

PRODUCT GUIDE 2009

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M-AUDIO **PRODUCT GUIDE** 2009



DJ Performance/Production Software

Torg® software is a fully equipped digital DJ workstation that provides innovative tools and production power to set you apart from the pack. In addition to essential cueing, beat matching and mixing, Torq goes beyond all other DJ applications by offering a host of real-time creative options unavailable anywhere else. Unique features like built-in effects processing, a 16-cell sampler, and ReWire support let you bring powerful studio processing to your live sets, as well as infuse your studio productions with live energy. Torq software is designed to work seamlessly with select M-Audio® DJ hardware, forming a completely integrated system for rock-solid performance. Version 1.5 brings dynamic new additions like Tempo Anchors, Tempo Master, zplane élastique time stretching, send/receive MIDI clock, a new toolbar and more.

Torq software is exclusively available as part of the Torq Xponent® and Torq Conectiv® DJ systems. Torg LE software is included with the X-Session® Pro control surface, Pro Tools[®] M-Powered[™] 8 and Pro Tools LE[®] 8.

cue, beat match and mix digital audio files (including MP3, AIFF, WAV. WMA, AAC and CD audio)

seamlessly integrates with select M-Audio DJ hardware

organizes digital audio files (including iTunes libraries) into an easily searchable database

supports optional third-party VST effects

ReWire support

2 virtual decks for loading and playing files

2-channel mixer with 3-band EQ, gain and level controls, headphone cueing and crossfader

snapshot function for instant recall of mixer and effects settings

16-cell, tempo-synced sampler

integrated performance recorder

vinyl, CDJ and hybrid emulation control modes control digital files with traditional DJ hardware

and/or MIDI controller

New Features in Tora 1.5

Improved Tempo/Pitch Alteration

Tempo Anchors allow marking tracks with varying tempos-Torg effects and samples follow the tempo changes in the song

Tempo Anchors enable you to eliminate tempo variations, making it easy to mix varied songs

Tempo Master assigns a playing track to serve as the master tempo source for all tempo-related functions

zplane élastique time-stretching/compression technology for sophisticated beat matching and

Advanced MIDI Control

transmit/receive MIDI clock for sync with other hardware/software and easy switching between DJs

MIDI soft takeover prevents parameter jumps when using external controllers

preview songs in headphones

warning message if loading a new

new database categories for bit

Musical Style Preferences improve

drag and drop songs into decks

before loading to deck

rate, file type, year and label

results when analyzing tracks

song onto a Deck that is playing

warning message when attempting to exit Torq

optimized control locations

intuitive toolbar buttons and

New User Interface Features

Hide Mixer option conserves screen space

while in browser max view

MIDI/tempo indicators

using Torq Control Vinyl and Torq Control CDs controllable via mouse, keyboard

compatible with Pro Tools M-Powered software

Global Tempo metronome can be monitored in headphones

Auto Gain feature matches the gain levels between decks

Enhanced Effects Processing

internal effects can be grouped into effects chains

VST effect tempo synchronizes with tempo of playing deck

effects can be used in a post-fader configuration

Rock-Solid Integration with Traditional

DJ Hardware Skip Protection keeps the music playing

even if the Control Vinyl or CD skips True Key Lock keeps the pitch constant when adjusting external turntable speed

New Looping Functionality

Loop Points can be set after a section plays new QuickLoop option for CDJ-style loop division Quantize function for Loop In/Out accuracy

For more information on Torq, including the new features in version 1.5, check out The Future of DJ Technology on pages 18-20 in the artist section of this magazine

Basics and Beyond

Torg software delivers everything you need to DJ in a single, easy-to-use workspace. The program features an integrated two-channel mixer with three-band EQs, level controls, headphone cueing and a crossfader-just like a traditional DJ setup. Then the power of digital DJing kicks in. Torq software's two virtual decks allow you to load, cue and mix digital files-plus automatically beat match tracks with the push of a button. Torq also organizes all of your files into an easily searchable database, so you can pull up the perfect song or change up your set in an instant. Torg software even gives you access to your entire iTunes library-literally putting all your music at your fingertips.

Enhanced User Interface

Torq 1.5 software incorporates several new features that improve the user interface and provide new creative possibilities. The Hide Mixer option conserves screen space by removing the mixer from the user interface and expanding the file browser window. The Browser Preview function lets you preview songs in your headphones before loading them into a deck. New file list categories such as "Year" and "Label" help with song organization-you can even hide and rearrange categories for improved searching through your database. Torq 1.5 also displays a warning message if you are exiting the program or loading a song onto a deck that's already playing, preventing accidental set interruption during a live performance.

Torg 1.5 takes your performance beyond simple beat matching and crossfading between tracks. Tempo Anchors let you set the phase grid exactly on the beats of a song, even if the song changes tempo. Tempo Anchors allow Torq to remove tempo variations from songs so you can easily blend old styles and new. For example, if you have an old '70s funk song that was recorded without a click, Tempo Anchors allow you to perfectly match that song with the steady beat of today's computer-driven productions. You can also assign a track you're spinning to serve as the Tempo Master for everything in Torq, so looped samples and other tempo-based features will automatically match the natural tempo changes in the song. These new features are enhanced by the addition of the zplane élastique time stretching algorithm, a time compression/expansion tool that performs extreme audio warping with exceptional precision and sonic quality, so you can keep your entire mix in key.

Audio Production Centerpiece

On stage and in the studio, Torq 1.5 provides the ideal centerpiece for your audio production setup. Want to hook up your outboard gear? Torg can send and receive MIDI clock, allowing total synchronization with drum machines, effects boxes and other computers. Multiple DJs can now connect with each other and share the same tempo across systems for improved collaboration between artists.

Torq Xponent

The professional-grade Torq® Xponent® system seamlessly integrates high-performance DJ software with a fully functional hands-on controller, representing a new era in digital DJ performance and production. The Xponent hardware combines the features of a two-channel DJ mixer with DJ CD-player-style tactile controls, letting you cue, mix and manipulate digital files without touching the host computer. In addition to the essentials, M-Audio® Torq 1.5 software delivers a variety of real-time creative tools to take your sets beyond the ordinary-and the Xponent hardware control surface puts all that power right at your fingertips. This integrated solution opens the door to new techniques that are simply impossible with traditional DJ setups.







"Xponent is an excellent physical representation of Torq software

perfect for gigs where space is tight. You just need Xponent and a

laptop. It's a great piece of hardware and integrates really well with



Dynamic Control Over Torq Software

The best DJ tools allow you to get your hands on your musicfrom cueing to beat matching to scratching. M-Audio designed the Xponent control surface with rugged, full-size, DJ CDplayer-style controls so you can get into the music and engage the crowd during each moment of your set. The unit includes two touch-sensitive scratch wheels for scratching, speeding up and slowing down the playback of digital files. And for indepth control over Torq, Xponent adds 64 assignable backlit buttons (including Play, Cue, Seek, Loop, Key and Sync controls), two volume sliders, eight assignable knobs for effect control and two 100mm pitch sliders. The built-in touchpad/ assignable X/Y controller allows you to control your mouse or the effects in Torg, so you never have to take your hands off the unit in the middle of a performance. Level and progress LED meters complete the list of professional features.



Made to Work Together



includes Torq DJ software

full-size, professional-grade controllers

Torq." -Johnny Juice (DJ; Public Enemy)

all hardware controls are pre-mapped to corresponding software functions in Torq

4-output USB audio interface:

- 4 RCA outputs for independent control of house system and booth monitors
- 1/4" headphone output with cue and volume controls
- 16-bit/48kHz max sample rate ASIO and Core Audio compatible
- advanced MIDI control surface:

mixer controls

- 2 vertical volume controls
- 2 3-band EQ control knobs with kill buttons
- 2 gain-control knobs with mute buttons · horizontal crossfader with transform buttons
- 2 LED volume meters
- - 2 touch-sensitive scratch wheels • 2 100mm pitch adjust controls
 - built-in touchpad/assignable X/Y control surface
 - · 8 assignable knobs for effect contro • 64 assignable backlit buttons, including Play, Cue, Seek,

Loop, Key and Sync controls 2 LED progress meters

compatible with any software that supports MIDI Learn

Pro Tools® M-Powered™ compatible* compatible with Mac and PC

*Requires Pro Tools M-Powered 7.4 or higher. Pro Tools M-Powered 7.4 and higher are paid upgrades from previous versions.



Torg Xponent provides much more than a transparent combination of hardware and software-it allows you to take DJing and live performance in new directions. With VST effect compatibility and easily assignable hardware controls, you can turn your plug-ins into expressive live performance tools. Call up a flanger effect and layer on LFOs-or mix in feedback loops from a delay effect and control it all from the assignable knobs on Xponent. The system also provides everything you need to loop, reorder and remix tracks on the fly. With built-in controls for looping as well as a 16-cell, tempo-synced performance sampler, Torq makes it possible to grab loops and one-shot samples and drop them wherever you like, all while staying perfectly in sync with the mix.



Torq 1.5 DJ software automatically configures itself to work with Xponent-every hardware control is pre-mapped to its corresponding function in Torg. It's easy to get

up and running with Xponent because all of the hardware controls are located exactly

where you'd expect them to be. The transition into digital DJing couldn't be simpler.

Torq Conectiv Vinyl & CD Pack

DJ Performance/Production System with Control Vinyl and CDs

The Torg® Conectiv® hardware/software system is designed for DJs who want to integrate the power of the computer with their turntables and DJ CD players. It features Torg 1.5 DJ software, the Conectiv USB bus-powered 4 x 4 audio interface and Torg Control Vinyl/CDs. Torg provides a creative new way to work with digital files, while Conectiv delivers high-fidelity audio with all the right kinds of I/O-including dual phono preamps. Use Torq Conectiv as a completely digital solution or process your vinyl and CDs on the fly. You can also create a hybrid digital/vinyl DJ rig with the included Torq Control CDs and Torq Control Vinyl discs. And unlike most digital DJ systems, the Conectiv interface works with third-party software as well.

For a complete description of Torg software, see page 6 in the product section, and The Future of DJ echnology on pages 18-20 in the artist section.



"The most fun I've had DJ'n since I learned to scratch." –DJ Revolution (DJ: Power 106)

includes	Tora	DJ	software

includes 2 Torq Control Vinyl Discs and 2 Torq Control CDs

- 4 x 4 USB audio interface with 16-bit/48kHz fidelity
- 4 RCA inputs configurable for line-level or phono signals
- 4 RCA line outputs with +10dBV peak output level
- 2 mix knobs for blending between the audio input and computer output

1/4" headphone output with volume control

crossfadable cue source with headphone split function

1/4" microphone input (dynamic mics only)

compatible with any Mac or PC software using ASIO or CoreAudio drivers

Pro Tools® M-Powered™ compatible*



Premium Interface Technology

The Conectiv USB interface delivers world-class M-Audio technology, outfitting your digital DJ rig with two pairs of RCA stereo inputs and outputs, phono preamps and a microphone input. Two large knobs allow you to seamlessly mix between the incoming analog audio from a turntable or CD player and the digital output of a connected Mac or PC. Conectiv also includes standard DJ cueing capabilities, so you can listen to any track in your headphones before sending it out to the house system. And with a maximum output level of +11dBV, Conectiv gives you enough volume to go toe-to-toe with any DJ hardware on the market.

The included Torq Control Vinyl records and Torq Control CDs allow you to manipulate digital audio files with the same feel and response as a traditional DJ setup. Simply connect your turntables or DJ CD players to Conectiv, enable external control in Tora, and you'll be able to cue and scratch your music just as if you were using the original records or CDs. Extremely low latency and super-fast response allow for battle-style scratching techniques and easy beat juggling. Torq also includes multiple operating modes that can make playback behave like a traditional turntable or a DJ CD player-and there is even a hybrid mode that blends the strengths of both vinyl and CD-style control.

Universal Compatibility

While Torq and Conectiv are designed to work together seamlessly, Conectiv is also compatible with any Mac or PC software using CoreAudio and ASIO drivers—delivering low-latency performance as a recording interface for DAWs like Ableton Live, Logic, Cubase, Digital Performer and more. Of course, it's also compatible with Pro Tools M-Powered software* as your entrance into the world of professional recording.

"A Serato killer?! Never say never... Seeing the combination of the Conectiv... and the powerful Torg software, dangerous, Watching it run on Windows XP on a MacBook Pro. serious, Finding out that you can also use your VST plug-ins seamlessly into your digital DJ software, priceless."



Torq Control Vinyl & Torq Control CD

Each Torq® Control CD and Torq Control Vinyl disc allows you to manipulate audio files in Torq using your favorite DJ CD player or turntable, respectively, with the same feel as the real thing. Simply connect your turntable or DJ CD player to a Torq-compatible audio interface like the M-Audio Conectiv®, enable external control in Torg, and you'll be able to cue and scratch your music just as if you were using the actual vinyl or CDs. Super-fast response allows for even the most intricate scratching techniques, and bi-directional positioning ensures that Torg precisely tracks the vinyl or CD for easy beat-juggling. Each disc or CD is sold individually.





X-Session Pro

USB MIDI DJ Controller

The X-Session® Pro USB MIDI controller is the easiest way to bring traditional DJ mixer-style tactile control to the world of computer DJing and live performance. This complete system features Torq® LE software and a hardware control surface that fuses the standard controls of a two-channel DJ mixer with the transport and pitch controls of a turntable or DJ CD player. The mixer controls allow you to make smooth crossfades and adjust level, cueing and EQ parameters-while the integrated transport and pitch controls allow you to start, stop and beat match digital files. X-Session Pro interfaces with any software that supports MIDI Learn, so you can bring DJ-inspired crossfades and filter sweeps into your live performances as well as studio productions.

4 MIDI sliders 12 MIDI knobs 10 MIDI buttons MIDI crossfader

DJ mixer-inspired layout includes Torq LE DJ software powered via USB

class-compliant with Windows XP. Vista and Mac OS X'

works with Torq and other software that supports MIDI Learn on/off switch

*Driver installation required for multi-client operatio



Torg LE Software

The included Torq LE software organizes everything you need to DJ into a single, easy-to-use workspace. Two virtual decks let you load, cue and play back digital files-and will even automatically beat match the files in an instant. And just like a traditional DJ setup, Torg LE includes an integrated two-channel mixer with threeband EQ, level controls, headphone cueing, a crossfader and three effects.

Torq-Made for **MIDI**

If you're the type of DJ who likes to get your hands on your music, Torq makes it easy. In addition to your M-Audio DJ hardware, you can use Torq with any standard MIDI controller–like the Trigger Finger™ or even the Oxygen and Axiom® line of keyboard controllers. Simply connect one of these devices and start mapping controller data. For instance, you could control the onboard effects within Torq with the sliders on the Trigger Finger while reserving the pads for snapshot settings that allow you to instantly switch between virtual mixer set-ups. The options are limitless. See page 52 for more information.



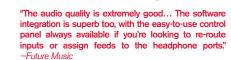
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ProFire 2626

High-Definition 26-In/26-Out FireWire Audio Interface with Octane Preamp Technology

Building on a legacy of award-winning FireWire audio interfaces, the M-Audio[®] ProFire™ 2626 delivers next-generation performance to your PC or Mac host-based recording system. This powerful interface provides 26 x 26 simultaneous I/O, complete with an onboard DSP mixer sourced from up to 52 audio streams. It features every kind of connection you need–including award-winning Octane™ preamp technology on all eight analog inputs, ADAT, S/PDIF, word clock and MIDI. Premium A/D-D/A converters and critically acclaimed JetPLL jitter elimination technology ensure pristine audio quality and reliable synchronization all the way up to high-definition 24-bit/192kHz resolution. Low-latency drivers and compatibility with most major DAWs–including Pro Tools® M-Powered™* software–complete the best-sounding, most flexible host-based audio interface available.



"Some vintage pres you get for a certain sound, but you can't use them on everything. I woudn't hesitate to use the ProFire 2626 pres on anything."—Buddy Miller (Grammy-nominated singer-songwriter)

up to 24-bit/192kHz for high-definition audio

- 8 analog inputs sourced from:
- . 8 XLR mic inputs with phantom power
- 8 1/4" TRS balanced line inputs
- 2 1/4" TS instrument inputs

optical I/O:

- 2 x 2 S/PDIF (on optical port B) or
- 16 x 16 ADAT (8 x 8 in S/MUX II mode, 4 x 4 in S/MUX
- 2 x 2 coaxial S/PDIF (via included breakout cable)

- 8 1/4" TRS balanced line outputs
- 2 1/4" TRS headphone outputs
- 1 x 1 MIDI I/O (via included breakout cable)
- word clock I/O (via included breakout cable)
- 8 preamps with award-winning Octane technology
- user-assignable master volume knob

flexible onboard DSP mixer:

- select from any of the 26 hardware inputs and 26 software returns as sources for its 18 input channels
- · create up to 8 stereo mixes

standalone 8 mic pre/8-channel A/D-D/A converter

JetPLL jitter elimination technology

sync options include internal clock, ADAT optical, S/PDIF and word clock

compatible with most popular DAWs including Pro Tools M-Powered*

includes Live Lite

*Requires Pro Tools M-Powered 7.4 (with downloadable update from m-audio.com) or higher. Pro Tools M-Powered 7.4 and higher are paid upgrades from previous versions. Maximum 96kHz and 18 x 18 I/O when using Pro Tools M-Powered.

Award-Winning Octane Preamp Technology

The quality of your preamps is critical to how your music sounds. ProFire 2626 features the same award-winning Octane technology that reviewers and audio professionals have praised in the M-Audio Octane eight-channel standalone preamp—giving you pristine audio fidelity right from the start of your signal path. In addition to exceptional sound, each preamp has a generous 75dB gain range, 48V phantom power and a -20dB pad.

User-Assignable Master Volume Control

The ProFire 2626 interface delivers a new level of flexible control via the programmable master volume knob. It's assignable to any or all of the four analog output pairs, so you can choose which output pair will feed your main monitors. You can also choose whether the master volume knob influences either or both of the two headphone outputs, or operates independently of them. In addition, you can assign the master level control to all four output pairs to uniformly attenuate up to a 7.1 surround mix.





"... used as a standalone outboard box, I found the pres clear enough to warrant running drums, guitar and vocals through it on a rock session without even hooking it up to my computer... and I have many, many boutique pres in my collection from which I can choose." –EQ (July 2008)

"As a superhighway of audio inputs and outputs for one's computer, the M-Audio ProFire 2626 is one of the most sophisticated interfaces I've ever used. It is the best sounding DAW interface I've ever owned, with converters and mic preamps of a quality that would have been undreamed of just a few years ago at this price level. I recommend it highly." – Dr. Frederick J. Bashour,

Standalone Operation

ProFire 2626 doubles as a standalone eight-channel preamp, eight-channel A/D-D/A converter and digital format converter. In standalone mode, the eight analog inputs route to the first eight ADAT optical outputs. This facilitates use of the line, instrument and mic ins—including preamps—with other gear. Similarly, the first eight ADAT optical inputs feed the eight analog outputs to provide D/A functionality for other digital devices. You can also disable the D/A conversion to allow the analog input channels to route to both the analog and ADAT optical outputs—transforming ProFire 2626 into a standalone eight-channel mic pre with eight discrete analog outputs plus eight-channel ADAT output. When the second optical port is not used for A/D conversion, ProFire 2626 can also convert coaxial S/PDIF to optical S/PDIF.

Onboard DSP Mixer and Router

Customize your setup and record the way you want thanks to the power and flexibility of the ProFire 2626 onboard DSP mixer and router. Use the router to select from up to 52 audio streams and send them to any of the 26 hardware outputs.* The adaptable onboard DSP mixer allows you to select from any of the 26 hardware inputs and 26 software returns as sources for any of its 18 input channels—allowing you to create up to eight stereo mixes. You can also create presets and instantly recall these setups at a later date.

*Routing in stereo pairs.

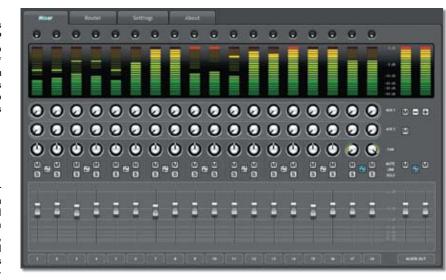
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Word Clock

High-Definition 192kHz Sound and Solid Performance

ProFire 2626 employs high-quality converters to deliver sampling rates up to 192kHz for pristine, high-definition sound quality. The unit also features critically acclaimed JetPLL technology to provide extremely stable synchronization and eliminate jitter that can degrade audio quality. In addition, ProFire 2626 allows master or slave operation via internal clock, ADAT optical, S/PDIF and word clock. M-Audio's mature, stable drivers deliver solid performance and ultralow latency. Complete with Octane preamp technology and premium A/D-D/A converters, you won't find a better sounding or more flexible host-based audio interface.

8 - 888



adjust the DSP mixer and router via the software control panel

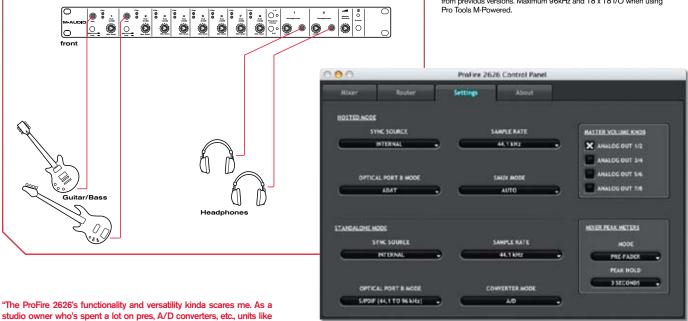


ProFire 2626 features essential connectivity for any studio, delivering rock-solid stability and optimal sound quality. Analog inputs are sourced from the eight combo jacks as either XLR mic or 1/4" TRS balanced line inputs. In addition, analog channels 1 and 2 provide easily accessible 1/4" TS instrument jacks on the front panel. Eight 1/4" TRS connectors provide balanced line outputs for monitoring and interfacing with other outboard gear. Digital I/O is available via 2 x 2 S/PDIF and 16 x 16 ADAT Lightpipe. The two front-panel headphone jacks feature separate sources and independent level controls for additional cue mixing flexibility.

Pro Tools M-Powered Compatible

ProFire 2626 is an extremely powerful interface for Pro Tools M-Powered.** Flexible I/O, top-quality sound and low latency round out an incredible recording environment that is file-compatible with many other pro and project studios around the world.

**Requires Pro Tools 7.4 (with downloadable update from m-audio.com) or higher. Pro Tools M-Powered 7.4 and higher are paid upgrades from previous versions. Maximum 96kHz and 18 x 18 I/O when using Pro Tools M-Powered.



Mic, Keyboard

Mixing Desk

"The ProFire 2626's functionality and versatility kinda scares me. As a studio owner who's spent a lot on pres, A/D converters, etc., units like this make me feel like I've wasted a lot of money. I mean, what's next? Large-format consoles for under \$5000?" –EQ (July 2008)

software control panel settings tab



ProFire 610

High-Definition 6-In/10-Out FireWire Audio Interface with Octane Preamp Technology

Designed by the company that revolutionized mobile music production, the ProFire™ 610 FireWire audio interface transforms your Mac or PC into a powerful 6-in/10-out recording studio. Premium digital converters deliver high-definition, 24-bit/192kHz audio throughout the signal path. Two preamps with award-winning Octane™ technology offer clean, transparent sound with low noise and exceptional headroom. Onboard DSP mixing provides a totally flexible recording experience-perfect for creating multiple, independent monitor mixes. The user-assignable master volume knob gives you flexible control over output levels. MIDI I/O allows you to connect keyboards and other outboard MIDI hardware. The half-rack-space unit also doubles as a standalone two-channel microphone preamp and A/D-D/A converter. ProFire 610 is compatible with most major DAWs including Pro Tools® M-Powered™* software, making it an unbeatable choice for personal, high-definition recording.

up to 24-bit/192kHz for high-definition audio

flexible onboard DSP mixer

2 preamps with award-winning M-Audio Octane technology:

- phantom power
- 20dB pad on each preamp
- signal/peak LED indicator lights
- 6 inputs including:
- 2 XLR/TS combo jacks for mics or instrument level inputs
- 2 1/4" TRS balanced line inputs
- stereo S/PDIF in

10 outputs including:

- 8 1/4" TRS balanced line outs
- stereo S/PDIF out

2 1/4" TRS headphone outputs

built-in 1 x 1 MIDI interface

2 FireWire (IEEE 1394) ports

JetPLL jitter elimination technology

standalone operation as 2 mic pre/2-channel A/D-D/A converter

user-assignable master volume knob

powered via FireWire bus or external power supply

half-rack-space form factor fits

in Universal Rack Tray

compatible with most major audio software including Pro Tools M-Powered*

includes Live Lite

*Requires Pro Tools M-Powered 8.0 or higher. Pro Tools M-Powered 8.0 is a paid upgrade from previous version

Onboard DSP Mixer

ProFire 610 features an onboard DSP mixer, giving you an extra level of flexibility and improving the overall performance of your system. Use the software control panel to create custom configurations for different recording setups, then save the settings and instantly recall them later. You can set up five unique stereo mixes sourced from any of the hardware inputs and software returns simultaneously-ideal for assigning separate monitor mixes to multiple musicians. The onboard DSP mixer also enables you to monitor while tracking, just as you would with your hardware mixer.

Professional-Quality Design-To Go

Once upon a time, choosing a mobile recording interface meant sacrificing some of the features and quality found in big-budget studio equipment. Now ProFire 610 changes the game by delivering the same advanced technology and design elements found in our acclaimed ProFire 2626 recording interface. Harness the power of award-winning Octane preamp technology, DSP mixing and 24-bit/192kHz audio resolution in a bus-powered unit perfect for recording on the go.





Award-Winning Octane Preamp Technology

Great recordings start with great preamps. ProFire 610 brings you the same award-winning M-Audio Octane preamp technology that audio professionals trust for clean, transparent sound. Each preamp delivers low distortion and an extremely high signal-to-noise ratio-providing pristine audio fidelity right from the start of your signal path. In addition to exceptional sound, each preamp has a generous 75dB gain range, 20dB pad and 48V phantom power.

	Connection	Maximum Speed
	USB 1.1	12Mbps
	USB 2.0	480Mbps
>	FireWire 400	400Mbps

High-Definition 192kHz Sound and Solid Performance

M-Audio designed the ProFire 610 using carefully selected components, including premium A/D-D/A converters that keep your audio signal uncolored and true to any input source. We engineered the electronic circuitry to provide an optimal signal path between input and output, resulting in extremely clean recordings that capture every nuance of your performance. ProFire 610 also features critically acclaimed JetPLL jitter-elimination technology for stable synchronization and exceptionally low audio band jitter. Building on M-Audio's time-proven FireWire driver technology-found in the best-selling FireWire 410 interface-ProFire 610 delivers solid performance and reliability at sample rates up to 24-bit/192kHz.

ProFire Lightbridge

ProFire™ Lightbridge interfaces up to four Lightpripe devices with a PC or Mac via FireWire—perfect for integrating digital mixers, A/D-D/A converters and/or ADATs with most popular DAW software. The unit is capable of 32 channels of Lightpipe I/O, two channels of S/PDIF I/O and two-channel analog output for a total of 34 x 36 concurrent channels. ProFire Lightbridge can output word clock as a master, or slave to external sync on any input, and MIDI I/O also accommodates MIDI Time Code and MIDI Machine Control, Independent volume controls are provided for the 1/4" balanced outputs and front-panel 1/4" headphone iack, ProFire Lightbridge is compatible with most DAWs including Pro Tools® M-Powered™* software.



"... the ProFire Lightbridge is an extremely cost-effective way to add many channels of digital I/O to a FireWire-equipped computer." -Tape Op

32-channel Lightpipe I/O at 44.1 or 48kHz

16-channel Lightpipe I/O at 88.2 or 96kHz using S/MUX

S/PDIF I/O (via included breakout cable, coax)

2 1/4" TRS analog outputs with dedicated level control

all analog and digital I/O active simultaneously (total of 34 x 36 concurrent channels at 44.1 or 48kHz; 18 x 20 at 88.2 or 96kHz)

word clock I/O (via included breakout cable, BNC) MIDI I/O (via included breakout cable)

four 8-channel preamps = 32 inputs

1/4" front-panel headphone out with dedicated level control

up to 24-bit/96kHz operation

FireWire connectivity to computer

indicators for sample rate, Lightpipe I/O, MIDI I/O, sync source and power

Kensington lock slot

half-space rack chassis, optional mounting tray

compatible with most major DAWs including Pro Tools M-Powered*

includes Live Lite

*Requires Pro Tools M-Powered 7.3 or higher. Pro Tools 7.3 and higher are paid upgrades from earlier versions of Pro Tools M-Powered. Pro Tools M-Powered currently supports a maximum of

Transfers

Need to transfer tracks between an ADAT system and your computer? ProFire Lightbridge does it with ease. Send up to 32 channels of Lightpipe simultaneously at 44.1 or 48kHz-or transfer up to 16 channels at 88.2 or 96kHz using the S/MUX protocol.

Live Multitrack Recording

Combine ProFire Lightbridge with up to four eight-channel preamps and you have a compact system ready to record up to 32 channels of input for live shows.







M-AUDIO

FireWire Solo

FireWire Mobile Audio Interface for Songwriters/Guitarists

FireWire Solo is an easy-to-use, high-quality interface for songwriters to record guitar, vocals and more on a Mac or PC computer. Compatible with most music software, this compact bus-powered* unit features inputs for guitar, microphone and line-level gear. FireWire connectivity, up to 24-bit/96kHz and digital I/O all translate to outstanding sonic performance.

guitar input (1/4" unbalanced) 2 line inputs (1/4" unbalanced) on rear panel 2 line outputs (1/4" TRS balanced/unbalanced) digital S/PDIF (coaxial) I/O w/ 2-channel PCM or pass-through of surround-encoded AC-3 and DTS material stereo headphone output (1/4" TRS) with level control flexible software-based routing/mixing with zero-latency monitoring includes GT Player Express software powered via FireWire bus* or optional 12V DC power supply Pro Tools® M-Powered™ compatible; includes Live Lite

6 x 4 24-bit/96kHz audio interface; 4 x 2 analog operation

microphone in (XLR) with 48V phantom power

*Requires powered 6-pin FireWire connection.

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ProjectMix I/O

Control Surface with Motorized Faders and 18 x 14 Audio Interface

Today, more professional music is produced at home than ever before—and the ProjectMix I/O control surface delivers what you need to take your Mac- or PC-based studio and productions to the next level. Seamless integration with all major DAW software. The ability to record directly into industry-standard Pro Tools® sessions. Faders so you can feel the mix with your fingertips instead of dragging a mouse. Onboard display of critical parameters for intuitive operation. Motorized control to craft more accurate mixes. And professional multichannel I/O including mic/instrument preamps, Lightpipe and S/PDIF. ProjectMix I/O is the universal solution that combines the best of the hardware and software worlds for a new standard in streamlined production. Compatible with Pro Tools M-Powered™ 7, Ableton Live, Logic,* Cubase,* Digital Performer* and SONAR*

Complete Production Experience

Computer-based DAWs seem to have virtually no limitations—except those imposed by using a mouse to do the job of dedicated hardware. ProjectMix I/O finally puts all that tactile control back under your fingertips. It also combines that control with M-Audio's acclaimed FireWire audio interface technology, eliminating clutter while integrating everything you need to transform your computer-based music system into a full-fledged digital audio workstation. ProjectMix I/O supports both audio and MIDI integration with most digital audio workstations. It also supports MIDI control only for all Pro Tools software.

Total I/O

ProjectMixI/O gives you just about every kind of I/O you might need for a session. Eight analog input channels all feature 1/4" balanced and XLR microphone inputs with individual mic/line switches. Of course, phantom power is provided for up to eight condenser mics and input 1 even features a front-panel instrument input so you can easily plug a guitar or bass right in. ADAT Lightpipe I/O brings the simultaneous input total to 16 and allows you to expand your system with devices like M-Audio's award-winning Octane preamp. This allows for eight more preamp channels via Lightpipe. S/PDIF handles all your other digital connectivity, and word clock synchronizes ProjectMix I/O with other digital devices.

ProjectMix I/O also serves as a 1 x 1 MIDI interface, allowing your computer to connect with MIDI hardware. Monitoring options include two 1/4" headphone outputs with separate volume controls for working with a partner, as well as two stereo output pairs for routing to monitors, processors and mixdown devices. There's even an A/B headphone source monitoring switch for cueing. ProjectMix I/O brings all of this to your computer with a single, simple FireWire connection.





"... the ProjectMix's cross-sequencer compatibilty, large LCD display and integrated audio interface make it a superior choice for project studios seeking a flexible alternative to keyboard-and-mouse mixing." –Future Music

Control

8 touch-sensitive motorized channel faders

1 touch-sensitive motorized master fader

8 assignable rotary encoders and large LCD for channel or plug-in functions

mic/line, record enable, select, solo and mute buttons on each channel

channel +/- and bank +/- keys

illuminated transport controls for record, play, stop, fast forward and rewind

jog/shuttle wheel with mode switch

zoom key with 4 directional keys dedicated keys for in/out points, locate, region nudge, loop and more

A/B headphone source switch for cue auditioning

function LED display for each channel

1/0

8 analog mic/line inputs (1/4" TRS and XLR balanced) with signal/peak indicators

4 analog outputs (1/4" TRS balanced)

8 x 8 ADAT Lightpipe I/O

front-panel instrument input (1/4" TS) switchable for input 1

2 x 2 S/PDIF digital I/O

1 x 1 MIDI I/O with activity LEDs

FireWire (IEEE 1394) connectivity

word clock I/O (BNC)

phantom power

2 stereo headphone outputs (1/4" TRS) with independent level controls

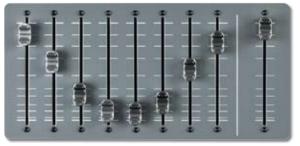
Kensington lock slot

compatible with Pro Tools M-Powered 7, Ableton Live, Logic,* Cubase,* Digital Performer* and SONAR*

includes Live Lit

*DAW software version must support Mackie Control, Logic Control or HUI protocol.

Total I/O ProjectMix I/O gives you just about every kind of I/O you might need for a session-and at the same price as many control surfaces lacking audio altogether.



10-Bit Touch-Sensitive Moving Faders

The motorized touch-sensitive faders on each of the eight channel strips and master strip record and play back your changes, making for seamless mix automation. And touch sensitivity means that when you touch a fader in motion, it immediately disengages for manual control and re-engages automation when you release. This provides for both smoother operation and the elimination of motor burnout prevalent in older technologies. The ProjectMix I/O faders also feature higher resolution than other control surfaces in the same class.

Channel Strip

Simply press the bank +/- and channel +/- controls to map the eight channel strips to as many channels as your project requires. Each channel strip features all of the functions you'd expect to help you zip through sessions, including buttons for mic/line, record enable, mute, solo and select. In addition to a touch-sensitive motorized fader, each strip has an endless rotary encoder knob whose function is assigned by one of 15 dedicated buttons for EO, aux send and more. Related information is displayed in the LCD window directly above each encoder. The encoders and LCD double to control assigned plug-in parameters as well.

"It's hard to thumb your nose at a 30-pound



software control panel



Transport Control

Dedicated transport controls are another feature that makes sessions with ProjectMix I/O a breeze. You get illuminated controls for record, play, stop, fast forward and rewind, plus a jog/shuttle wheel and locate buttons for easy project navigation. Sets of keys for in/out points, zoom, region nudge, looping and more will having you flying through sessions with incredible efficiency.



"Songwriters now need just one mixer for both the home studio and performance. The NRV10 is clean enough to serve as a front end, and FireWire audio is versatile enough to be mixed well in any venue." -Performing Songwriter

"Large-sounding... killer stage-mixing capabilities, and thorough FireWire integration make the [NRV10] the jack-of-all-trades nerve center that it should be" -Remix

> "There are subtleties in my music that can sometimes get lost through a live sound system, but the NRV10 has helped me enhance these details." -Kate Havnevik (singer/songwriter)

"If you're looking for a versatile audio interface but don't want to give up the hands-on aspects. of a physical mixing console, you'll definitely want to check out the NRV10." -Electronic Musician

If you have a laptop studio that needs to be mobile, the NRV10... simplifies the setting up of a home studio, while saving the cost of buying a separate audio interface, monitor controller and either a small mixer or a number of dedicated mic preamps." -Sound On Sound

nigh-quality 8 x 2 analog mixer with integral 10 x 10 FireWire audio interface 24-bit/96kHz on all channels

4 mono channels, 2 stereo channels, each with:

- 1/4" TRS balanced line input
- · channel/FireWire selector
- gain control
- monitor send
- effect send
- pan/balance
- volume fader with peak indicator LED
- mute/cue button

balanced XLR mic inputs with award-winning Octane technology, each featuring mic/line switches

1/4" insert jacks on all mono channels

3-band EQ on each channel

integrated digital effects processor with 16 effects, plus variations, mute. and peak LED

2 mono aux sends and 2 stereo aux returns

both XLR balanced and 1/4" TRS balanced main outputs

headphone output with flexible source assignment and DJ-style pre-listening

independent volume for mix, control room and headphones

included NRV10 interFX software facilitates live mixing with included dynamic effects, VST compatibility and software recall:

- add real-time insert effects to each NRV10 channel
- built-in effects include compression and expansion-gating for each channel, flanger, chorus,
- host two VST plug-ins per channel
- save and recall setups to easily change effect configurations between songs
- instant setup through automatic configuration with NRV10 hardware

also functions as a standard analog mixer

durable metal chassis with optional rack-mount bracket

compatible with major DAWs including Pro Tools M-Powered* (demo included); includes Live Lite

*Requires Pro Tools M-Powered 7.3 or higher. Pro Tools M-Powered 7.3 and higher are paid upgrades from earlier versions of Pro Tools M-Powered.

The Missing Link for Live Computer Tracks

without ever needing to change cabling or connections.

NRV10

10 x 10 FireWire Digital Audio

gap between studio and stage as well.

Your Studio's Nerve Center

Interface | 8 x 2 Analog Mixer with Effects

NRV10 combines an 8 x 2 analog mixer and a 10 x 10 24-bit/96kHz FireWire

digital audio interface in one convenient package—the perfect tool for computer-

based recording and performance. Now you can route and record mixer

channels directly to individual tracks in most popular software studios-including Pro Tools® M-Powered™ software.* You can also return multiple discrete audio

channels from your Mac/PC to NRV10 for CPU-free mixing and monitoring

complete with onboard digital effects-great for doing live performances

without changing settings like EQ and reverb in your files while you play. The

included NRV10 interFX™ application even allows you to turn your NRV10 and

host computer into a powerful live mixer with access to all of your VST effects. NRV10 seamlessly integrates your computer with your studio-and bridges the

The eight analog inputs on NRV10 let you keep all of your favorite instruments

and other gear connected while composing or practicing-even when your

computer is off or disconnected. And, thanks to its integral 10 x 10 FireWire

interface, you can record any or all of them to computer-based tracks any time you wish. Conversely, you can also discretely return multiple channels of pristine

digital audio to the mixer for flexible monitoring, mixing and processing. Inline

monitoring on the same channels from which you recorded means you can just

leave your levels set where they are. Use the two aux busses to route to external processors or create a custom headphone mix. In the studio, the NRV10 lets

you move seamlessly between writing, recording, practicing, editing and mixing

Using computer-based tracks on live gigs used to require making and saving software adjustments for parameters like levels, EQ and effects just so everything sounded right. Now that's history. NRV10 gives you easy analog mixer control over direct feeds from multiple computer-based tracks. Perfect your studio tracks, then just make temporary tweaks on NRV10 for live performances. As a bonus, you can run live vocals and more through the same processing you used on those studio tracks. You can even add built-in digital effects to tracks without modifying your files. NRV10 puts you back in control of the mix that you and your audience hear live.

NRV10 interFX Turns Your Computer into a **Multi-Effects Processor** for the Stage

The included NRV10 interFX software for Windows and Mac turns the NRV10 and your host computer into an even more powerful mixing console complete with multi-effects processing. The application adds a compressor, expander/gate and two VST effect slots to each mixer channel-letting you process your live instruments and mics with your favorite computer-based effects. Of course, you can also save and recall settings for later use. Now, with the NRV10 and NRV10 interFX, you can bring the VST plug-in effects you use in the studio to the stage.

More Live Uses

There are endless live applications for NRV10. Its flexible monitor source assignment can send the drummer a headphone click trackand even lets your DAW's automation change effects and other parameters on the fly for different song sections. Or if you're creating live dance music, the monitoring section lets you audition material in your headphones DJ-style before routing it to the main mix. Have some soft synths? A laptop, NRV10 and an M-Audio MIDI controller are a great way to perform with them on stage. And if you're recording a band live, NRV10 lets you easily record different players to separate tracks using an application like Pro Tools M-Powered* software.

Analog Heart, Digital Mind

A FireWire digital interface is not a pricey extra option on NRV10. It's built right in, giving you the best of analog mixing and digital recording in a single package. The ability to route analog inputs and outputs in conjunction with digital

tracks and busses provides for seamless creative flow without multiple devices or constantly plugging and unplugging cables. The FireWire interface and software drivers build on M-Audio's years of experience perfecting digital audio interfaces to critical acclaim. You get unbeatable specs like pristine digital audio quality all the way up to 24-bit/96kHz. And unlike other FireWire mixers, NRV10's ability to return multiple channels from your computer gives you much more flexibility compared to simple



Professional Mixer

Featuring a durable metal chassis with optional rack-mount bracket, NRV10 can be used either with a computer or as an excellent standalone mixer. M-Audio's award-winning Octane preamp technology assures best-of-class performance. The unit features four stereo and four mono channels-and the first five inputs have phantom-powered XLR mic inputs as well. Its design features everything you'd expect in a professional mixer including balanced inputs, channel inserts for inline processing, two aux busses, 3-band EQ on each channel and a flexible monitor section. There's also a built-in effects section with 16 effects so you don't even need to burden the computer for basic effects-especially handy live or when practicing.







"The Fast Track Ultra 8R could be the perfect combination of fidelity, flexibility, and friendliness." -Remix

"If you want to record multiple sources simultaneously, this rack-mountable interface is a winner." –Musicradar.com

High-Speed 8 x 8 USB 2.0 Interface with 8 Preamps and MX Core DSP

Fast Track Ultra 8R

Designed for comprehensive studio work, the Fast Track[®] Ultra 8R audio/MIDI interface delivers 8 x 8 I/O, high-speed USB 2.0 connectivity, MX Core™ DSP mixing/ effects and eight preamps with award-winning Octane™ technology. Record drum kits and full bands on all eight inputs simultaneously—with superior 24-bit/96kHz fidelity. The onboard MX Core DSP mixer processes eight hardware inputs and eight software returns to the eight hardware outputs, expanding the total I/O to an impressive 16 x 8 configuration and delivering flexible routing and monitoring with delay and reverb. With reliable, low-latency M-Audio drivers and compatibility with most digital audio workstations including Pro Tools® M-Powered™ software, the Fast Track Ultra 8R puts the core of your powerful project studio in a single rack-mount unit.

24-bit/96kHz professional audio on all I/O simultaneously

high-speed USB 2.0 interface

MX Core DSP technology:

- monitor with effects while recording
- flexible channel routing options
- expands DSP mixing interface to 16 x 8

2 front-panel 1/4" instrument inputs

2 analog inserts

8 preamps with award-winning M-Audio Octane technology:

- 8 XLR/TRS combo jacks for mic or line-level inputs
- 20dB pad on each preamp
- signal/peak LED indicator lights
- +48V phantom power

8 balanced TRS analog outputs

S/PDIF digital I/O

built-in 1 x 1 MIDI interface

2 independent headphone outputs with individual volume control knobs

rugged, lightweight metal chassis

full ASIO 2.0, WDM, MME and Core Audio support

compatible with most major audio production software including Pro Tools M-Powered* includes Live Lite

Maximum Speed

12Mbps

480Mbps

400Mbps

*Requires Pro Tools M-Powered 7.4 (with downloadable update from m-audio.com) or higher. Pro Tools M-Powered 7.4 and higher are paid upgrades from previous versions.





Shared Features

High-Speed USB 2.0 Technology

The next generation of recording interfaces has arrived—the Fast Track Ultra and Fast Track Ultra 8R interfaces feature high-speed USB 2.0 technology for advanced performance. Like FireWire, USB 2.0 delivers the bandwidth for 24-bit/96kHz audio resolution throughout the recording and monitoring process. The increased data speed lets you enjoy exceptional fidelity on all of your inputs and outputs simultaneously.



Built-in Effects Processing Tired of tracking vocals and inst too much latency? Built-in MX C

Tired of tracking vocals and instruments dry because plug-in effects introduce too much latency? Built-in MX Core DSP technology provides reverb and delay on the headphone outputs with separate sends for every channel. Now you can inspire creativity by adding effects—without taxing your CPU resources.

Connection

USB 1.1

USB 2.0

FireWire 400





Professional Preamps with Octane Technology

The quality of your recordings is only as good as the initial signal. Fast Track Ultra 8R features eight preamps with multi-award-winning M-Audio Octane technology—while Fast Track Ultra features four such preamps—to create a premium front end that delivers exceptional audio clarity. You get professional, low-noise, high-gain preamplification with enough channels for various applications. Features include +48V phantom power for condenser mics, signal/peak LED indicator lights and a pull-out gain knob that activates a 20dB pad for recording at high volumes.

"The Fast Track Ultra provides a solid set of affordable studio interface functions, and gets multitrack audio interfacing into the hands of Windows-based users who might have been frustrated in the past by FireWire's quirks. Mac users will also benefit from its seamless and powerful feature set... there's plenty to love in the Fast Track Ultra." —Recording



Fast Track Ultra

High-Speed 8 x 8 USB 2.0 Interface with MX Core DSP Technology

The Fast Track[®] Ultra 8 x 8 audio/MIDI interface takes M-Audio's acclaimed mobile recording line to the next level with high-speed USB 2.0 connectivity, MX Core™ DSP mixer and four preamps with award-winning Octane™ technology. Featuring both analog and digital I/O, it allows recording on all eight channels simultaneously with pristine 24-bit/96kHz fidelity—while M-Audio's mature low-latency drivers ensure rock-solid stability and performance. The onboard MX Core DSP mixer processes eight hardware inputs and eight software returns to the eight hardware outputs, delivering flexible routing and monitoring with delay and reverb. Compatible with most major digital audio workstations including Pro Tools® M-Powered™* software, Fast Track Ultra lets you record the way you want—in the studio or on the go.

24-bit/96kHz professional audio on all I/O simultaneously

high-speed USB 2.0 interface

MX Core DSP technology:

- digital effects processing with assignable parameters
- flexible channel routing options

• expands DSP mixing interface to 16 x 8

6 balanced line inputs

- 4 preamps with award-winning M-Audio Octane technology:
- 4 XLR microphone inputs-including 2 XLR/TRS combo jacks for mic or instrument level inputs
- 20dB pad on each preamp
- signal/peak LED indicator lights
- +48V phantom power

· gain control for each preamp

· headphone volume control

• 4 XLR microphone inputs -

*Requires Pro Tools M-Powered 7.4 (with downloadable update from m-audio.com) or higher. Pro Tools M-Powered 7.4 and higher are paid upgrades from previous versions.

ear view
• 2 analog inserts ——

6 balanced analog outputs

built-in 1 x 1 MIDI interface

software; includes Live Lite

power supply required for full 8 x 8 operation.

2 analog inserts

S/PDIF digital I/O

USB bus power**

- 6 balanced line inputs
- S/PDIF digital I/O
 1 x 1 MIDI interface
- 1 x 1 wildi interlace

lanced analog outputs

2 independent headphone outputs with individual volume controls

compatible with Pro Tools M-Powered* and most major audio production

**USB bus power enables analog channels 1 and 2. S/PDIFI/O, and headphone output 1: included

full ASIO 2.0, WDM, MME and Core Audio support

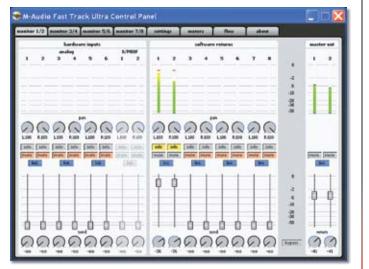
Dual Headphone Amp

Whether you usually track vocals or an ensemble, you're likely to need multiple headphone mixes. The Fast Track Ultra interfaces each have two completely independent headphone outputs so you can take advantage of the DSP mixer's powerful routing options and create the right mix for your performers—complete with reverb and delay.



MX Core DSP Technology for Sophisticated Channel Routing

The Fast Track Ultra interfaces make it easy to create multiple monitor mixes when recording other musicians. The robust MX Core mixer employs an array of eight individual DSP cores to deliver zero latency and a matrix for sophisticated channel routing options—accessible via an intuitive control panel that utilizes a separate mixer for each pair of outputs. The mixer expands the total I/O to an impressive 16 x 8 configuration, allowing you to connect synths and outboard gear to any of the interface's hardware inputs, then mix them with the eight audio streams coming from your computer.



Fast Track Pro

4 x 4 Mobile USB Audio/MIDI Interface with Preamps

Fast Track[®] Pro delivers all the mobile recording flexibility of our popular Fast Track USB, plus even more professional features. You get two front-panel mic/line inputs complete with phantom power for condenser microphones, inserts for outboard effects, balanced and unbalanced analog outputs, S/PDIF digital I/O, MIDI I/O and more. Fast Track Pro also features zero-latency direct hardware monitoring and low-latency ASIO software monitoring, plus an A/B source switch and dual output pairs for DJ-style cueing. Fast Track Pro is compatible with most popular PC and Mac music creation applications including Pro Tools[®] M-Powered™ software. USB connection and bus power along with class-compliancy for Mac OS X* make it a breeze to set up.

onnection	Maximum Speed
SB 1.1	12Mbps
SB 2.0	480Mbps
reWire 400	400Mbps
	SB 1.1 SB 2.0 reWire 400



"... with its all-in-one design, flexible inputs and outputs, and a useful collection of bundled software, the Fast Track Pro gives you everything you need to stay focused on capturing your music."

—Acoustic Guitar



24-bit/96kHz USB audio interface

2 front-panel mic/instrument preamp inputs (combo XLR / 1/4" TRS) with:

- gain controls
- phantom powe
- signal/clip LEDs
- · pads for each input
- insert jack (1/4" TRS) for outboard effects or dynamic processors

2 balanced outputs (1/4" TRS)

4 unbalanced outputs (RCA)

digital S/PDIF (coaxial) I/O w/ 2-channel PCM or pass-through of surround-encoded AC-3 and DTS material

1 x 1 MIDI I/O with activity LEDs

headphone output (1/4" TRS) with level control

headphone A/B source switch for DJ-style cueing

input/playback mix control for hardware direct monitoring

mono switch for input/playback direct monitoring

master output level control

zero-latency hardware direct monitoring

low-latency ASIO software monitoring

powered via USB or optional power adapter

Kensington lock slot

Mac and PC compatible; class-compliant with Mac OS X 10.3.9 and higher*
Pro Tools M-Powered compatible: includes Live Lite

*Class-compliancy supports up to 16-bit/48kHz 2 x 4 operation on Mac OS X 10.3.9 and higher. Driver installation required to access more I/O.



Fast Track USB

Record Guitar and Vocals on Your Computer

Fast Track[®] USB is the easiest way to record with professional results. It has an input for instruments like guitar, bass and keyboards, plus a microphone input for recording vocals or other acoustic sounds. Use it with the included Ableton Live Lite software for PC and Mac or with other popular software including GarageBand. The included GT Player Express software also gives you killer effects.

professional 24-bit/48kHz sound
dynamic microphone input (XLR)
switchable instrument/line input (1/4")
stereo headphone output (1/8")

stereo output jacks (RCA)

level control for headphones and main outputs

USB bus-powered; class-compliant with Windows XP, Vista and Mac OS X

compatible with GarageBand and most other Mac/PC software

direct hardware monitoring for synchronized overdubs

Pro Tools[®] M-Powered[™] compatible; includes Live Lite and GT Player Express software

MobilePre USB

USB Bus-Powered Preamp and Audio Interface

MobilePre USB is the preamp with a built-in audio interface that's designed for laptop work including field recording and sampling expeditions. In fact, it's one of the only completely bus-powered preamps that offers phantom power. (Of course, you can use it on the desktop as well.) A complement of XLR and 1/4" inputs as well as 1/4" line and 1/8" headphone outputs provides maximum

flexibility for any application—including two onboard microphone/instrument preamps and high-impedance instrument inputs ideal for connecting guitars and basses. And unlike many competitors, zero-latency direct monitoring provides you with a streamlined professional recording experience.

2 x 2 16-bit/48kHz analog I/O with preamps

2 mic inputs

2 high-impedance line inputs for guitar, bass, etc.

stereo line outs and headphone out compatible with Macintosh and Windows operating systems USB bus power for total mobility

Pro Tools[®] M-Powered[™] compatible; includes Live Lite



"... a real winner. Its performance far outstripped its laughably low price tag, and we can see many gigging musicians finding it a viable solution to their laptop audio needs." —Electronic Musician

Transit

High-Resolution Mobile Audio Interface

Small enough to fit in your pocket, the bus-powered Transit brings high-resolution 24-bit/96kHz recording and playback to any USB-enabled compatible PC or Mac computer. Digital I/O lets you transfer pristine audio between your computer and other devices such as MiniDisc and DAT. The digital output can also deliver AC-3 and DTS from your computer to an external decoder such as a surround receiver. And the compact design allows you to play and record virtually anywhere your laptop can go.

Pro Tools on the Go

The Transit is the most compact and affordable Pro Tools® M-Powered™ interface available, allowing you to hit the road with the industry standard in audio/MIDI production software. Pro Tools M-Powered has the same user interface as Pro Tools® and Pro Tools LE®, making it easy to transfer your sessions between pro, project and mobile personal studios. While many small-scale mobile devices offer limited functionality and restricted compatibility, Transit provides everything you need for using Pro Tools M-Powered on the go. Now there's no excuse for not having the most popular DAW in your creative toolbox.

"One of my favorite pieces of hardware is the Transit for Pro Tools M-Powered production on the go." —Carmen Rizzo (Grammy-nominated producer, mixer and recording artist; Coldplay, Paul Oakenfold, Alanis Morisette)

mobile 24-bit/96kHz USB audio interface

stereo analog/optical digital input (1/8")

stereo line/headphone output (1/8")

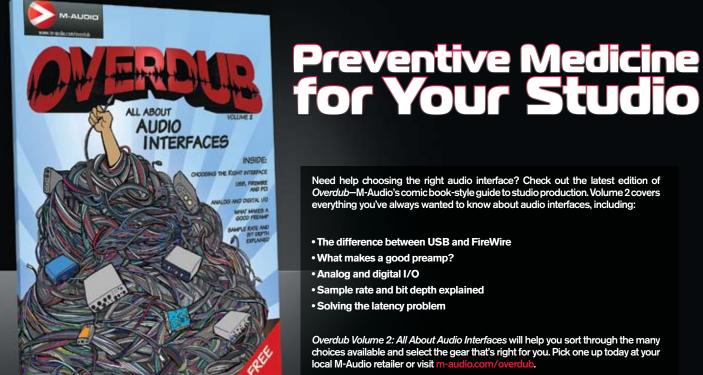
TOSLINK optical digital output allows AC-3 and DTS pass-through accommodates self-powered stereo microphones

single USB connection for input and output

includes 3.5mm (male) to TOSLINK (female) adapter

completely USB powered (no power supply needed)

compatible with Pro Tools M-Powered





Delta 1010



10-In/10-Out Digital Recording System with MIDI and Digital I/O

Recognized around the world as the top choice in host-based PCI digital audio solutions for Mac and PC, the rack-mount Delta 1010 is designed to handle the needs of the most demanding project and professional studios. Superb engineering and ultra-high-quality converters deliver fidelity that surpasses systems costing many times more. Multiple inputs accommodate recording numerous tracks simultaneously, including zero-latency monitoring for seamless overdubs. Multiple outs route individual tracks or soft synth/sampler timbres to a mixing console, or directly drive up to 7.1 surround sound systems. The analog outs also remain live for monitoring while mixing or transferring digital signals via S/PDIF. MIDI and word clock I/O round out everything you need to take advantage of today's music software. Combine up to four 1010s for as much I/O as you need. Compatible with Pro Tools®

Delta-The Standard in PCI Digital Recording

Standard features like 24-bit/96kHz fidelity, zero-latency monitoring and our legendary driver compatibility have made Deltas the best-selling audio card line in the world. Delta cards span a wide range of studio configurations, feature sets, operating systems and price points-and are unparalleled in manufacturing and audio quality at every price.



8 x 8 analog I/O (balanced/unbalanced 1/4" TRS); 1 x 1 MIDI I/O

digital S/PDIF (coaxial) I/O w/ 2-channel PCM or passthrough of surround-encoded AC-3 and DTS material

directly drive up to 7.1 surround (software bass management included)*

word clock I/O for sample-accurate synchronization of external devices

Pro Tools M-Powered compatible

"How does it sound? In a word, stunning. This has to be



Delta 1010LT

10-In/10-Out PCI Virtual Studio

The Delta 1010LT delivers much of the same universal connectivity, high fidelity and seamless performance as the popular Delta 1010 on a single PCI card-and at a fraction of the price. Multiple analog I/O, MIDI, S/PDIF and surround sound support are all here. Two inputs even have mic preamps on XLR connectors, saving the expense of outboard preamps. It's all on a compact, half-size PCI card with two color-coded breakout cables. Combine up to four Deltas in your system for a maximum of 32 analog and four stereo S/PDIF inputs and outputs. Compatible with Pro Tools[®] M-Powered[™] on Mac and PC.





Delta 1010LT analog connector cable



6 RCA inputs, 8 RCA outputs

digital S/PDIF (coaxial) I/O w/ 2-channel PCM or pass-through of surround-encoded AC-3 and DTS material

1 x 1 MIDI I/O

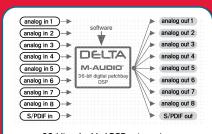
directly drive up to 7.1 surround (software bass management included)**

+4dBu/-10dBV operation individually switched in software

word clock I/O for sample-accurate synchronization of external devices Pro Tools M-Powered compatible

*Inputs switchable to line level.

**PC only.



36-bit embedded DSP puts a mixer inside the Delta 1010

36-Bit Embedded DSP

All Delta cards contain a 36-bit embedded DSP enabling a software-driven patchbay/router for all analog and digital I/O-all with extremely fast A single unified control panel provides settings for clock and sample rates, buffer sizes, individual signal levels for every input and output, adjustable +4dBu/-10dBV level controls and digital I/O control on up to four installed Delta cards. All controls are also easily accessible through mos professional audio software applications fo seamless integration and operation.

Delta Control Panel

The Delta control panel offers full control over features -for up to four Delta card



Audiophile 2496

4-In/4-Out Audio Card with MIDI and Digital I/O

One of the best-selling digital audio cards in the industry, the Audiophile 2496 delivers a level of audio fidelity and performance that's unequalled by other audio cards in its class. This critically acclaimed PCI card features premium digital audio converters, elegant board design and ultra-stable drivers just like the rest of the Delta line, but with a simpler I/O configuration. As a member of the Delta family, the Audiophile 2496 supports both Mac and PC computer platforms and most major software programs, ensuring seamless integration and rocksolid performance. Compatible with Pro Tools® M-Powered™ software.



2 x 2 analog I/O (gold-plated RCA jacks)

S/PDIF (coaxial) I/O w/ 2-channel PCM or pass-through of surround-encoded AC-3 and DTS material

1 x 1 MIDI I/O

Pro Tools M-Powered compatible



Audiophile 192

High-Definition 4-In/4-Out Audio Card with MIDI and Digital I/O

Building on the legacy of the Audiophile 2496-one of the world's most popular audio cards-the Audiophile 192 features high-definition 192kHz resolution, digital I/O, balanced analog I/O and an amazing signal-to-noise ratio. The Audiophile 192 represents a new benchmark in audio performance for music production, mastering and critical listening.

up to 24-bit/192kHz audio

2 x 2 balanced analog I/O on PC and Mac digital S/PDIF (coaxial) I/O w/ 2-channel PCM or pass-through of surround-encoded

direct hardware input monitoring

1 x 1 MIDI I/O

AC-3 and DTS material

Pro Tools® M-Powered™ compatible



Delta 44

Professional 4-In/4-Out Audio Card

Why pay for more than you need? The Delta 44 offers the same features and high performance as the larger Delta interfaces but is designed for the user who does not require digital I/O. The rugged external breakout box gives you the convenience of making connections to the four 1/4" TRS analog inputs and outputs right on your desktop—no more fumbling behind the computer. Compatible with Pro Tools® M-Powered™ software.

4 x 4 analog I/O (1/4" TRS)

balanced/unbalanced operation

professional 24-bit/96kHz audio quality

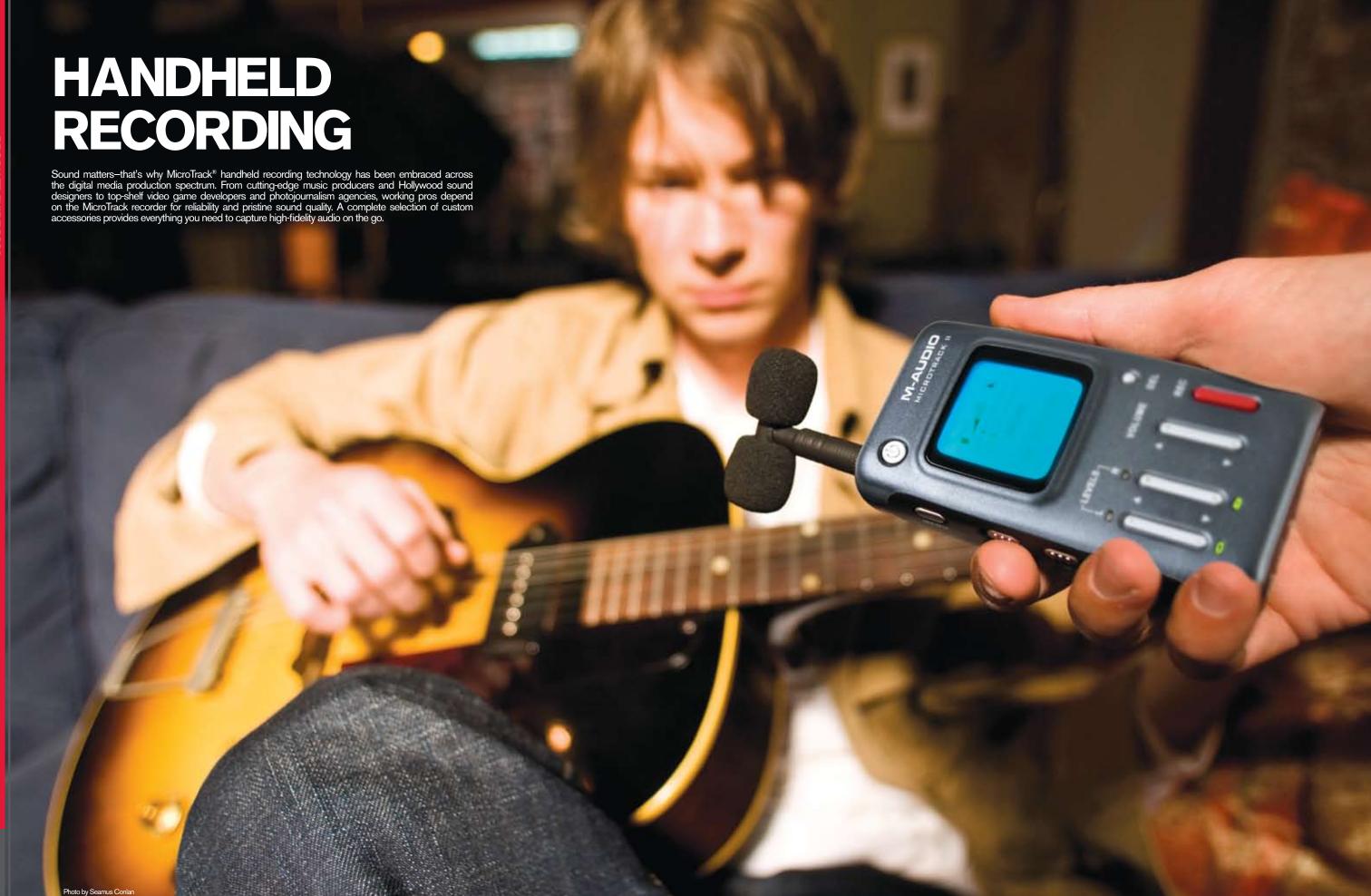
convenient breakout box

Pro Tools M-Powered compatible

M-AUDIO PRODUCT GUIDE 2009

Audio Interface Comparison Chart		Total simultaneous channels of I/O	Analog inputs	Analog outputs	48V phantom power	Resolution	ADAT optical I/O channels	S/PDIF digital	MIDI I/O	Word clock I/O	Headphone output	Connection	Bus- powered	Integrated control surface	Compatible with Pro Tools M-Powered	Hardware direct monitoring	Mic preamps	Drivers	Additional features
:-?e?eeeeeeeee.	ProFire 2626	26 x 26	8 (8 combo XLR-1/4" TRS)	8 (8 1/4" TRS)	yes	24-bit/192kHz*	16 x 16	1 RCA input / 1 RCA output (optical input/ output port B capable of S/PDIF)	1 x 1	yes	yes (2 1/4")	FireWire	no	no	yes (7.4 with downloadable update* and higher)	yes	8	ASIO, Core Audio, WDM, GSIF2	rack-mountable, DSP mixer/router, 8 mic pres, standalone A/D-D/A converter
20 20 00	ProFire 610	6 x 10	4 (2 combo XLR-1/4" TRS, 2 1/4" TRS)	8 (8 1/4" TRS)	yes	24-bit/192kHz*	no	yes (RCA I/O)	1 x 1	no	yes (2 1/4")	FireWire	FireWire (6 pin only)	no	yes (8.0 and higher)	yes	2	ASIO, Core Audio, WDM, GSIF2	DSP mixer/router, 2 mic pres, standalone A/D-D/A converter
• ॐ व्यवसम्बद्धः • ,	ProFire Lightbridge	34 x 36	none	2 (2 1/4" TRS)	no	24-bit/96kHz	32 x 32	yes (RCA I/O)	1 x 1	yes	yes (1/4")	FireWire	FireWire (6 pin only)	no	yes (7.3 and higher)*	yes	no	ASIO, Core Audio, WDM, MME, GSIF2	16-channel Lightpipe I/O at 88.2 or 96kHz using S/MUX
●:ō- •·ċ •·ċ •·	FireWire Solo	4 x 4	2 (1 XLR, 1 inst. 1/4" TS, 2 1/4" TS)	2 (2 1/4" TRS)	yes	24-bit/96kHz	no	yes (RCA I/O)	no	no	yes (1/4")	FireWire	FireWire (6 pin only)	no	yes	yes	1	ASIO, Core Audio, WDM, MME, GSIF2	dual FireWire ports for easy device chaining, software-controlled digital mixing/routing
	ProjectMix I/O	16 x 12	8 (8 XLR, 8 1/4" TRS/ TS, 1 inst. 1/4" TS)	4 (4 1/4" TRS)	yes	24-bit/96kHz	8 x 8	yes (optical I/O, RCA I/O)	2 x 2 (1 x 1 DIN5 MIDI I/O)	yes	yes (2 1/4")	FireWire	no	yes	yes	yes	8	ASIO, Core Audio, WDM, MME, GSIF2	10-bit touch-sensitive motorized faders, LCD display, headphone A/B source switch, transport controls, assignable rotary encoders, support for Mackie Control and HUI protocols, software controlled digital mixing/routing
	NRV10	10 x 10	8 (5 XLR/ TRS, 8 1/4" TS)	4 (2 XLR/ TRS and 2 1/4" TS)	yes	24-bit/96kHz	no	no	no	no	yes (1/4")	FireWire	no	no	yes (7.3 and higher)	yes	5	ASIO, Core Audio, WDM, MME, GSIF2	includes interFX live mixing application with VST support
-: : \$! \$ \$ \$ \$ \$ \$ \$ \$ \$ \$ \$ \$ \$ \$ \$ \$	Fast Track Ultra 8R	8 x 8	8 (8 combo XLR-1/4" TRS)	8 (8 1/4" TRS)	yes	24-bit/96kHz	no	yes (RCA I/O via adapter)	1 x 1 (via adapter)	no	yes (2 1/4")	USB 2.0	USB	no	yes (7.4 with downloadable update and higher)	yes	8	ASIO, Core Audio, WDM, MME, GSIF2	rack-mountable, MX Core DSP mixer with built-in FX
0.0.0.0.0.0.0.0	Fast Track Ultra	8 x 8	6 (2 mic/ line/inst, 2 mic/line, 2 line)	6 (6 1/4" TRS)	yes	24-bit/96kHz	no	yes (RCA I/O)	1 x 1	no	yes (2 1/4")	USB 2.0	USB	no	yes (7.4 with downloadable update and higher)	yes	4	ASIO, Core Audio, WDM, MME, GSIF2	MX Core DSP mixer with built-in FX
	Fast Track Pro	4 x 4	2 (2 combo XLR/ 1/4" TRS/TS)	4 (2 1/4" TRS, 4 RCA)	yes	24-bit/96kHz	no	yes (RCA I/O)	1 x 1	no	yes (1/4")	USB	USB	no	yes	yes	2	ASIO, Core Audio, WDM, MME	class-compliant, headphone A/B source switch
CHORD	Fast Track USB	2 x 2	2 (1 XLR, 1 inst. 1/4" TS)	2 (2 RCA)	no	24-bit/48kHz	no	no	no	no	yes (1/8")	USB	USB	no	yes	yes	1	ASIO, Core Audio, WDM, MME	class-compliant
000 000	MobilePre USB	2 x 2	2 (2 XLR, 2 inst./line 1/4" TRS/ TS, stereo 1/8")	2 (2 1/4" TS, stereo 1/8")	yes	16-bit/48kHz	no	no	no	no	yes (1/8")	USB	USB	no	yes	yes	2	ASIO, Core Audio, WDM, MME	class-compliant
- O Francis	Transit	2 x 2	2 (2 stereo 1/8")	2 (2 stereo 1/8")	no	24-bit/96kHz	no	yes (optical I/O, in via adapter)	no	no	yes (1/8")	USB	USB	no	yes	no	no	ASIO, Core Audio, WDM, MME	accommodates self-powered stereo microphone
the state of the s	Delta 1010	10 x 10	8 (8 1/4" TRS/TS)	8 (8 1/4" TRS/TS)	no	24-bit/96kHz	no	yes (RCA I/O)	1 x 1	yes	no	PCI	n/a	no	yes	yes	no	ASIO, Core Audio, WDM, MME, GSIF2	rack-mountable, software- controlled digital mixing/routing
	Delta 1010LT	10 x 10	8 (8 RCA)	8 (8 RCA)	no	24-bit/96kHz	no	yes (RCA I/O)	1 x 1	yes	no	PCI	n/a	no	yes	yes	2	ASIO, Core Audio, WDM, MME, GSIF2	software-controlled digital mixing/routing
mann Mar	Delta 44	4 x 4	4 (4 1/4" TRS/TS)	4 (4 1/4" TRS/TS)	no	24-bit/96kHz	no	no	no	no	no	PCI	n/a	no	yes	yes	no	ASIO, Core Audio, WDM, MME, GSIF2	software-controlled digital mixing/routing
	Audiophile 192	4 x 4	2 (2 1/4" TRS/TS)	2 (2 1/4" TRS/TS)	no	24-bit/192kHz*	no	yes (RCA I/O)	1 x 1	no	no	PCI	n/a	no	yes	yes	no	ASIO, Core Audio, WDM, MME, GSIF2	direct hardware input monitoring via separate balanced 1/4" TRS monitor outputs, software- controlled digital mixing/routing
	Audiophile 2496	4 x 4	2 (2 RCA)	2 (2 RCA)	no	24-bit/96kHz	no	yes (RCA I/O)	1 x 1	no	no	PCI	n/a	no	yes	yes	no	ASIO, Core Audio, WDM, MME, GSIF2	software-controlled digital mixing/routing

*Maximum 96kHz sample rate and 18x 1











MicroTrack II gives you even more professional recording options than ever before. Capture high-SPL sources without clipping thanks to an extended input gain range and built-in analog input limiter. Transfer files faster than ever before via the high-speed USB 2.0 connection. Drop in markers while recording BWF files and easily find your place when editing your content. Organize and locate your files in the field with a user-created folder structure. Overcome the limitations of FAT32 and create seamless recordings beyond 2GB in size. Dim the backlight to save battery life and be discreet in a dark room. Let the recorder hibernate, then wake it up to record on a moment's notice. A CompactFlash test ensures that the card can record at the settings you desire.

Post MP3s to the Web-Instantly

Since MicroTrack II lets you record directly to MP3 files and transfer them instantly to the computer via high-speed USB 2.0, you can e-mail or post high-quality recordings to the Web immediately. Musicians can e-mail demos to writing/production partners and record companies when the inspiration strikes. Business users can easily upload meetings and presentations for review and wider dissemination. Educators can post classes, lectures and seminars without delay. Faith-based organizations can quickly broadcast worship services and speakers. Professional presenters can create and sell back-end collateral content almost instantly-and the list goes on.

MicroTrack II

The redesigned MicroTrack® II brings even more professional features to the original high-fidelity mobile digital recorder that's been embraced by audio and film professionals worldwide. With an extended input gain range, analog input limiter, 48V phantom power, headphone monitoring of S/PDIF, faster file transfer rate and other enhancements, MicroTrack II delivers the highest quality mobile recording experience available today. Simply record WAV (BWF) and MP3 files to CompactFlash or Microdrives (sold separately) through balanced line inputs or built-in microphone preamps-then drag and drop recordings to your computer via high-speed USB 2.0 for immediate editing or Web posting. MicroTrack II is perfect for a wide variety of applications including field recording, capturing live shows, songwriting,

2-channel WAV (BWF) and MP3 recording and playback for pro recording, capturing live shows, songwriting, training, education, worship and more

new features include extended input gain range, analog input limiter, multi-part recording of files beyond 2GB in size, BWF file marking ability, headphone monitoring of S/PDIF and more

storage via convenient CompactFlash or Microdrives (sold separately)

immediate drag-and-drop file transfer to PC and Mac via high-speed USB 2.0 mini-connector

powered via USB, built-in rechargeable lithium-ion battery or included power supply

separate left and right input level controls with signal and peak indicators

professional balanced 1/4" TRS inputs capable of mic or line-level signals

dual microphone preamps with 48V phantom power for studio microphones

1/8" TRS input with 5V power for use with stereo electret microphone (microphone included)

S/PDIF coaxial input for digital transfers

monitoring via RCA line outputs or 1/8" stereo headphone output

output level control

dedicated buttons for navigation, record, hold, delete, menu and power

includes electret T-shaped microphone, software for editing and file format conversion, carrying pouch, 1/8" stereo extension cable with lapel clip, power supply and USB cable





MicroTrack 10dB Pad

Passive In-line 10dB Pad for MicroTrack 24/96

The MicroTrack 10dB Pad is a specially designed attenuator for the 1/8" microphone input on the first-generation MicroTrack 24/96 recorder—or any device with similar output impedance and phantom power characteristics. It reduces the output of a connected electret condenser microphone by approximately 10dB, which protects the input stage of your MicroTrack 24/96 from becoming overloaded and distorting when recording high-SPL sources. The in-line pad attenuates the audio signal from the mic while passing unaltered the 5V supply needed to power the internal preamp of electret condenser microphones. Patent pending.



5V pass-through for electret condenser mics designed for MicroTrack 24/96 recorder

Like many of today's ultra-compact digital cameras, MicroTrack II records to CompactFlash or Microdrives (sold separately). Capacity is based on the recording format selected and the size of the currently inserted media. For example, a 1GB CompactFlash card or Microdrive will hold approximately 100 minutes of uncompressed CD-quality 16-bit stereo 44.1kHz WAV files. That same 1GB media will store 1500 minutes of speech-quality stereo MP3 audio at 96kbps. Of course, you can use and swap out whatever capacity media suits your needs.

Simple Transfers

You can record as many unique files as the capacity of your available storage allows. Then just connect MicroTrack II to your PC or Mac via high-speed USB 2.0, and it appears as a USB mass storage device showing the unique identification number of each file. Just drag and drop the files to your computer's hard drive and your audio will be ready to edit and/or post to the Web. The included software even gives you the power to easily edit your files and convert between file formats.



Power to Go
The MicroTrack II recorder is powered via a long-life lithium-ion battery. You can record for approximately four to five hours on a single charge (or about three hours with phantom power engaged). You can also use an optional USB battery pack for additional hours of top-quality recording. Recharge the battery by simply connecting MicroTrack II to a PC or Mac computer via USB, or by using the included power supply.



"I use my MicroTrack on practically every episode of Lost. The sound quality is top notch and it's incredibly versatile-whether I'm on the mix stage grabbing an FX or Foley pick up, 'worldizing' effects, or even recording an ADR line. I carry it everywhere I go for recording sounds in the field." -Tom de Gorter (supervising sound editor, Lost)

"Whether you're recording live gigs off the mixing desk, capturing interview material for annotation or simply sampling the the outside world, the MicroTrack II is a quality solution at an even better price than ever-get one." -Future Music

"The compact size of the MicroTrack allows me the luxury of having it with me almost everywhere. It's enabled me to capture recordings that I normally wouldn't be able to get-and at 24/96!" -Tim Larkin (Academy Award-winning sound designer; Mvst. Half-Life 2)

At about the size and weight of a deck of cards, MicroTrack II will go anywhere you need to record. But don't let its small size follows. For example, the same class as our critically acclaimed audio interfaces. Pro-quality preamps complete with full 48V phantom power will work with your favorite. condenser microphones virtually anywhere you need to capture audio-all the way up to 24-bit/96kHz.

When you carry MicroTrack II as your songwriter's notebook, you'll never miss a moment of inspiration. The recorder's mobility and mic preamps also make it ideal for recording practice sessions and gigs, not to mention capturing film-quality sound effects in the field. Balanced 1/4" TRS line inputs also make it easy to take a feed directly from a studio or club mixer. S/PDIF input means that you can even record the output of digital mixers and do transfers from other digital recording/storage devices. And regardless of how you choose to record, you can monitor via the 1/8" stereo headphone jack or RCA line outs.

Optimized for Condenser Microphones

The MicroTrack II delivers optimized performance when coupled with high-quality condenser microphones. The 1/4" TRS inputs allow you to use balanced XLR-to-TRS cables for professional-quality results.*



*For information on using dynamic mics with the MicroTrack II recorder, please visit the MicroTrack II product page on m-audio.com.

MicroPack

Custom Carrying Case for the MicroTrack Recorder

Protect your M-Audio MicroTrack recorder with the M-Audio MicroPack. Constructed from high-quality ballistic nylon, the M-Audio MicroPack is the only all-in-one carrying solution for the MicroTrack that allows you to access every function and connection without having to remove the unit from the bag. And with a full complement of zippered pockets, the MicroPack allows you to keep all of your cables, mics and accessories in one place.



BX8a

M-AUDIO



To a second

Pro Tools M-Powered 8

Production Software

Pro Tools® M-Powered™ 8 software puts the industry standard in audio/MIDI production on the fast track, letting you enter the powerful world of Pro Tools on your own terms. Pro Tools M-Powered has the same user interface as Pro Tools|HD® and Pro Tools LE® software, all while expanding your creative hardware options to dozens of M-Audio interfaces.* Session compatibility with all current versions of Pro Tools software means that you can easily move your projects between pro, project and mobile personal studios. And at just \$299.95 MSRP, anyone can become a member of the Pro Tools community. Pro Tools M-Powered 8 is the serious music production choice for power, compatibility, options and ease of use-without spending serious money.

supports a wide range of M-Audio interfaces* including all current Delta PCI, USB/USB 2.0 and FireWire products
work anywhere, anytime—seamless workflow between studio, stage, home and the road
award-winning Pro Tools mixing environment
industry-standard session compatibility with countless Pro Tools- equipped project and professional studios around the world
48-track** audio recording, editing and mixing up to 24-bit/96kHz
fully non-destructive editing with world-class tools
powerful, intuitive MIDI sequencing
real-time audio and MIDI processing
advanced automation features
import MIDI, REX, ACID, WAV, AIFF, AAC, MP3 and CD audio files
built-in DigiBase file management tool
QuickPunch functionality
powerful loop recording options
low-latency monitoring while recording

STREET, STREET, ST.

RTAS® real-time effects processing

advanced virtual instrument integration

easily accepts ReWire output stream directly from other applications for further mixing and processing

unique Beat Detective™ LE automatic groove analysis and correction tool

supports M-Audio ProjectMix I/O and Digidesign Command|8® control surfaces for hands-on mixing

wide range of compatible creative software options (AudioSuite and RTAS plug-ins, ReWire-compatible applications) available from Digidesign and Digidesign Development Partners

Windows XP, Vista and Mac OS X^ compatible

includes 70 effects plug-ins (EQ, dynamics, delays, reverb and more), plus 7 powerful virtual instruments (6 from the Digidesign Advanced Instrument Research (A.I.R.) group)

*Pro Tools M-Powered requires a supported M-Audio hardware interface to function. Visit m-audio.com for a current list of compatible M-Audio hardware peripherals.

**Up to 64 simultaneous stereo or mono audio tracks with the Music Production Toolkit 2 option. Owners of the original Music Production Toolkit software option who upgrade to Pro Tools M-Powered

8 will automatically get support for up to 64 stereo audio tracks.

*Visit m-audio.com for complete compatibility information

Real Pro Tools

Pro Tools M-Powered is Pro Tools. There are actually three flavors of Pro Tools, all of which share the same user interface and file format. The primary distinction is the hardware they complement. Pro ToolsIHD runs on elite DSP-powered Pro ToolsIHD hardware. Pro Tools LE works with a variety of Digidesign hardware including the Mbox 2 family. And Pro Tools M-Powered* delivers even more options via compatibility with dozens of M-Audio interfaces. And at just \$299.95 MSRP, now there's no reason for not having the industry-standard DAW in your creative toolbox.

For more information on Pro Tools M-Powered, check out Pro Tools on Your Terms (pages 10-13) in the artist section of this magazine.

"I can't imagine anyone who owns a supported M-Audio interface not wanting to acquire the Pro Tools M-Powered software. For the price, I know of no comparable DAW system that combines such high sound quality with recording and editing software of such sophisticated elegance. The combination of M-Audio hardware with Pro Tools M-Powered software can take on anything from singer-songwriter demos, to multitrack live band recordings." -Pro Audio Review

"There's no question that Pro Tools M-Powered lives up to Digidesign's reputation... Now, with Pro Tools M-Powered on the scene, excluding Pro Tools because of budget concerns is past"

Complete Your Pro Tools M-Powered System with Any of the Following Hardware Options:





Fast Track Ultra* High-Speed 8 x 8 USB 2.0 Interface with MX Core DSP Technology

Fast Track Pro 4 x 4 Mobile USB Audio/MIDI

Interface with Preamps

Fast Track USB

Delta 1010 System with MIDI and Digital I/C

Delta 1010LT



Audiophile 192 High-Definition 4-In/4-Out Audio Card with MIDI and Digital I/O



New in Pro Tools M-Powered 8

Get ready for a revolutionary new way to work with Pro Tools[®] software. Pro Tools M-Powered™ 8 delivers a streamlined, customizable interface along with many new production tools and creative options. Work with up to 48 stereo audio tracks*. Create with five new A.I.R. instruments and 30 more plug-in effects. Create sophisticated notation with the new Score Editor based on the Sibelius engine. Work MIDI magic with the new MIDI Editor. Change audio pitch with the new Elastic Pitch real-time pitch transposer. It's time to upgrade to the most powerful version of Pro Tools M-Powered ever.

Stunning New Look and Interactivity

With a sleek new look, Pro Tools M-Powered 8 is as easy on the eyes as it is to use. All of the familiar Pro Tools M-Powered functionality is still in place—now with double the inserts per channel, more customizability, easier access to editing options and more. Customize the toolbar, tile or cascade your window arrangement, change the color of your channel strips, tracks, regions, groups and markers, and much more.



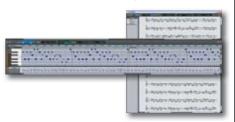
A Well-Stocked Studio

Pro Tools M-Powered 8 comes packed with a huge, comprehensive collection of music creation and sound processing plug-ins-giving you a wellstocked studio right out of the box. Create and play music with groundbreaking new virtual instruments. Dial up awesome guitar tone with Eleven™ Free and SansAmp. Play DJ with Torg® LE. Add character to tracks with 20 new A.I.R. effects. Make music with nearly 8GB of pro-quality loops. And fix, enhance and polish your mixes with ease.



Complete MIDI Production

Pro Tools M-Powered 8 features a comprehensive array of new MIDI tools to streamline production with both virtual and traditional instruments. Gain extensive MIDI editing power through MIDI Editor windows. Work with new features that let you separate, consolidate and mute MIDI notes; scrub and shuttle through parts; view superimposed MIDI and Instrument tracks; and play MIDI notes when tabbing. Edit MIDI automation and continuous controller (CC) data through multiple Automation and Controller lanes, and more.



New Features in Pro Tools M-Powered 8

up to 3 times more audio tracks-now 48 stereo audio tracks* (expandable to 64 tracks)

dedicated MIDI Editor window greatly simplifies and streamlines MIDI sequencing

includes 5 new A.I.R. virtual instruments and FXpansion BFD Lite-7 VIs total

30 more effects plug-ins-70 total

powerful new Score Editor window based on Sibelius notation engine

beautiful, redesigned user interface with new enhancements and customizability

Elastic Pitch real-time pitch transposition

For more information on the new Elastic Pitch and Score Editor features, see pages 12 and 13, in the artist section of this magazine.

incredible new track compositing workflow allows for easily constructing a perfect performance from multiple takes

access up to 10 inserts (plug-ins) per channel

view multiple Automation and Controller lanes for faster, easier editing

improved plug-in parameter mapping allows easier hands-on control of effects and virtual instruments with supported control surfaces and MIDI controllers

automatic Check for Updates feature keeps Pro Tools and plug-ins up to date

QuickStart dialog allows for quickly, easily creating new sessions from templates, from scratch or open existing sessions

improved, reduced-price Music Production Toolkit 2 option includes more valuable plug-in collection and adds support for up to 64 stereo audio tracks

*Up to 64 simultaneous stereo or mono audio tracks with the Music Production Toolkit 2 option. Owners of the original Music Production Toolkit software option who upgrade to Pro Tools M-Powered 8 will automatically get support for up to 64 stereo audio tracks.

Take Pro Tools M-Powered 8 Even Further

Music Production Toolkit 2

FireWire Solo

MobilePre USB

Preamp and Audio Inte

Interface for Songwriters/Guita

Music Production Expansion Option for

The Digidesign $^{\mathbb{R}}$ Music Production Toolkit 2 expands the creative power of your Pro Tools $^{\mathbb{R}}$ M-Powered $^{\mathbb{T}}$ system up to 64 stereo audio tracks plus even more creative tools. It features over \$1,600 worth of professional plug-ins including the Digidesign Eleven™ LE hyper-realistic, vintage guitar amp emulator, Hybrid™ 1.5 high-definition synthesizer, Smackl™ LE compressor and limiter, and Structure® LE advanced sample player, as well as the TL Space™ Native Edition

Producer Factory Bundle

Audio Production Suite for Pro Tools

The Producer Factory™ bundle augments your Pro Tools plug-in collection, offering seven professional Digidesign®/ Bomb Factory® plug-ins-separately worth nearly \$1,100-at incredible savings. Includes Cosmonaut Voice™; JOEMEEK SC2 Photo Optical Compressor and VC5 Meequalizer®; and the Moogerfooger 12-Stage Phaser, Analog Delay, Lowpass Filter and Ring Modulator.

Producer Factory Pro Bundle

Professional Audio Production Suite for

The Digidesign® Producer Factory™ Pro bundle features 17 Digidesign®/Bornb Factory®/Trillium Lane Labs® effects plug-ins is worth over \$3,550–yours for a fraction of the price. The bundle includes all seven plug-ins found in the Producer Factory bundle, plus the Bomb Factory BF-2A and BF-3A; Fairchild[®] 660 and 670; Purple Audio MC77; SoundReplacer™; TL EveryPhase™; Tel-Ray[®] Variable Delay; and Voce Chorus/Vibrato and Spin



NRV10^

10 x 10 FireWire Digital Audio Interface | 8 x 2 Analog Mixe



Tora Xponent dvanced DJ Performance/Production System ProjectMix I/O Control Surface with Motorized Faders and 18 x 14 Audio Interface

sit m-audio.com for the complete list of hardware impatible with Pro Tools M-Powered. ompatible with Pro Tools M-Powered 7.4 with

High-Definition Synthesizer for Pro Tools Systems

Developed by the Digidesign® Advanced Instrument Research (A.I.R.) group, Hybrid™ is a high-definition RTAS® software synthesizer for Pro Tools® M-Powered™ and other Pro Tools® systems that combines the warmth of classic analog waveforms with digital wavetables. Emulate your favorite classic synth or create something no one has ever heard before.

> 256 preset patches combining analog and digital sounds user-adjustable parameters to easily create unique sounds sample-accurate, high-definition synth engine two simultaneous parts for complex patches RTAS format for tight Pro Tools system integration

"A lot of sound design work went into giving Hybrid the kind of basses, pads and keyboard comps that mainstream pop music producers are sure to love... Hybrid is simply a great synthesizer, capable of both classic analog warmth and modern digital edge." -Mix

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Strike

Virtual Drummer Instrument for Pro Tools Systems

Strike™ from the Digidesign® Advanced Instrument Research group is a revolutionary RTAS® instrument plug-in that makes it easy to create, control and produce professional drum performances in Pro Tools® systems with uncanny realism. One part intelligent virtual drummer and one part professional drum module, Strike features a wide range of high-definition drum kits and style choices, and allows for unprecedented control over the virtual drummer's performance. Comprising 20GB of 24-bit source material, Strike allows you to mix and match kits, room tones, effects and playing styles, as well as explore the full dynamic range of a professionally tuned and miked drum kit. And unlike other virtual drummer instruments, Strike is designed expressly for the Pro Tools environment and delivers a level of stability, efficiency and ease of use that no third-party product can touch.



"Using Strike is not unlike sitting behind the glass in a studio and recording a real trap-set drummer-but with some extras that session producers can only dream of... Strike is a super addition to the Pro Tools family, and is sure to be showing up before long on major-label releases near you." -Electronic Musician

easily produce professional, highly realistic drum performances within Pro Tools systems full, real-time performance control, including intensity, complexity, timing, groove, dynamics and much more groundbreaking new real-time performance engine allows for gradually adjusting dynamics

doubles as a powerful MIDI drum module for use with your own drum patterns and sequences

5 high-definition drum kits, each featuring up to 12 instruments and up to 300 samples per instrument



"[Velvet's] user interface is a breeze to navigate, and the onboard signal processing and effects are perfect for spicing up your sounds. Its presets are so convincing, you'd swear you're hearing the real thing. The raw pianos are modeled to perfection, and the ability to add the idiosyncrasies of the originals gives Velvet a huge advantage." -Electronic Musician

Velvet

Virtual Electric Piano for Pro Tools Systems

Developed by the Digidesign Advanced Instrument Research group, Velvet® is a worldclass virtual instrument that delivers highly realistic emulations of four of the most legendary electric pianos. Using dynamic modeling-a revolutionary proprietary combination of both sampling and modeling techniques-Velvet emulates these original electric pianos, both in sound quality and playing feel. Velvet also includes built-in preamp/EQ and multi-effects sections, making it perfect for reproducing or customizing any kind of electric piano sounds-from classic to contemporary. Available in RTAS® plug-in format for Pro Tools® and Pro Tools M-Powered™ 7 and higher.

highly accurate emulations of four legendary electric pianos unique dynamic modeling technology combines sampling and modeling over 100 presets plus powerful preamp section and multi-effects highly optimized RTAS instrument for Pro Tools and Pro Tools M-Powered



Professional Sampler for Pro Tools Systems

Developed by the Digidesign® Advanced Instrument Research (A.I.R.) group and optimized exclusively for Pro Tools® systems, Structure™ is a powerful RTAS® virtual instrument that redefines the art of sampling. Freely create, sculpt and refine everything from simple acoustic instruments to highly complex soundscapes with amazing ease-in real time.

integrated stereo and surround multi-effects processing engine

professional RTAS sampler designed and optimized for Pro Tools systems includes comprehensive, premium factory sound libraries (nearly 20GB) compatible with unencrypted SampleCell, Kontakt 1 and 2, and EXS24 sound libraries powerful database and integrated file browser



"... a solid, simple, unintimidating, easy-to-use, easy-to-understand, great-sounding sampler that integrates with Pro Tools and comes with a phenomenal library... It speaks volumes that something I've only had for a couple of weeks has already barged its way to the front of my plug-in queue as a one-stop shop for all my music requirements..." -Sound On Sound



"[Transfuser's] really ill... It's just bananas. It's kinda blowing my mind right now." -Junior Sanchez (artist, remixer, producer; Madonna, Gorillaz, New Order

"[Transfuser's] just amazing... really incredible! It looks infinitely expandable... you can get really deep into it" -Ken Jordan (The Crystal Method)

Transfuser

Loop and Groove Creation Workstation for Pro Tools Systems

Transfuser™ is an innovative, real-time loop, phrase and groove creation workstation that allows you to quickly and easily create, manipulate, and perform loop- and rhythm-based music on the fly-right within Pro Tools® M-Powered™ and other Pro Tools® systems. Whether you're looking to create and tweak your own rhythms, mash up something funky, or breathe new life into existing loop libraries, Transfuser offers a ton of unique tools that'll have you creating, exploring and experimenting with music and loops in brand new ways.

easily create, tweak, arrange and perform grooves on the fly right within Pro Tools systems

musically intelligent randomization lets you instantly explore inspiring groove variations at the click of a button

supports full drag and drop of Pro Tools regions and audio files from

comes with nearly 2GB of loops and phrases to get you started includes 20 real-time effects ideal for electronic, hip-hop and dance music creation



Eleven LE

Hyper-Realistic Guitar Amp Emulations for Pro Tools Systems

Eleven LE™ is a powerful RTAS®/AudioSuite™ plug-in that sets a new standard for recording guitar amp sounds. Eleven enables you to achieve highly realistic, jaw-dropping guitar tones based on the world's most coveted vintage and modern tube amps, speaker cabinets and mics-all within your Pro Tools® M-Powered™ rig.

Meticulously Modeled Classic Amps

A great guitar amp doesn't just amplify your sound-it can breathe life into a simple riff or chord change, or viciously attack when it's time to solo. The Eleven team went to great lengths to find the most lusted-after vintage and modern amps, speaker cabinets and mics-in their original form-to create a plug-in so realistic that you can practically feel the tubes burn.

"I've never found anything that simulates a speaker cabinet like this before. It's really got the bite and the attack that's missing from most plug-ins, and it's as responsive to my playing as a real tube amp." -Dave Navarro (artist/guitarist; Red Hot Chili Peppers, Jane's Addiction)

create realistic mic'd guitar amp/cabinet sounds within Pro Tools M-Powered or other Pro Tools systems

choose from vintage and modern amp, cabinet and mic models including presets multi-dimensional tones complete with power amp sag, ghost notes, cabinet resonance and speaker cone breakup

innovative amp cloning technique emulates each component to behave like its hardware counterpart

convolution-based speaker cabinet and mic models provide incredibly rich tones





"The behavior of the TimewARP 2600 software-both module by module and integrated into patches-is effectively indistinguishable from that of the analog hardware that it emulates." -Jim Michmerhuizen (director/ founder, Boston School of Electronic Music; author, ARP 2600 and TimewARP 2600 owners manuals)

"The instrument is remarkably authentic and delivers a broad palette of unique and evocative sounds that just plain feel different than most virtual synths out there today... It's a spot-on emulation of an extraordinarily deep and complex instrument that puts all of the charm and character of the classic at your fingertips." -Future Music

"This is one great application. It takes the sound of the ARP 2600 and makes it available to musicians for about a tenth of what the 2600 cost new... This is the most impressive synthesizer emulation I have heard to date." -Recording

"The TimewARP 2600 has all the charm, the complexity and magic of the original but with none of the drawbacks."

KikAxxe

Virtual Vintage Analog Synthesizer

The KikAxxe is a faithful emulation of the classic ARP Axxe* synthesizerand much more. Built with Way Out Ware's award-winning analogmodeling technology, KikAxxe combines the Axxe synthesizer, analog-style 16-step sequencer, programmable drum machine and a retro-themed delay/echo effect. Chock full of modern conveniences like presets, MIDI sync and comprehensive MIDI-mapping capabilities, KikAxxe has all of the features you'd expect from a top-shelf virtual instrument. KikAxxe is compatible with RTAS®, VST and AU hosts, making it a welcome addition to any personal studio.

meticulous emulation of the classic ARP Axxe*

16-step analog-style sequencer

sample-based drum machine with 5 kits and integrated sequencer analog-style delay/echo effect

synth and drum machine can be assigned



TimewARP 2600

Virtual Vintage Voltage-Controlled Synthesizer

The TimewARP 2600 from Way Out Ware is a meticulous emulation of the classic ARP 2600* analog synthesizer-and the only one endorsed by the original instrument's creator, Alan R. Pearlman. In addition to faithful replication of every original feature, TimewARP 2600 features up to eight-voice polyphony and extremely sophisticated velocity and aftertouch control. Mapping of MIDI controller parameters to sliders, knobs and switches is virtually unlimited, putting an unprecedented amount of control at your fingertips. And even though you can create incredibly complex sounds with TimewARP 2600, the unit's comprehensive preset manager ships with hundreds of patches to get you started. Standalone or RTAS®, VST and AU host operation on Mac and Windows makes this gem the ideal addition to your sound-whether as a synthesizer or a unique signal processor.

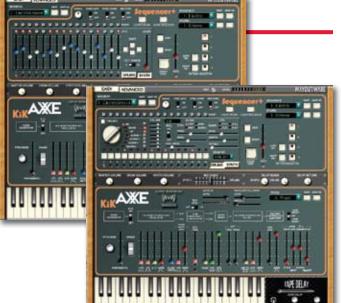
faithful emulation of all original ARP 2600 features and sound* extensive control over velocity and aftertouch unlimited MIDI assignment of all controls to MIDI controllers responds to MIDI bank- and patch-select commands up to 8-voice polyphony LFO synchronization to MIDI Beat Clock

instant access to complex sounds via hundreds of great factory presets extensive patch manager with sort, import and export capabilities

incredibly smooth controls in both operation and sound

standalone or RTAS/VST/AU operation on Mac and Windows









Complete Masterina System

When you're ready to master your mix, Ozone is all you need. Ozone 4 combines seven essential mastering processors into one complete system, letting you give your mixes a finished, full and professional sound. Ozone's critically acclaimed IRC™ (Intelligent Release Control) Loudness Maximizer gives you commercial volume and fullness without distorting or coloring your mix. An eight-band Paragraphic EQ combines linear phase precision with the warmth and character of analog equalizers and even lets you intelligently match your mix to reference tracks. Innovative Multiband Dynamics, Harmonic Exciter and Stereo Imaging tools add depth and polish to your projects. And because these modules are integrated in a single powerful interface, Ozone maintains the highest quality processing throughout the entire signal chain-while also making the mastering process more efficient and intuitive than ever before.



complete mastering system in a single plug-in
includes 7 essential mastering tools: Loudness Maximizer,
Paragraphic EQ, Multiband Dynamics, Multiband Stereo
Imaging, Harmonic Exciter, Dithering, Mastering Reverb
analog-modeled processing combined with linear phase precision
elegant, highly efficient user interface
extensive metering and spectrum analysis tools
comprehensive preset system with over 100 professionally designed presets

integrated undo history with comparison tools user-definable signal routing

extensive automation support

CPU-efficient and optimized for today's low-latency DAW environments

64-bit internal processing chain

supports sampling rates up to 192kHz

excellent documentation including a comprehensive mastering guide that helps demystify the process



iZotope RX

Complete Audio Restoration Software

iZotope RX is the most complete audio restoration product on the market-a unique application designed from the ground up to combat a range of audio problems. Featuring both standalone and plug-in operation, RX includes innovative restoration modules, detailed analysis tools and time-saving workflow features that help you get the most out of your recordings. RX is ideal for audio restoration and archiving, recording and mastering, broadcasting and podcasting, video production, forensics, and any application that demands spotless results and a truly complete range of restoration tools.

"It has been difficult to limit the word count on this review. I could write about the numerous features buried in the software. But I have to go put my Waves Restoration Bundle on eBay."

"The sound quality is top-notch, but RX is surprisingly affordableyou need never sacrifice that near-perfect take again. -Alexander Posell, XLR8R

"... even with heavily clipped material, RX can produce results that will have you wondering whether the audio had ever been clipped in the first place... RX is a thoroughly excellent solution for sound restoration, not to mention an unusual creative tool. Top marks! -Computer Music

complete software solution designed for audio restoration

features both standalone and plug-in operation

ideal for recording and music production, broadcasting and podcasting, restoration and archiving, video production and forensics

suppress broadband and tonal noise without the harsh artifacts of other solutions

remove intermittent noises, corrupted intervals and gaps with Spectral Repair re-synthesis

eliminate overload distortion by automatically rebuilding clipped sections of audio

clean up hum and buzz caused by poor wiring and other electrical problems

remove impulse noises like clicks, crackles and pops, digital artifacts and more select by time and frequency to isolate and repair noise with surgical precision

process multiple settings in parallel and revert to earlier settings with an undo

"Absolutely amazing!" -Sound On Sound

Hum Removal

RX Hum Removal uses precise filtering to isolate electrical hum as well as any resulting harmonics. In addition to 50Hz and 60Hz hum removal, a Free mode allows for hum removal at other frequencies.

The RX Declipper can save previously unusable recordings by removing analog and digital overload distortion. Declipper uses advanced multiband processing to actually rebuild damaged peaks in clipped audio.

Spectral Repair

RX Spectral Repair can remove hard-to-fix intermittent noises and even intelligently fill in gaps in an audio file. Spectral Repair can learn from the audio surrounding a selection and resynthesize missing audio, even if the content includes changes in pitch, vibrato and harmonics.

Advanced Spectrogram

The RX software's unique spectrogram display can represent audio with an unparalleled level of detail and sharpness. The spectrogram uses newly developed technology to attain the greatest time and frequency resolution available. Use the spectrogram display to identify clipping, buzz, intermittent noises and other problems at a quick glance.

Declicker

The RX Declicker can automatically suppress clicks, crackles and pops as well as similar digital impulse noises. Ideal for cleaning up audio archived from vinyl, cylinders and record masters, Declicker also features a manual mode for isolating and repairing hard-to-fix clicks and pops.

The RX Denoiser is designed to remove broadband and steady-state noise including tape hiss, environmental background noise, camera motor noise, buzz and more. Denoiser can be trained with a sample of noise, or it can run in automatic mode. Denoiser uses special techniques based on image processing technology to achieve natural-sounding results with a minimum of artifacts.

RX Restoration Plug-ins

Each module from the RX standalone application has been faithfully reproduced in plug-in form, allowing you to use cutting-edge restoration tools in conjunction with your digital audio workstation. The individual plug-ins include Hum Removal, Declicker, Declipper, Denoiser and Spectral Repair-offering compatibility with VST, AU, MAS, RTAS, AudioSuite and DirectX formats.



Complete Distortion Processor

iZotope Trash is the ultimate tool for selectively adding distortion to your tracks. Of course it features tons of analog-modeling guitar rig simulators, including rectified overdrive and a stunning array of 85 cabinet models. And you'll find just as much use for the subtler tone enhancements to your other tracks-tape saturation for vocals, overdrive for keyboards or a pinch of fuzz on a lead guitar, for example. Chain pairs of distortions together or apply distortion independently to individual frequency bands. Trash is compatible with most Mac and PC hosts, supports 192kHz and employs 64-bit processing for great fidelity regardless of how nasty you like your sound.





64-bit internal processing up to 192kHz

realistic guitar rig simulator

85 speaker cabinet models from vintage to experimental; 3 microphone models

Multiband, Dual-Stage Distortion

Have you used all the "standard" distortion plug-ins out there, only to find that you still end up with a thin, canned and basically unrealistic sound? Do you wish you could combine the ease of use of the "virtual stack" with the intelligent control of studio effect processors? Trash combines multiband, dual-stage distortion and an intuitive user interface to prove that all distortions are not created equal. Add a little crunch in the midrange without muddying the lows and over-hyping the highs, give the low end some more thump without distorting the top. Go one step beyond standard formats and layer not one, but two types of distortion on each of the four available frequency bands.

36 sweepable filter types-analog, resonant, clean, saturated

multiband, distortion, compression and gating

Turn It Up to 11

We'll give it to you straight: Trash is a killer amp simulator. Now you can get that real overdriven amp sound and still keep the peace with your neighbors. With 85 box models to choose from, you'll never run out of options to get the sound you've been searching for. In addition, we went beyond the normal boxes you find in the music stores. You'll find models for telephones, clock radios, sheet metal, plexiglass tubes, piano cabinetsand those are just a few of the extra models included with Trash. To get the creative juices flowing, we also created a new class of cabinets that don't exist in the material world. When it comes to amp modeling, Trash can be as real or unreal as you want.

extensive delay modeling-analog, lo-fi, tape-style CPU-efficient for Mac and PC

Pro Tools[®] M-Powered[™] compatible

Beyond Distortion

In addition to amp modeling. Trash has 36 sweepable filters and six delays to choose from. The pre-filter lets you add a tremolo, wah, or any number of more exotic effects to your sound. The buzz module gives you six different types of delay lines ranging from a clean, crisp digital delay to a gritty, organic tube-amplified tape delay. In addition to all the controls you would expect, a "trash" slider lets you adjust the amount of distortion applied to the delayed signal and an output filter lets you shape the wet signal even more.

"With no cabinet selected, I got some of the nastiest sounds I'd ever heard. We are talking ear-ripping, fried-capacitor, broken-speaker, overdriven digital recorder-type distortions-cool!" -Guitar One

"I love Trash. It's all over Bleed Like Me." -Butch Vig (producer/musician; Garbage, Nirvana)

"Whether you want tape-like saturation to warm up vocal tracks or extremely saturated distortion that sounds like the Judas Priest Marshall backline melting into a nuclear reactor, Trash provides just about any flavor or distortion you can imagine."

"Ozone, Spectron and Trash totally rock."

-Jonathan Davis (singer; Kom)



Spectron

Spectron from iZotope is the next generation of spectral effect plug-ins, allowing unmatched control over audio by frequency. Add delay to only the snare or pan only the hi-hat-on a drum track that's already mixed. Give your guitar a delay, but only on the high notes. Make your bass ring down low, but stay clear and dry up high. For effects that are subtle or out of this world, Spectron is the one-stop spectral toolbox. Spectron splits audio into thousands of frequency bands, applies independent delay, morph, filter and pan effects, and then resynthesizes the processed audio into an entirely new sound. Using this unique control over individual frequencies, Spectron can sound like a phaser, flanger, chorus, ensemble, vocoder or an entirely new class of effects. Spectron is perfect for spots, loop/sample manipulation, sound design and music production.

up to 192kHz sampling rate spectral domain processing morphing for vocoder-style effects delay with independent feedback control

4-stage chorus/flanger effects

fully automatable, triggered filters

frequency-selective panning

CPU-efficient for Mac and PC Pro Tools[®] M-Powered[™] compatible

Spectron's Secret-Spectral Nodes

Even though Spectron's approach to processing is unique, the Spectron interface relies on standard controls. The Spectron effects are manipulated by Spectral Nodes, which behave like the nodes of a parametric EQ with control over frequency, amount, bandwidth and shape. Beyond the basic node operation, you can use Spectron Nodes with LFOs and envelope triggers to open another level of creativity. The frequency, amount, shape and width automate with advanced control over speed and patterns not possible with normal automation.

Looking for a New Sound?

Spectron was designed with creativity and originality in mind. With Spectron, you can easily create effects that aren't possible with analog hardware, thus inspiring new ideas for creating intricate and evolving soundscapes. Looking for something a little more subtle? It's just as easy to use Spectron to add coloration with flanges, spatialization or light filter sweeping.

Hundreds of Effects in an Instant

Looking for a crazy effect right away? Check out Spectron's integrated preset system. Straight out of the box, you get tons of quality presets and the list is constantly growing. With the presets, Spectron is a virtual stomp box for guitars, keys, bass, vocals and anything else that needs new aural life. Experienced users will find the presets to be very useful starting points for creating their own sounds with Spectron. Be sure to check the iZotope web site for new presets.

"[Spectron's] ability to split audio into thousands of frequency bands and then independently apply effects to these bands is amazing. I love it in particular for delays, filters and panning for creating some great sounds and effects?

-Tania Mann (DJ/producer/engineer)



Virtual String Machine

Virtual Vintage String Synthesizer Collection

The Virtual String Machine from GForce puts the evocative tones of the finest retro string synthesizers of yesteryear at your fingertips. Far more than a preset device, Virtual String Machine makes it possible to layer two sample sets and apply an intuitive synthesis engine to eachcreating different envelopes, filter settings, pitch LFO, pan positions, fine tune amounts and more. You can even apply a vintage-style phaser and/or ensemble to the final patch. The result is a truly authentic and highly versatile range of textures, perfect for just about every musical genre.

features over 3200 individual samples from over 60 sample sets

includes emulations of Freeman String Symphonizer, Eminent 310, ARP Omni II, ARP Quartet, Crumar Multiman, Polymoog, Elka Rhapsody, Korg PE2000, Logan String Melody, Eminent Solina, Roland RS202, Yamaha SS30 and more*

fully programmable with 500+ factory patches

standalone or host operation

Pro Tools® M-Powered™ compatible



"GForce aim their hammer with unerring accuracy in realizing Virtual String Machine... VSM hits the nail squarely on the head, combining the immediacy of the original sampled instruments with some well chosen sound shaping tools that provide plenty of scope for creativity while still retaining the spirit of what string machines were all about"

"It's quickly apparent that the sky's truly the limit for sonic choices, flexibility, and potential combinations

"Virtual String Machine sounds wonderful and is a joy to use... There's plenty of opportunity to be a purist and stick with single instruments and plenty more to 'mess with history' and create genera spanning hybrid sounds," -Future Music







In the 1970s-well before polyphonic synthesis and digital sampling was commonplace-dozens of string machines evolved to emulate sections of violins. They changed the face of music in the hands of Pink Floyd, Stevie Wonder, The Cure, Parliament, Herbie Hancock, Air, Jov Division, Jean Michel Jarre, Thomas Dolby, Genesis, Lonnie Liston Smith and many others.* More recently, these vintage instruments have seen resurgence with artists like Kasabian and Goldfrapp. The Virtual String Machine puts those same sounds and much more right in your Mac or PC.

Oddity

Virtual Vintage Analog Synthesizer

The Oddity from GForce is a meticulous emulation of the classic ARP Odyssey synthesizer popularized by artists as diverse as Herbie Hancock, Styx, Tangerine Dream, Ultravox and Portishead.* This legendary dual-oscillator, duophonic vintage synth is back in all its glory, crafted with incredible realism. Modern improvements have been added including programmability, preset morphing, host-syncable LFO and automation along with velocity control of the amp and filter envelopes. You can use the Oddity in standalone mode (Mac/PC) or as a plug-in for most popular host applications. You simply can't get this sound anywhere else today.



... a remarkable imitation of a classic synth. Given that Odysseys are rare, expensive unreliable, often have damaged faders, always have scratchy faders, often drift out of tune, and are always a nightmare to tune... I very STRONGLY recommend you to try it" -Gordon Reid, Sound On Sound

"You guys did a fantastic job of getting every detail right, including incorporating some features that we couldn't have even dreamed of at the time. Having [an Odyssev] completely emulated in software is really the ultimate! Thanks for this wonderful labor of love." -David Friend (co-founder ARP Instruments and lead designer of the original ARP Odyssey)

"... an aficionado's plug-in... a product for synth connoisseurs... a labor of love... somebody put a lot of time into getting the details right." -Craig Anderton, EQ









extremely realistic analog modeling of classic ARP Odyssey* including:

- 2 syncable oscillators
- fully tunable across a 6-octave range
- sawtooth, sine, square and variable pulse-width waveforms and sync
- white and pink noise generators · monophonic and duophonic modes
- ring modulator
- resonant 24dB/octave low-pass filter
- · high-pass filter
- sample-and-hold
- · portamento
- 2 envelope generators • flexible modulation routing

host-syncable LFO

programmable including 384 presets timed morphing between presets

Flying Slider feature

full automation support

standalone or host operation

Pro Tools® M-Powered™ compatible

*M-Audio is a trademark of Avid Technology, Inc. All other product names are trademarks of their respective owners, which are in no way associated or affiliated with Avid Technology, Inc. They are used solely to identify the products of those manufacturers whose tones and sounds were studied during GForce's sound model development. References to artists and bands are for informational purposes only and do not imply endorsement or sponsorship of these products by these artists and bands.



"... totally authentic... I use [M-Tron] both in the studio and live." -Rick Wakeman (keyboardist: Yes)

"If you're into retro sounds, the M-Tron is a must" -Paul White, Sound On Sound

M-Tron Pro



Virtual Vintage Keyboard

the unique tape-playback keyboard used by such luminaries as The Beatles, Yes, David Bowie, Led Zeppelin, The Moody Blues and Radiohead* The 2.5GB sound library features over 160 tape banks (sample sets) including 19 of the original GForce M-Tron tape banks, which have been remastered at London's legendary Abbey Road Studios. M-Tron Pro also delivers 45 brand-new tape banks and over 500 patches, many created by world-class recording artists and programmers. Select from the patches or open the lid to access G:sampler's easy editing controls for layers, splits, reverse, half-speed and more. Use it in standalone mode (Mac/PC) or as a plug-in for most popular host applications.

incredible emulation of vintage Mellotron-and beyond*

160+ tape banks including remasters, loops and all-new libraries

500+ presets featuring many from acclaimed artists and programmers new layer, split, synth and editing features for custom sounds

standalone or plug-in operation

Pro Tools® M-Powered™ compatible

Minimonsta:Melohman

Virtual Vintage Analog Synthesizer

The Minimonsta: Melohman from GForce is an expert emulation of the classic Minimoog that was at the forefront of the analog synth revolution and made famous by artists like Jan Hammer, Rick Wakeman, Keith Emerson, Chick Corea, Gary Numan and Kraftwerk* Every component of the triple-oscillator vintage synth has been faithfully modeled with stunning realism-and then some. Minimonsta is fully programmable, ships with over 6000 presets and can dynamically morph between 12 patches within a meta-patch bank for unprecedented control. There's also an additional LFO and ADSR that can be applied to just about any parameter, as well as delay. You can even play Minimonsta in monophonic, polyphonic or unison modes. This true players' instrument runs as a standalone application (Mac/PC) or as a plug-in for most popular host applications.



all original Minimoog* features:

- 3 VCOs
- pink/white noise generator
- famous Moog 4-pole 24dB/octave low-pass VCF
- 2 ADSRs external input
- mixer • alide
- additional LFO and ADSR for matrix modulation of almost
- monophonic, polyphonic, legato and unison trigger modes

delay effect

fully programmable with over 6000 factory patches real-time morphing between up to 12 patches via keyboard control external input allows filter to be

used as plug-in effect MIDI Learn with storable

Continuous Controller maps

standalone or host operation

full automation support Pro Tools® M-Powered™ compatible "There's simply nothing out there that comes close." -Rick Wakeman (keyboardist; Yes)

"I closed my eyes and could not believe it was not the real thing." -Steve Winwood (singer/songwriter multi-instrumentalist; Traffic, Blind Faith)

"The ability to sculpt the sound using both hands makes the Minimonsta a real musical instrument, rather than just a programming tool." -Will Gregory (keyboardist; Goldfrapp)

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"... an extremely powerful soft synth-one of the most powerful, honestly, that I've ever come across. A brilliantly designed, greatsounding virtual OSCar, Buy it today," -Future Music

Virtual Vintage Analog Synthesizer

The impOSCar from GForce is a faithful reproduction of the classic OSCar synthesizer used by artists including Stevie Wonder, Billie Currie (Ultravox), BT, Keith Emerson and Geoff Downes (Yes and Asia)* Unique for its digital oscillators with Additive Wave Matrix, dual filters and onboard sequencer, this legendary vintage synth has been faithfully reproduced to the last detail-along with some new innovations such as additional LFO and filter modes, polyphony, user-definable waveforms, effects and over 500 designer presets. Use it in standalone mode or as a plug-in for most popular host applications.

extremely faithful analog modeling 2 digitally controlled oscillators, 13 waveforms user-programmable Additive Wave Matrix

mono, duo and 4 polyphonic arpeggiator modes 5 keyboard triggering modes

6 portamento modes

9 filter types with overdrive, cutoff, Q and separation

2 envelope generators extensive LFO section with sync

powerful effects section programmable velocity response

MIDI Learn with storable Continuous Controller maps 500+ designer patch library

full automation support

standalone or host operation

Pro Tools[®] M-Powered[™] compatible

*M-Audio is a trademark of Avid Technology, Inc. All other product names are trademarks of their respective owners, which are in no way associated or affiliated with Avid Technology, Inc. They are used solely to identify the products of those manufacturers whose tones and sounds were studied during GForce's sound model development. References to artists and bands are for informational purposes only and do not imply endorsement or sponsorship of these products by these artists and bands.



Axiom Pro 61

Advanced 61-Key USB MIDI Controller with HyperControl Technology



- 9 sliders
- 9 buttons
- 8 rotary encoders
- 8 velocity-sensitive pads
- 12 keypad-style buttons
- 6 transport buttons
- 7 function keys

Axiom Pro 49

Advanced 49-Key USB MIDI Controller with HyperControl Technology



- 9 sliders
- 9 buttons
- 8 rotary encoders
- 8 velocity-sensitive pads
- 12 keypad-style buttons
- 6 transport buttons
- 7 function keys

Axiom Pro 25

Advanced 25-Key USB MIDI Controller with HyperControl Technology



- 8 rotary encoders
- 8 velocity-sensitive pads
- 6 transport buttons
- 4 function keys

designed for **PRO** TOOLS

TruTouch semi-weighted, piano-style, professional-action keybed

proprietary HyperControl MIDI mapping and host control automatically maps controllers to popular DAWs and software instruments-and maintains a constant live link between them

compatible with DAW software including Pro Tools, Cubase, Logic and Reason systems*

graphic LCD groups and names controls modularly with 4 dedicated profiles per group for easy recall

dedicated button toggles between Mixer and Instrument control modes

Transport-only mode option enables DAW transport control and track navigation combined with user MIDI controller mappings

ASCII messages for DAW keyboard shortcuts

50 onboard memory locations

USB bus-powered

MIDI Out and MIDI In port for extended external connectivity

4 programmable keyboard zones

Axiom Pro MIDI Controllers

M-Audio® Axiom® Pro USB MIDI keyboard controllers build on the acclaimed Axiom series with premium TruTouch™ action, proprietary HyperControl™ MIDI mapping technology and other cutting-edge enhancements. HyperControl automatically maps the keyboard's controls to commonly accessed parameters in DAWs including Pro Tools®, Cubase, Logic and Reason,* as well as software instruments. The constant two-way link with your host DAW means the keyboard's controls are always in sync with your software's active parameters. The intuitive graphic LCD constantly updates the current values, which ensures seamless editing and prevents parameter jumps—even when plug-ins are closed. Toggle instantly between Mixer and Instrument control modes. Map buttons to send QWERTY key commands right from the Axiom Pro. Save settings to 50 memory locations—each with four profiles of quick recall via the intuitive graphic LCD. Now you can control your entire session right from your Axiom Pro.

Enhanced DAW Integration with HyperControl Technology

Thanks to revolutionary M-Audio HyperControl technology, your Axiom Pro will automatically map to commonly used parameters of popular software instruments and digital audio workstations including Pro Tools, Cubase, Logic and Reason* systems. HyperControl updates the virtual instrument or track parameter mappings of each control in real time as you sequence—it doesn't get any easier.

Expressive Playability

With great feel in a lightweight chassis, the Axiom Pro controllers provide a new level of creative inspiration. The M-Audio TruTouch keybed combines professional semiweighted action with advanced high-speed keybed scanning technology for an ultra-expressive playing experience.

Keyboard Shortcuts

Complete MIDI Control

external MIDI devices.

The Axiom Pro keyboard controller line is the first to combine MIDI control with ASCII keystrokes. You can assign any button on Axiom Pro to send ASCII messages accessing any of the available QWERTY keyboard shortcuts within your host DAW. Imagine the possibilities with Copy, Paste, Duplicate, New Track-anything you want to assign.

sensitive trigger pads integrate acclaimed M-Audio Trigger Finger™ technology for laying down rhythm tracks or triggering loops and samples. Smoothly control parameters like filters, EQ and

panning with the eight non-detented rotary encoder knobs. The nine 40mm sliders (Axiom Pro 61 and 49 only) feature auto-mute to prevent jumping-ideal for controlling mixes, virtual

organ drawbars and more. In addition to USB connectivity, you'll

DAW Transport and Track Navigation

The six dedicated transport buttons on the Axiom Pro controllers give you immediate hands-on access to all of your host DAW's playback and recording controls. Unlike other all-or-nothing systems, Axiom Pro also features a transport-only mode. This option lets you access DAW transports and track navigation via HyperControl technology, while creating and controlling your own custom MIDI assignments for the trigger pads, buttons, sliders and

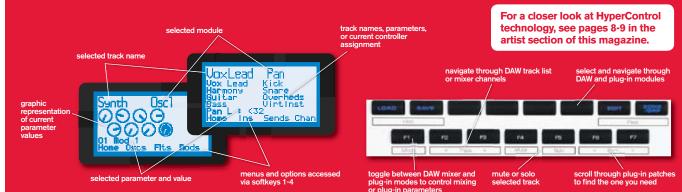
The Ultimate Pro Tools Keyboard Controller

Whether you are working on a Pro Tools M-Powered™, Pro Tools LE® or Pro Tools|HD® system, the Axiom Pro MIDI controllers set a new standard for Pro Tools keyboard control. Advanced M-Audio HyperControl technology puts **HYPERCONTROL** instant access to all of your Pro Tools parameters at your fingertips-including mixing, individual track functionality, and unparalleled control of virtual instruments such as Xpand!®, Strike™ and Hybrid™.



HyperControl-The Most Advanced, Intuitive Real-Time Link with Your Software

While other controllers offer mapping, only the M-Audio Axiom Pro line has HyperControl technology. HyperControl synchronizes Axiom Pro with your DAW, ensuring that your hardware and software collaborate to form a single work environment. The technology utilizes the Axiom Pro graphic LCD to group software controls together in a modular fashion. Controls are continually updated to reflect the mapped parameters. A dedicated button toggles between Mixer and Instrument modes for instant hands-on control of all of your software's functionality. Dedicated Track buttons allow sequential navigation of your mixer and instantaneous mapping of software instruments within the selected track. You can even control software instruments without opening them.





"I used the Axiom 49 and Axiom 61 for the synthesizer work I did on films such as The Good German, Alpha Dog and Bridge to Terabithia. I used the sliders and the keys to trigger synthesizers and the faders for filter resonance, volume and to control effects-it worked really well!" -Steve Tavaglione (artist/composer; Road to Perdition, American Beauty, CSI)

Advanced 61-Key Semi-Weighted USB MIDI Controller

The Axiom® 61 advanced 61-key USB mobile MIDI controller features semi-weighted action and assignable aftertouch. You get eight trigger pads, nine sliders, eight endless rotary encoder knobs, 15 MIDI-assignable buttons-including six transport controls-and more. Store and manage your setups with 20 onboard memory locations.

61-key, velocity-sensitive, semi-weightedaction keyboard with assignable aftertouch front-panel programming

8 MIDI-assignable trigger pads

8 MIDI-assignable rotary encoder knobs

9 MIDI-assignable sliders

15 assignable buttons, including 6 transport buttons

assignable pitch bend and modulation wheels octave up/down; transpose up/down

4 independent configurable keyboard zones

15 buttons and backlit LCD screen for total

assignable sustain and expression pedal jacks

USB MIDI interface including standard MIDI In and Out jacks

powered via USB or optional power supply class-compliant with Windows XP, Vista and

Mac OS X 20 non-volatile memory locations

memory dump via SysEx

all controllers fully programmable to MIDI controller number and channel

Controller Mute function mutes controller output to avoid parameter jumps

snapshot function transmits all current controller settings

weighs only 12 lbs. includes Live Lite

Axiom 49

Advanced 49-Key Semi-Weighted USB MIDI Controllei

If you need total MIDI control and playability in a compact package, check out the Axiom® 49 controller. This advanced 49-key USB mobile MIDI controller delivers semi-weighted action plus assignable aftertouch, along with eight rubberized trigger pads that are perfect for hands-on drum programming and performance. Nine sliders complement eight endless



rotary encoder knobs to control just about any software studio parameter you can think of. 15 MIDI-assignable buttons-including six transport controls-and assignable wheels and foot pedals round out total control. The dedicated front-panel keypad and backlit LCD screen provide intuitive setup. There's room for 20 non-volatile memory locations for instant access to setups.

49-key, velocity-sensitive, semi-weightedaction keyboard with assignable aftertouch

8 MIDI-assignable trigger pads

8 MIDI-assignable rotary encoder knobs

9 MIDI-assignable sliders

15 assignable buttons with 6 transport buttons

assignable pitch bend and modulation wheels plus sustain and expression pedal jacks

15 dedicated function buttons and backlit LCD screen for total front-panel programming built-in USB bus-powered MIDI interface

including standard MIDI In and Out jacks

class-compliant with Windows XP, Vista and Mac OS X

20 non-volatile memories

includes Live Lite



Axiom 25

"[The Axiom 25] boasts channel aftertouch and an excellent feel, and I reveled in what I reckoned to be the perfect compromise between a weighted piano action and the lighter response of a typical synth." -Sound On Sound

Advanced 25-Key Semi-Weighted USB MIDI Controller

The Axiom® 25 advanced 25-key USB mobile MIDI controller features semi-weighted action and assignable aftertouch, plus eight rubberized trigger pads for drum programming. You also get eight endless rotary encoder knobs, six transport controls, assignable pedal jacks, 20 memory locations, bus-powered operation, backlit LCD

25-kev. velocity-sensitive, semi-weightedaction keyboard with assignable aftertouch

8 MIDI-assignable trigger pads

8 MIDI-assignable rotary encoder knobs 6 reassignable transport buttons

assignable pitch bend and modulation wheels plus sustain and expression pedal jacks

backlit LCD screen

built-in USB bus-powered MIDI interface including standard MIDI In and Out jacks

class-compliant with Windows XP. Vista and Mac OS X

20 non-volatile memories

includes Live Lite



"M-Audio's modus operandi has always been to outwhen it's time to program them." -Keyboard

61-note, velocity-sensitive,

synth-action keyboard with

9 fully assignable MIDI sliders

15 assignable buttons, including

assignable pitch and mod wheels

USB MIDI interface

6 transport buttons

controller knobs

8 fully assignable MIDI

feature... the competition, and their MIDI keyboards tend to offer not just the generous quantity of controls for the price, but impressive flexibility, and clarity about what's going on,

> octave up/down; transpose up/down; preset up/down

MIDI channel/program change up/down

7 top-panel function buttons plus additional programming functions on keybed

3-segment LED screen sustain pedal jack

MIDI Out jack

powered via USB or optional power supply

class-compliant with Windows XP. Vista and Mac OS X

10 non-volatile memories all controllers fully programmable to individual MIDI channels

controller mute function mutes controller output to avoid parameter jumps snapshot function transmits all current controller settings

all controllers fully programmable to MIDI controller number

complement of MIDI controllers will also allow you to go

as deep as you want to go. Store 10 different setups in

onboard non-volatile memory. Bus-powered operation

and a weight of just 9 pounds means that you can use

Oxygen 61 anywhere your computer goes.

weighs only 9 lbs. includes Live Lite

Oxygen 49

49-Kev USB MIDI Controller

Oxygen 49 gives you the same control, flexibility and power as Oxygen 61, just with an octave fewer keys to accommodate tight spaces and even greater mobility. You get the same quality keyboard and built-in MIDI interface, plus knobs, sliders and buttons totaling 35 MIDI-assignable controllers for complete control of your hardware and software. Store 10 different setups in onboard non-volatile memory. Buspowered operation and a weight of only six pounds lets you play, compose, record and mix with ultimate mobility.

"If you are looking for a great balance of price, quality and features, the Oxygen line of keyboards definitely rates a close look" -MacDirectory



49-note, velocity-sensitive, synth-action keyboard with USB MIDI interface

8 fully assignable MIDI controller knobs

9 fully assignable MIDI sliders 15 assignable buttons, including 6 transport buttons

assignable pitch bend and modulation wheels: octave up/ down; transpose up/down

10 non-volatile memories; SysEx dump

3-segment LED screen with programming via 7 function buttons and keybed

rear-panel sustain pedal jack and MIDI Out jack

Windows XP, Vista and Mac OS X class-compliant

powered via USB or optional power supply

includes Live Lite

"[Oxygen 8 v2] packs a great degree of functionality into a tiny space and is great fun to use." -Future Music

Oxygen 8 v2

25-Kev USB MIDI Controller

The Oxygen 8 v2 is an updated version of the bus-powered MIDI controller that started the mobile studio revolution. Features include a MIDI keyboard with 25 full-size keys, pitch and mod wheels, eight MIDI-assignable knobs, six transport controls, 10 memory locations, software patch management and more-perfect for controlling software synths and DAWs. Compatible with Windows and Macintosh operating systems.

25-note, velocity-sensitive, synth-action keyboard with USB MIDI interface

8 fully assignable MIDI controller knobs

6 transport buttons also assignable to any MIDI parameter

assignable pitch bend and modulation wheels; octave up/ down; transpose up/down

3-segment LED screen with programming via 5 function buttons and keybed

rear-panel sustain pedal jack and MIDI Out jack

Windows XP, Vista and Mac OS X class-compliant

powered via USB or optional power supply 10 non-volatile memories; SysEx dump

includes Live Lite

Keystation Pro 88



"M-Audio have really filled a gap in the market for this kind of keyboard and, yet again, delivered a very practical product at an incredible price." -Future Music

The Keystation Pro 88 is the 88-key hammer-action USB master MIDI controller designed specifically to let you perform, program and mix music directly with your computer. Regardless of whether you're a seasoned pro or just ready to take your music to the next level, our hammer action is so expressive that you simply won't want to stop playing it. And the complement of 59 (yes, 59) assignable controls delivers more than enough tactile command over your favorite soft synths and DAWs like Pro Tools® M-Powered™ software. This bus-powered master controller unifies everything you need to experience the full potential of today's virtual studios-and at a light weight.



88-kev. hammer-action, velocitysensitive keyboard

24 MIDI-assignable rotary controllers

22 MIDI-assignable buttons

9 MIDI-assignable Alps faders MIDI-assignable pitch bend and modulation

wheels, foot switch and foot pedal inputs controllers assignable to MIDI controllers, notes,

transport controls, program changes (including LSB and MSB), SysEx and RPN/NRPN

independent MIDI channel assign for all controllers

four keyboard zones for layers and splits

multiple dynamic velocity curves

large custom LCD displays controller number, assigned controller, edit icons and edit values

10 memory locations for saving all controller assignments

MIDI In port

MIDI Out port can drive external MIDI gear or be used as USB to MIDI Out interface

USB class compliant-no drivers required for Windows XP, Vista (32 bit), or Mac OS X

powered via USB bus (cable included) or optional 9V power supply

lightweight for its features-only 47 lbs

includes Live Lite



Keystation 88es

88-Kev Semi-Weighted USB MIDI Controller

If you appreciate the full range of a piano keyboard in a lightweight package, the 88-note Keystation 88es controller is for you. You get semi-weighted action that's velocity-sensitive to convey all the nuances of your playing to your computer and most popular music education and studio software. Class-compliancy with Windows XP, Vista and Mac OS X ensures easy plug-and-play setup.



sturdy keyboard that's hard to beat at this price point" -Keyboard

88-note, velocity-sensitive, semi-weighted action

built-in USB MIDI interface with MIDI Out jack

pitch bend and modulation wheels

powered via USB or 9V DC

Mac OS X. Windows XP and Vista class-compliant includes Live Lite



Keystation 61es

61-Kev Semi-Weighted USB MIDI Controller

The Keystation 61es is a 61-note USB keyboard with velocity-sensitive, semi-weighted keys that is designed to easily integrate into any computer music environment. Class-compliancy with Mac OS X, Windows XP and Vista delivers true plug-and-play setup. Keystation 61es is also compatible with many music education and music creation software titles, making it ideal for classrooms and studios alike. More advanced users can control software synths, external sound devices and more with the assignable slider, and pitch and mod wheels. This sleek, compact keyboard is USB bus-powered and requires no external power supply.

61-note, velocity-sensitive, semi-weighted action

pitch and modulation wheels; volume/control slider

advanced function button for programming

sustain foot pedal input octave +/- buttons

MIDI Out jack routes computer MIDI to control external devices

powered via USB or 9V DC power adapter (sold separately)

includes Live Lite

KevRiq 49 turns your Mac or PC into a versatile keyboard workstation. Featuring a 49-key, synth-action USB MIDI keyboard and the award-winning Key Rig software for the PC, KeyRig 49 makes it easy to start composing and performing music with a computer. Play KeyRig's virtual instrument sounds, or audition loops and compose songs with Ableton Live Lite (included) or other music-creation programs. The keyboard also provides an easy way to harness the power of GarageBand's virtual instrument collection on your Mac. It connects via a single USB cable and works instantly with computers running Windows XP, Vista or Mac OS X.



49-note, synth-action keyboard includes award-winning M-Audio Key Rig software for PC assignable modulation wheel and volume slider USB compatible pitch bend wheel connection and power via a single USB cable assignable octave +/- buttons instant installation on Mac OS X. Windows edit mode button XP and Vista computers-just plug and play sustain pedal input includes Ableton Live Lite music creation software



KeyRig 25

KeyRig 25 turns your Mac or PC into a mobile keyboard workstation. Featuring a compact, half-action, 25-note USB MIDI keyboard and the award-winning Key Rig software for the PC, KeyRig 25 makes it easy to compose and perform computer-based music anywhere you want. Play Key Rig's virtual instrument sounds, or audition loops and compose songs with Pro Tools® M-Powered™ software, Ableton Live Lite (included) or other music-creation programs. The keyboard also provides an easy way to harness the power of GarageBand's virtual instrument collection on your Mac. It connects via a single USB cable and works instantly with computers running Windows XP, Vista or Mac OS X.

25-note, half-action keyboard assignable volume slider modulation and pitch bend controls eight fully assignable knobs and buttons octave +/- buttons sustain pedal input powered via USB or external 9V DC adapter (sold separately) built-in 16-channel MIDI interface instant installation on Mac OS X, Windows XP and Vista-just plug and play includes award-winning M-Audio Key Rig software for PC includes Ableton Live Lite music creation software



Key Rig software from M-Audio covers all the essential needs of today's keyboardist in a single virtual rack. Compatible with Windows environments, Key Rig delivers four great-sounding modules for standalone or plug-in use-SP-1 Stage Piano, MS-2 Polyphonic Synthesizer, MB-3 Tone Wheel Organ and GM-4 General MIDI Module. The master section lets you route, split, layer and mix these four powerful modules in whatever way best suits your performance and recording needs. A master effects section also allows for adding a common effect in addition to the individual effect section included with each instrument.

four top-quality virtual sound modules:

- SP-1 Stage Piano
- MS-2 Polyphonic SynthesizMB-3 Tone Wheel Organ
- GM-4 General MIDI

use in standalone mode or with a sequencer

variable velocity and controller settings, optimized for

independent MIDI channels

integral mixer

split and layer modules for performance combination

dual effects processor for each module

master effects section in addition to individual module effects compatible with most popular software including Pro Tools M-Powered



Trigger Finger

The Trigger Finger™ control surface puts the power to program and perform expressive percussion and drum parts at your fingertips. Its 16 velocity-sensitive pads are perfect for playing the drum sounds in your favorite software, launching loops and samples, or even controlling video projections-and applying pressure to the pads can generate any MIDI controller you wish. You also get eight knobs and four faders freely assignable to MIDI parameters such as volume, pan, pitch and effects. Pre-programmed maps for Live, Reason, GM Drum, XG Drum, iDrum and more make setup a snap. Connection and power is via a simple USB cable.

16 velocity- and pressure-sensitive pads 8 assignable knobs, 4 assignable faders pre-programmed maps for Live. Reason. GM Drum, XG Drum and iDrum velocity control including fixed velocity mode

program/bank change capability

powered from USB connection or optional DC power supply

16 MIDI presets

compatible with Windows and Mac OS X includes Live Lite

Trigger Finger is designed to put you in full command of your drum programming sessions. 16 pads give you immediate access to plenty of drum sounds or other events you wish to trigger. You'll find the pad size provides just the right balance of compact space and perfect touch. Speaking of perfect touch, Trigger Finger's pads are both velocity and pressure sensitive-and, unlike many control surfaces, you can map the pressure sensitivity to any MIDI parameter you desire for that extra measure of expressive control. Its eight knobs and four faders are also fully assignable to any MIDI parameter you want. With Trigger Finger, you're in control.

Control Anything You Want

While our main focus in designing Trigger Finger was drum programming, full programmability means that you can use it to control any MIDI devices you wish. Launch Clips or Scenes in Ableton Live, trigger video clips in VJ software, or fire off samples in your favorite DJ software-or anything else you can think of. There's even a MIDI Out jack that lets you connect external



"I used to lug around an MPC2000 on tours and it was a staple part of my studio. Now it's collecting dust because the Trigger Finger replaced it years ago. M-Audio gear helps me capture ideas quickly and easily." -Morgan Page (DJ/ producer/remixer; Delerium, Nelly Furtado, Stevie Nicks)

"With today's computer platforms packing all the drums, samples, and synths you need, the Trigger Finger is the missing link between you and your first great computer















"...a monitor rivaling the reproduction characteristics that cost much more."

INTRODUCING THE

Studiophile DSM Studio Monitors

The Studiophile® DSM1 and DSM2 studio monitors represent a strategic collaboration between M-Audio and Digidesign. M-Audio's heritage spans 20 years of award-winning innovation and the best-selling reference monitors in the United States. Digidesign's legacy includes the industry-standard Pro Tools® high-end audio production systems with DSP technology. As sister Avid® companies, our engineering teams combined forces to deliver uncompromising acoustic performance with flexible DSP-based EQ controls. The end result is a line of technologically advanced monitors with a sound you can trust-ideal for a broad array of creative and professional environments.

- co-developed by M-Audio and Digidesign > stra
- DSP-based EQ controls > custom (
- digital crossover technology > 1
- digital inputs up to 24-bit/192kHz > coni
- anodized aluminum-cone LF drivers > exce
- 1" Teteron soft fabric HF driver > accura
- 180-watt Class D bi-amplification > low distortion, high effic











ProKeys 88

Weighted Hammer-Action Premium Stage Piano

ProKeys 88 is the premium digital stage piano that doubles as a great MIDI controller. It's loaded with large, realistic samples—14 world-class instruments in all, including split/ layer capabilities. We paired those great sounds with an 88-key weighted hammer-action keyboard, delivering a realistic feel to satisfy the most demanding pro. ProKeys 88 is also a great master controller keyboard, including a built-in USB MIDI interface for easy direct connection to your PC or Mac, MIDI In and Out jacks to communicate with other MIDI gear, pitch and modulation wheels, sequencer controls and more. ProKeys 88 pulls out all the stops in creating the ultimate playing experience.



88-key, weighted, hammer-action keyboard stunning stereo grand piano samples with

3 velocity layers large sound bank with masterfully tweaked

samples of 14 world-class instruments:*

- Piano1: stereo grand piano based on Yamaha C7
- · Piano2: bright grand piano
- E Piano1: electric piano
- E Piano2: electric piano based on Wurlitze
- FM Piano 1: bright electric piano based on Yamaha DX7 • FM Piano2: warm electric piano based on Yamaha DX7
- Clay: instrument sound based on Hohner Clavinet D6
- . Vibes: vibes with hybrid mallets
- Organ1: percussion organ based on Hammond B3
- Organ2: rock organ based on Hammond B3
- A Bass: acoustic upright bass*
- E Bass: fingered electric bass (including harmonics)** · Strings: orchestral strings
- · Warm Pad: warm, lush synth pad

126-note polyphony single, split and layer modes

pitch and modulation wheels

sustain, sostenuto and expression pedal inputs

3 reverb spaces plus chorus and tremolo dedicated button to instantly recall

single-layer stereo grand piano

2-stage equalizer to easily optimize tone

12-key data entry pad with 3-digit LED screen

dedicated volume faders for master, layer and split

3 velocity curves plus fixed velocity to optimize feel

self-demo mode

+/- octave transpose keys

built-in USB MIDI interface

MIDI In/Out jacks to interface with external

dedicated remote MIDI controller buttons including:

- MIDI Clock Stop/Start
- Tempo control for MIDI Clock including tap tempo
- Program Change Channel
- Bank LSB
- Bank MSB
- Local On/Off

General MIDI compliant

stereo out (1/4" unbalanced)

stereo headphone out (1/4" TRS)

class-compliant drivers for Windows XP, Vista and Mac OS X

*This product is not connected with, or approved or endorsed by, the owners of the Wurlitzer, Clavinet, B3, and DX7 names. These names are used solely to identify the instruments emulated by

**Optional ride cymbal layer.

Full Pedal Complement

Every digital piano has a sustain pedal jack. But what about the sostenuto pedal (the middle pedal on an acoustic piano)? ProKeys 88 has it. There's also an expression pedal for foot control of volume And all three of these pedals serve as MIDI controllers for outboard

"Clearly there are gigging keyboardists on the design team... The complement of gozintas (sustain and soft pedal as well as expression pedal ins) and gozoutas (USB and MIDI) is best-in-class, and the front panel features are impressive for the money too. I've never met a [stage piano under \$1,000] with all these features, let alone one that sounded so good? -Ken Hughes. Kevboard



ProKeys 88sx

88-Key Lightweight Stage Piano with Semi-Weighted Action

At just 17 lbs., the ProKeys 88sx digital stage piano delivers excellent sound in a package so light that you can carry it under one arm. Its must-have complement of instruments-grand piano, electric pianos, organ and clav-deliver world-class sonic quality. The piano's semi-weighted action also delivers the best feel you can get without the added cost and weight of hammer action. Two stereo headphone jacks are perfect for private practice or teaching. The USB MIDI interface and pitch/mod wheels combine with other features to make it a great MIDI controller as well. If you want a pristine-sounding digital stage piano that feels great to both your fingers and your back, ProKeys 88sx is for you.

Dual Headphones Jacks. Pro Audio Outputs

ProKeys 88sx includes not one but two 1/4" stereo headphone jacks right up front-perfect for practicing or teaching privately. Of course, it wouldn't be an M-Audio product without high-fidelity 1/4" unbalanced audio output with specs like 20Hz-20kHz frequency response and 108dB dynamic range.



The ProKeys 88sx is a great go-to axe; keep it around for those gigs in town, rehearsals, practicing and writing. Put it on your desk in the studio, throw it over your shoulder with your lapton and take it on your boat, or carry it onstage with one hand and your drink in the other." -Keyboard

88-note, semi-weighted-action keyboard

7 premium sounds including multi-layer stereo grand USB MIDI interface plus pitch and mod wheels

2 front-side headphone jacks

sustain, sostenuto and expression pedal inputs

Portable Digital Pianos

with Audio Interface Technology

ProKeys Sono 88

ProKeys Sono 61



Write. Perform. Record.

Bridge the gap between writing, performing and recording music with M-Audio's family of portable digital pianos with built-in audio interfaces. The 88-, 61- and 49-note models all feature premium keybed action, advanced scanning technology and a stereosampled Steinway grand sound* for an expressive piano experience. ProKeys Sono 88 and 61 each feature additional instant-access sounds like electric piano, organ and clav-as well as a full onboard General MIDI sound set-providing top-notch standalone performance solutions.** You can also connect these bus-powered instruments to your computer and record your keyboard performance, vocals and instruments-or perform with all the keyboard sounds and instruments in your computer. M-Audio portable digital pianos take you from the studio to the stage and beyond.

"I ditched all my old sampling equipment in favor of KeyStudio 49i and a laptop. My new setup is simple to use and has been rock-solid on stage." -Gram Lebron (guitarist/keyboardist; Rogue Wave)



KeyStudio 49i



semi-weighted, velocity-sensitive keyboard^ advanced keybed scanning technology

onboard, stereo-sampled Steinway grand piano sound*

additional instant-access instrument sounds^

built-in 2 x 2 USB 16-bit/44.1kHz audio interface XLR microphone, 1/4" instrument and RCA inputs

1/4" jacks for stereo line output

61 features bright piano, electric piano, organ and clav

dual front-mounted headphone jacks zero-latency hardware direct monitoring General MIDI compatible with 128 onboard sounds plus drum kits and sound effects^

40-note (maximum) polyphony ^

assignable mod wheel and voice volume control pitch bend wheel

transpose +/- buttons assignable to alter octave, transpose, program, Bank LSB, Bank MSB, MIDI channel or master tune

Edit Mode button for advanced MIDI functionality MIDI Out from USB mode allows the keyboard to be used as a MIDI interface

USB bus-powered operation-no power supply required when connected to your computer

standalone use via 9V or USB power supply class-compliant MIDI requires

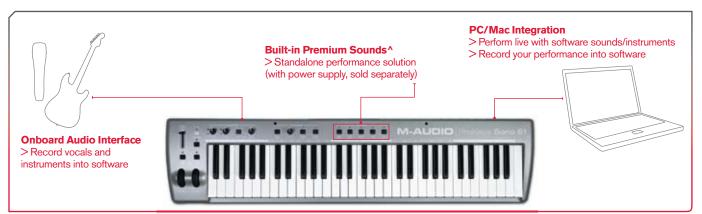
no additional drivers includes low-latency drivers for top performance sustain pedal jack

55

MIDI Out jack

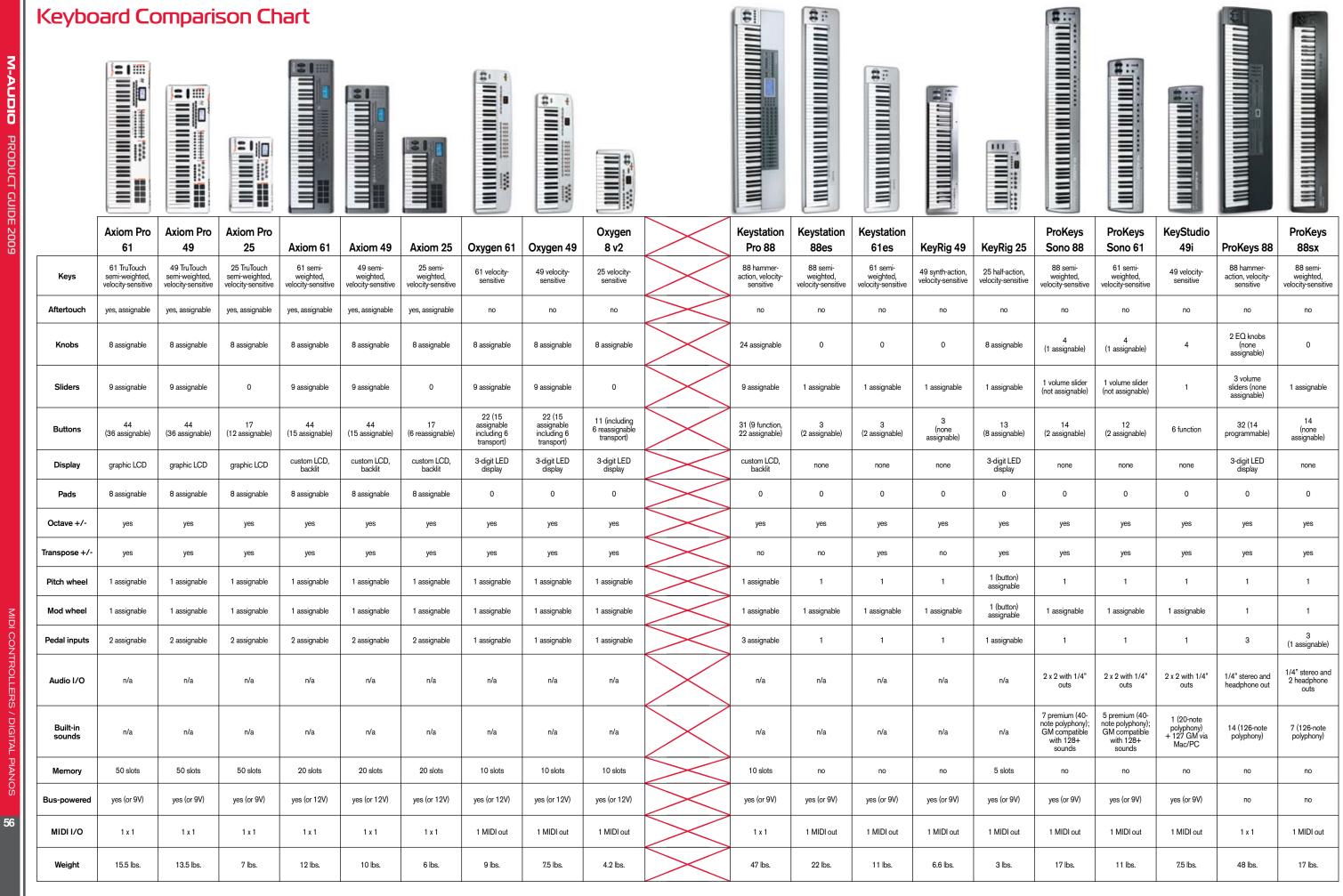
Piano Reset button restores all settings to piano sound (default state) includes Ableton Live Lite music software

^Semi-weighted action, additional instant-access instruments, 40-note polyphony and onboard GM sounds featured on ProKeys Sono 88 and 61 models only. KeyStudio 49i features premium synth action and 20-note polyphony, and includes M-Audio GM sound module software (which increases available sounds to 128). ProKeys Sono 88 features bright piano, electric piano, organ, clav, strings and choir. ProKeys Sono



*This product is not connected with, or approved or endorsed by, the owners of the Steinway name. This name is used solely to identify the instrument emulated by the ProKeys Sono keyboards. **Optional power supply required for standalone use.







STUDIO MONITORS

You can have the best gear in the world and still create less-than-perfect mixes unless you have monitors that tell your ears what's really happening. With that in mind, our design team set out to prove that world-class reference monitors don't have to cost a fortune. The result is the M-Audio® EX and Studiophile® line of reference monitors, delivering exceptional quality performance. And when you want to make it personal, check out the Studiophile® headphones and the IE reference earphones.

Studiophile DSM2

Professional High-Resolution 8-Inch Near-Field Studio Monitor with DSP



Studiophile DSM1

Professional High-Resolution 6.5-Inch Near-Field Studio Monitor with DSP



The Studiophile® DSM1 and DSM2 studio monitors are engineered to provide superior monitoring to discerning audio professionals. The design results from collaboration between two powerhouse companies in music technology: M-Audio, the top U.S. manufacturer of reference monitors,* and Digidesign, the industry leader in DSP technology. The DSM monitors feature DSP-based crossover management, an assortment of onboard digital EQ controls, proprietary high-grade driver designs, and analog and digital inputs for high-resolution audio up to 24-bit/192kHz. The end result is a line of technologically advanced monitors with a sound you can trust—ideal for a broad array of creative and professional environments.

1" high-frequency driver composed of a Teteron soft fabric, high-frequency dome with Neodymium magnet, ferrofluid cooling and built-in heat sink-accurate to 27kHz

6.5" or 8" custom low-frequency driver with anodized aluminum cone provides exceptionally low distortion

onboard DSP manages the complex digital crossover for pristine imaging

DSP-based EQ and Acoustic Space Controls optimize frequency response for any production environment

optimally tuned bass-reflex design with dual-flared port for controlled extended bass response with reduced turbulence

"Accuracy. Accuracy. Accuracy... These speakers stand up to the rigors of current, competitive design criteria, meaning that they sound really good next to similar speakers in their class and above. They're punchy in the extended low end without being tubby and accurate on top. Imaging is solid. The DSM2s would be a welcome addition to any production environment. At this price point, this level of accuracy is a bargain." —*Mix*

Redefining the World-Class Near-Field Monitor

Over the last 20 years, M-Audio has achieved critical acclaim across diverse lines of music production tools. We attribute much of this success to the fact that each product group operates like a small boutique shop within the organization. The Los Angeles-based engineering team that designed the industry-leading Studiophile line applied their expertise to the DSM drivers, power amps, cabinetry and electronics. Along the way, they worked with top recording engineers and producers to refine the Studiophile DSM monitors for optimal performance.

Powerful DSP

M-Audio engineers worked closely with their counterparts at Digidesign to bring their experience to the Studiophile DSM studio monitors. The powerful onboard DSP engine controls a complex 4th-order digital crossover. It also provides clear and accurate audio imaging by compensating for phase differences between the two drivers at the crossover point—delivering perfect acoustic alignment and exceptional imaging. And as a result of exhaustive tuning and listening sessions, the DSM monitors provide incredible tonal accuracy, excellent dynamic range and exceptionally low distortion.

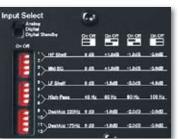
high-density cabinet yields good damping and reduced acoustic interference

professional digital inputs (up to 24-bit/192kHz): S/PDIF and AES

balanced analog inputs: XLR and 1/4"

bi-amplified with Class D power amps: 100 watts (low frequency) and 80 watts (high frequency)



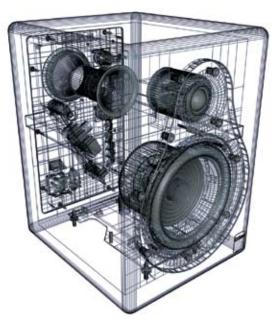


Tailor the Speaker to Your Room

Your listening environment and speaker placement greatly impact monitor performance. To ensure optimal monitoring in your studio, the onboard DSP also manages a broad range of customizable performance filter settings that allow you to tailor each speaker to your room. The DSM monitors offer DSP-controlled high and low frequency roll-offs along with an assortment of EQ settings from acoustic space to Q that help optimize the speaker for placement within even the most difficult acoustic environments.

Best-in-Class Drivers

To ensure the greatest sonic clarity, the Studiophile DSM studio monitors feature a custom-designed, anodized aluminum cone woofer that actively reduces distortion by pushing the breakup mode out of the pass band for maximum low-frequency detail at any volume level. The 1-inch ferrofluid-cooled Neodymium tweeter is made of a soft Teteron dome, which strikes the perfect balance between efficiency and damping to produce a controlled high-frequency breakup out to 27kHz. The drivers and their carefully matched power amps provide a wide frequency response and outstanding dynamic range, ensuring that you hear an accurate representation of your audio at all volumes.

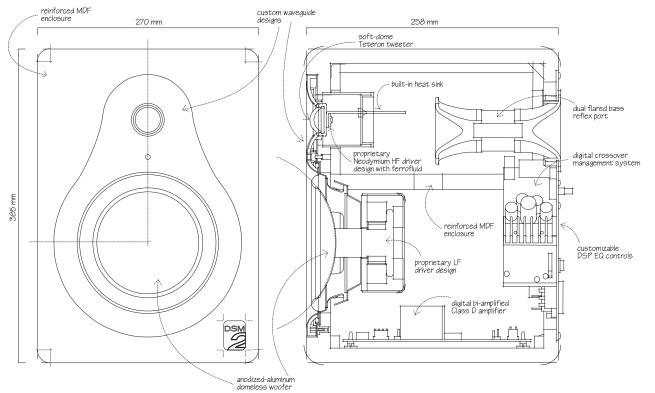


Strategically Integrated Design

Each DSM monitor component is customized to deliver outstanding sound quality in a compact form factor. Featuring an exclusive design, the sturdy cabinet is constructed from high-density material with internal bracing that renders it acoustically inert. The optimally tuned bass-reflex design uses a dual-sided flared port, which offers increased SPL with extended bass response and low turbulence, while the carefully matched Class D power amps deliver ample power with low distortion and excellent efficiency. The custom-voiced components ensure hours of accurate listening without fatigue. And the professional analog and digital inputs support studio-grade audio resolutions up to 24-bit/192kHz, revealing every detail that you've created with your digital audio workstation.

Uncompromised Reliability

Far beyond the industry standards, the DSM monitor design endures a strenuous 100-hour continuous pink noise qualification process, ensuring its reliability in the harshest studio environments. This 100-hour reliability test starts with each individual driver design. Once each driver design successfully passes, the entire assembly is qualified as a system via the same 100-hour procedure. This highly rigorous design qualification process ensures that the DSM monitors will perform at the highest level—for hours on end, every time you turn them on.



M-Audio EX Series EX66

Professional High-Resolution Active MTM Reference Monitor

The M-Audio® EX66 monitor is the culmination of years of research developing cutting-edge high-definition active reference monitor technology. Our midwoofer-tweeter-midwoofer (MTM) design yields a wide sweet spot while minimizing room reflections. The unique low-frequency drivers and custom titanium high-frequency driver deliver exceptional smoothness and transparency across the entire audio band, without resonant peaks. Onboard DSP handles cabinet resonance tuning and crossover optimization to round out an incredibly accurate frequency response. Digital inputs compatible with up to 24-bit/192kHz signals and high-impedance balanced analog inputs insure that your signal path is pristine. Sold individually, the M-Audio EX66 monitor represents a new high-resolution standard for both two-channel and multichannel applications.

Dual Woofers

The M-Audio EX66 monitor uses a midwoofer-tweeter-midwoofer (MTM) vertical array configuration to provide optimal imaging. Traditional two-way speaker designs can cause significant early reflections from consoles, floors and ceilings, resulting in coloration and smeared details in the audio. In the EX66 monitor's MTM design, the interaction of the two large drivers with the tweeter serves to channel the sound into a much narrower vertical pattern, thereby eliminating many of these reflection problems. The design simultaneously provides a large, coherent horizontal radiation pattern, giving you and your colleagues a much wider sweet spot.

"I'm so sick of hyped 'pro' speakers; it's so nice to find monitors... that aren't deceiving and actually tell me the story I need to get a good mix. You need to listen to the EX66s for yourself."

-Recording

MTM configuration yields wide sweet spot with reduced room reflections

two 6" custom linear-piston woofers deliver high durability with exceptional definition and precision

1" titanium dome tweeter produces smooth transparency without high-frequency smearing

bi-amplified with 200 watts continuous power per channel

DSP-tuned cabinets and crossovers for extremely accurate frequency response with no resonant peaks

"They serve a perfect function—just the right size with excellent imaging. They sound smooth and fall in a perfect space, which is important when you're working at a desk. I absolutely love them."

-Patrick Leonard (Grammy-nominated producer; Madonna)

high-impedance analog inputs (XLR and 1/4" TRS balanced)

S/PDIF and AES/EBU digital input up to 24-bit/192kHz for total digital studio connectivity

32-bit digital processing

Optlmage II high-frequency waveguide minimizes diffraction

Acoustic Space Control optimizes low-frequency response based on room placement soft limiter circuit to prevent clipping

calibrated volume control/input sensitivity control high-frequency boost/attenuation switch

mid-range "presence" switch (low-Q boost) low-cutoff switch (37Hz, 80Hz, 100Hz)

custom-designed MDF cabinet

internal adiabatic foam absorbs standing waves and enhances lowfrequency response and efficiency

dual-flange rear-port design to minimize air turbulence and associated friction noise

"One of the first things I listened to on the EX66 monitors was the title track off our last record, *Bleed Like Me*, which we mixed on Genelecs. I was amazed because I heard lots of things on the EX66s that I didn't hear when we mixed the record. I love mixing on them." —Butch Vig (artist/producer; Garbage, Smashing Pumpkins, Sonic Youth)

"The EX66s blew me away. If you make music, they will actually inspire you. Don't believe me? I bought the review units and gave my old monitors to my best friend. Color me converted."

-Keyboai

Unique Low-Frequency Drivers

The M-Audio EX66 woofers use our proprietary linear-piston technology to minimize driver break-up modes and inertial effects. Our drivers have exceptional intrinsic damping characteristics with absolutely no frequency response peaks or resonances in the audible band. They provide truly linear pistonic action over the entire musical spectrum, bringing all portions of each musical transient into correct alignment and revealing more of music's subtle details.

Titanium Tweeter

Our 1" tweeter sounds as smooth as it looks. Titanium is known in the aerospace industry for its high tensile-strength-to-weight ratio and its resistance to corrosion. We craft the M-Audio EX66 tweeters from titanium in order to create a stiff, yet responsive piston that moves natural resonant modes well outside the audible range. This allows the tweeters to reproduce every nuance of musical detail with crisp, transparent accuracy without any high-frequency energy smearing or the harshness sometimes associated with rigid high-frequency drivers. The titanium tweeter is a natural complement to the M-Audio EX66 low-frequency drivers, providing smooth, coherent reproduction over the entire audio band.

Noise-Free Bass-Reflex Port

Many loudspeaker enclosures use a port to form a Helmholtz resonator to improve bass response. In some ported speakers, however, this has the side effect of audible friction noise from air moving in and out of the enclosure. Not so with the M-Audio EX66 monitors. Our engineers developed a dual-flanged rear-cabinet port to minimize the vibrations normally generated by low-frequency signals. This custom port is extremely efficient in its air transfer characteristics and is virtually noise free. Rear placement also prevents port turbulence from interfering with the front dispersion of the midwoofers and tweeter.

Optimal Enclosure

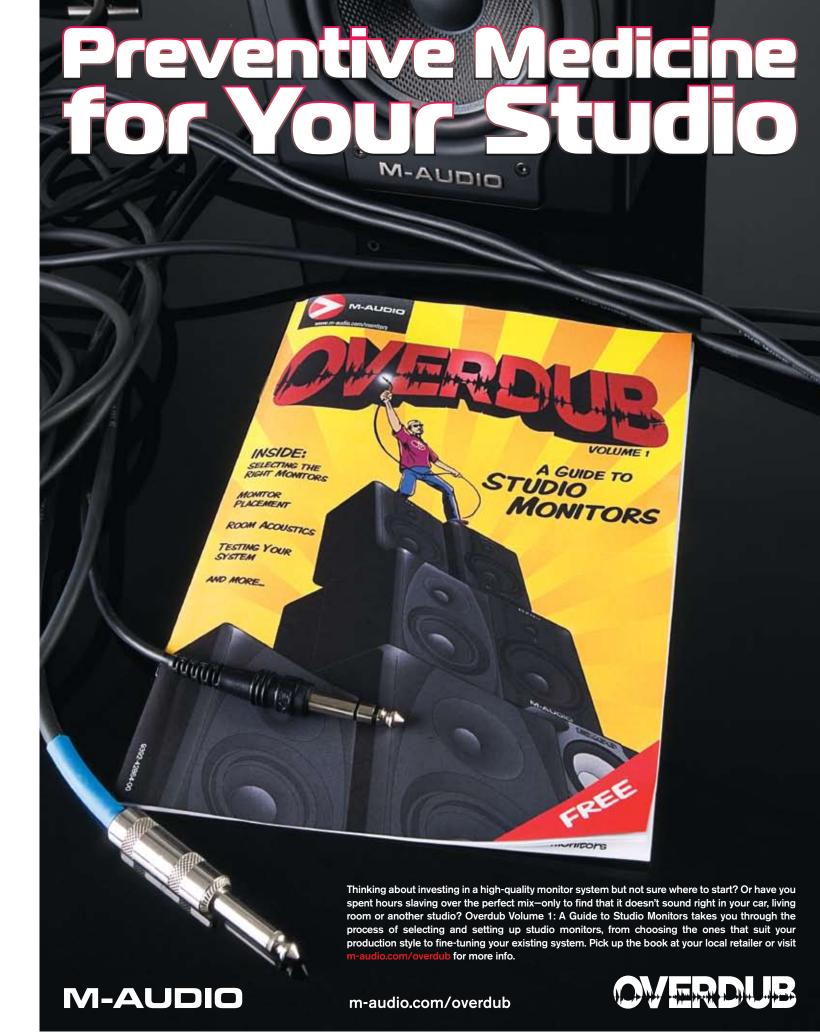
Like the other components, the M-Audio EX66 enclosure plays an important role in shaping the overall sonic response. In order to provide more stable performance, we designed an enclosure made of a special high-acoustic-efficiency medium-density fiberboard (MDF), along with unique interior adiabatic foam reinforcement designed to absorb extraneous vibration and standing waves. Furthermore, our custom Optlmage II waveguide on the front panel minimizes diffraction and dramatically improves stereo imaging.

DSP-Tuned Cabinets

All speaker cabinets exhibit resonant frequencies that color the timbre of the audio being reproduced. The extent to which they are mitigated is much of what allows critical listeners to distinguish between exceptional monitors and mediocre ones. The M-Audio EX66 monitors employ sophisticated onboard digital signal processing (DSP) technology to eradicate all resonances and tune the cabinet with exceptional precision. The M-Audio EX66 monitors also use DSP filters to achieve a precise and maximally flat 4th-order Linkwitz-Riley crossover for the smoothest transition between the high and low driver frequency ranges.

Superior Bi-Amplification

The M-Audio EX66 monitors use two separate 100-watt PWM power amplifiers to drive the 4-ohm woofer combination and 4-ohm tweeter separately in an audiophile-quality bi-amp structure. The ultra-low distortion at 200 watts continuous per channel rounds out the M-Audio EX66 monitor's profile as a truly world-class active loudspeaker monitor.







Studiophile BX8a Deluxe

130-Watt Bi-Amplified Studio Reference Monitors

Recording professionals around the world trust M-Audio Studiophile® reference monitors for exceptional sonic accuracy. The Studiophile BX8a Deluxe system builds on this legacy with new waveguides and enhanced driver integration for greater high-frequency clarity and refined sound. The magnetically shielded monitors feature 8" curved Kevlar low-frequency drivers plus high-temperature voice coils and damped rubber surrounds for durability and fidelity. The 1-1/4" natural silk, waveguide-loaded tweeters deliver a high end that's both clear and easy on the ears. Custom-tuned crossovers distribute 130 watts of bi-amplified power for optimal frequency handling by each driver.

"[The BX8a Deluxe monitors] sounded great-I could hear the clear definition of vocals and bass, especially in that range where they start to blend together. The bass was kickin' but it was still vibrant. At the same time, the samples and vocals sounded crisp and clear without being piercing." -9th Wonder (Grammy Awardwinning producer; Jay-Z, Mary J. Blige, Destiny's Child)

"I use the M-Audio BX monitors and I just love them. I can listen to them for long periods of time and never get any fatigue. I don't feel like I have to be sitting in a particular place to be in their sweet spot. I really find them to be very flat and accurate while at the same time being pleasant to listen to." -Vinnie Colaiuta (drummer/recording artist; Sting, Faith Hill, Quincy Jones, Steely Dan, Joni Mitchell)

two-way studio reference monitors

130 watts of continuous bi-amplified power

8" low-frequency drivers with:

- · curved Kevlar cones
- · high-temperature voice coils damped rubber surrounds

1-1/4" waveguide-loaded high-frequency drivers

magnetically shielded XLR balanced and 1/4" balanced/

unbalanced inputs

volume control and power indicator

Deluxe Edition

The M-Audio engineering team focused on several critical components to ensure that the Studiophile BX Deluxe monitors sound balanced at all volume levels. First, the waveguide design provides detailed imaging. The fine-tuned power amplifier sections provide optimal headroom and extra clarity. Cohesive frequency integration allows the drivers to work together so your ears hear the monitors as one source, not as a collection of drivers. It all adds up to a new standard from the industry leader in reference monitor technology.

Studiophile BX5a Deluxe

70-Watt Bi-Amplified Studio Reference Monitors

The Studiophile® BX5a Deluxe monitors update our bestsellers with new waveguides and enhanced driver integration for a cohesive and refined sound. The 5" low-frequency driver cones are crafted from Kevlar, an extremely resilient material. Curved cone design coupled with high-temperature voice coils and damped rubber surrounds deliver excellent fidelity and durability. The 1" waveguide-loaded, high-frequency drivers are made of magnetically shielded silk domes for a crisp top end that's gentle on the ears. 70-watt bi-amped power and expertly tuned crossovers round out a first-rate pair of reference monitors with a footprint small enough to fit just about anywhere.



two-way studio reference monitors

70 watts of continuous distributed power

5" low-frequency drivers with:

- · curved Kevlar cones
- · high-temperature voice coils · damped rubber surrounds
- 1" waveguide-loaded high-frequency drivers
- magnetically shielded
- XLR balanced and 1/4" balanced/ unbalanced inputs volume control and power indicator

"My M-Audio monitors are accurate and easy on the ears at a variety of volumes and listening positions, even after 12-hour sessions. They let me give the mixing engineer tracks that don't require much tweaking-so I know my music remains faithful to my vision." -Mark Isham (composer; Crash, A River Runs Through It)

> "The spec sheet doesn't lie: these brawny speakers kick out a generous amount of low end." -Future Music

"I have to go between analog and digital all the time and the M-Audio monitors have become my workhorse." -Terry Howard (Grammy-nominated engineer/producer; Ray Charles)



Studiophile BX10s

The Studiophile® BX10s active subwoofer is ideal for any monitoring environment that requires an accurate low-frequency response down to 20Hz. Comprising a 10" composite driver, 240-watt internal amplifier and a variable 50 to 200Hz crossover, BX10s delivers tight, clean bass in a compact chassis—perfect for project studios where space is often at a premium. The BX10s subwoofer is well-suited to either two-channel stereo or multi-channel surround monitoring environments and is designed to integrate with any powered directfield studio monitors, including the M-Audio BX5a and BX8a Deluxe active reference monitors. BX10s even includes a subwoofer bypass function so you can judge how a mix will sound without a dedicated subwoofercontrollable via the included footswitch.

10" composite driver

240-watt discrete amplifier

rear bass-reflex port

variable 50-200Hz crossover with easy settings for Dolby Digital®, DTS® and THX® compliance

low frequency response down to 20Hz adjustable gain (-30dB to +6dB)

"A footswitchable subwooferhow cool is that?

balanced XLR and 1/4" TRS inputs and outputs subwoofer bypass function with 1/4" footswitch input (footswitch included) auto on/off "sleep" function phase switch (0°/180°) magnetically shielded

"... I give high marks to the BX10s, especially for its very reasonable price."

removable grille and feet

Studiophile AV 40

Compact Desktop Speaker System

The compact Studiophile® AV 40 powered reference monitors deliver the proven M-Audio technology that's favored by top producers, recording engineers and musicians around the world. The combination of 4" polypropylene-coated woofers and 3/4" ferrofluid-cooled silk dome tweeters provides punchy lows, crisp highs and an even, uncolored response. And for the highest possible fidelity, the Studiophile AV 40 monitors also feature Optlmage III waveguide technology and a 20 watt-per-channel internal amplifier that utilizes Class A/B architecture. The Studiophile AV 40 reference monitors are the perfect solution for traveling musicians or anyone who wants top-notch sound from their desktop multimedia system.





Professional Components-Professional Results

The Studiophile AV 40 monitors are designed with the same quality materials and internal electronics that are used in M-Audio's acclaimed professional monitors. You'll find features like audiophile-grade Class A/B amplifier architecture and balanced 1/4" TRS inputs in addition to RCA connections. It all adds up to a big sound that belies the speakers' small size. With the Studiophile AV 40s, you don't have to give up professional-grade fidelity to enjoy the ease and mobility of a compact monitoring system.

Extended Low End

If you require extra bass response, the Studiophile AV 40 monitors have you covered. MDF wooden cabinets, bass reflex design and an integrated bass boost switch let the Studiophile AV 40s crank out plenty of low end-perfect for urban/dance music, DVDs and more.

4" polypropylene-coated woofers

3/4" ferrofluid-cooled silk dome tweeters

Optlmage III tweeter waveguides for superior imaging 20-watt-per-channel amplifier with Class A/B architecture magnetic shielding

1/4" TRS balanced and RCA unbalanced inputs

1/8" headphone output

1/8" stereo auxiliary input

front-panel controls for volume and power

bass boost control

MDF cabinet with bass reflex port design



"... if sound quality is high up on your list, you'll find a way to squeeze these bad boys onto your desktop" - CNFT.com

"... an excellent high-quality monitor system at an affordable price."



Studiophile Q40

Whether you're tracking, mixing or listening, you can expect sonic accuracy from your Studiophile® Q40 headphones-designed by M-Audio, a leading provider of reference monitor technology. With professional-grade dynamic drivers and ear cups that completely surround the ears, the Q40 headphones deliver full-range specs with excellent isolation from outside noise. Lightweight construction and comfortable earpads make them ideal for long studio sessions or hours of listening. These sturdy phones are collapsible for easy travel and even include a detachable/replaceable 3-meter cable to protect your investment. From making music to enjoying it, the Studiophile Q40 headphones let you get the most out of the experience.



40mm drivers provide exceptional frequency response closed-back circum-aural design delivers excellent isolation lightweight construction with comfortable earpads detachable, replaceable 3-meter cable with 1/8" connectors copper-clad aluminum voice coil wire for optimum efficiency includes 1/8"-to-1/4" adapter collapsible design for easy travel

Comfort and Convenience

M-AUDIO

The Studiophile Q40s are designed to go the distance with you-from traveling the globe to marathon studio sessions. A collapsible design means they'll easily fit in your carry-on luggage. The included storage bag protects them from dust and damage. The convenient removable and replaceable cord extends the life of your investment. And the large, padded ear cups completely surround your ears, ensuring a comfortable fit hour after hour

M-AUDIO



Full Frequency Response

Headphones deliver a level of sound quality matched only by speakers costing a great deal more-especially when it comes to low frequencies. With the Studiophile Q40 headphones, you'll feel like you're listening to a full-sized set of reference monitors in a professional studio environment. The Q40s provide excellent imaging and a wide sound stage for accurate mixing-without the hassle of worrying about speaker placement or room acoustics.

Superior Drivers

working in a shared space.

Full frequency response owes much to the size and construction of a speaker's driver. The Studiophile Q40 headphones feature extralarge 40mm-diameter drivers-bigger than most other headphonesfor extended low frequencies. Bass response is also

> enhanced by efficient components like neodymium magnets, copper-clad aluminum voice coils and vented diaphragms made of low-mass treated mylar. These elements come together in M-Audio's custom enclosure, which has

been specially tuned to further maximize low-frequency extension.

Closed-Back Circum-Aural Design

With sealed backings and comfortable ear cups that completely surround the ears, the Studiophile Q40 headphones deliver excellent sonic isolation for accurate monitoring. You'll hear more detail at lower volumes, so you can protect your hearing while tracking, mixing or listening in any noisy environment. In addition, the Q40 headphones afford privacy when

"I've found that the O40s are incredibly smooth across all frequencies. In both my mixing and tracking, the Q40 headphones provide me with a different perspective, without altering the quality of the playback." -Lyle Workman (guitarist composer; Sting, Beck, Superbad)

"What immediately sets them apart from most is that the cord screws in and out of the left cup, so you can replace it if it's damaged without scrapping the whole headphone set... the Q40s present a large and pleasant sound with a wide stereo image." -Remix

"M-Audio's Studiophile Q40s are compact, collapsible and well designed, and have very balanced, clear sound." -Electronic Musician



IE Reference Earphones

IE-40

High-Definition Professional Reference Earphones with Triple-Driver Technology

M-Audio® IE-40 reference earphones employ patented Ultimate Ears® technology to deliver an unparalleled personal listening experience-like carrying studio-grade reference monitors with you wherever you go. With the same triple-driver architecture, dual-bore design and sonic signature as Ultimate Ears' renowned custom monitors, the IE-40s provide a premium solution that fits any ear. The earphones operate like professional threeway monitors by splitting the frequency range across three drivers, which allows them to reproduce highs, mids and lows with pristine accuracy and superb balance. And with 26dB sound isolation, the IE-40s transform any location into an ideal monitoring environment—whether you're on stage, in the studio or traveling through a noisy public space.

"Having lived with the IE series for a while, I can honestly say that I will never go back to regular headphones for any of my mobile work, period. Laptop producers, DJs, and anyone who uses headphones regularly, absolutely must check out the IE series. I was utterly blown away, and everyone at Keyboard agreed they were a clear Kev Buv winner." -Kevboard



"These monitors are as good-if not better thanthe custom earphones available today... All the artists absolutely love them. They can't believe the low-end response and overall hi-fi sound qualitythe IE-40 monitors sound fat, crisp and clean with amazing isolation." - Marty Strayer (monitor engineer; Madonna, Seal, Tina Turner, Paul McCartney)

triple-armature driver design for separate high-, mid- and low-frequency reproduction patented dual-bore technology delivers highs and lows via separate canals for maximum sound field integrated passive crossover

to appropriate drivers 26dB isolation from outside noise

network directs sound frequencies

professional ear loops for a secure fit universal fit kit with silicone and foam tips for custom fit and maximum isolation

user-serviceable 46" cable gold-plated 1/8" input connector

with 1/4" adapter attenuator/limiter for overload protection

IE-30

"I use the IE-30s for monitoring live and in the studio. They have great isolation and tone. My bass always sounds fat and clear" - Milo Decruz (bassist: Duncan Sheik

High-Definition Professional Reference Earphones with Dual-Driver Technology

The IE-30 professional reference earphones employ a dual-driver design to deliver full frequency response and 26dB of isolation for extraordinary detail and accuracy, Stage-proven Ultimate Ears® technology makes them ideal for both performance and critical listening. Flexible ear loops and universal fit kit comfortably secure the earphones in place. Case included.



dual-armature driver design for separate high- and lowfrequency reproduction

dual-driver crossover audio wave shaping

26dB isolation from outside noise patented dual-bore design delivers highs and lows via separate canals for maximum sound field

universal fit kit with silicone and foam tips for custom fit and maximum isolation user-serviceable 46" cable

gold-plated 1/8" input connector with 1/4" adapter airplane attenuator/limiter for overload protection



IE-20 XB

High-Precision Professional Reference Earphones with Enhanced Bass

The M-Audio® IE-20 XB professional reference earphones use the same revolutionary Ultimate Ears® technology trusted by top touring acts-and feature enhanced bass designed for music that is normally monitored on systems with a subwoofer. The IE-20 XB benefits from a dual-speaker design with integral crossover: The dynamic lowfrequency driver delivers full, punchy bass while the precision-balanced armature driver provides crystal-clear mid-range and high frequencies. This results in a high-performance reference monitor system designed for critical listening during mobile recording/production of bass-enhanced music such as dance and urban genres.

dual-speaker design:

- armature driver for highs/mids
- · dynamic driver for extended
- low-frequency reproduction

audio wave shaping

custom crossov

16dB of isolation from outside noise

patented dual-bore design delivers highs and lows via separate canals for maximum sound field

universal fit kit with silicone and foam tips for custom fit and maximum isolation user-serviceable 46" cable gold-plated 1/8" input connector with 1/4" adapter airplane attenuator/limiter for overload protection



Professional Reference Earphones

Designed for musicians, the M-Audio® IE-10 professional reference earphones deliver studio-quality sound anywhere you need it. Using stage-proven Ultimate Ears® technology and providing 26dB of isolation, they're great for live and mobile monitoring-not to mention a serious upgrade for your personal audio player. Set features pro ear loops and universal fit kit.

ultra-lightweight polycarbonate housing 26dB of isolation from outside noise universal fit kit with silicone and foam tips for custom fit and maximum isolation user-serviceable 46" cable

gold-plated 1/8" input connector



Studio Monitor Comparison Chart





















· _		LF driver	HF driver	Power	Frequency response	Crossover frequency	Cabinet	Analog inputs	Digital inputs	Acoustic space controls	Weight	Dimensions	Max SPL at 1 meter	Additional features
	Studiophile DSM2	8" (203 mm) anodized aluminum	1" soft Teteron	100W LF / 80W HF	42Hz to 27kHz	2.7kHz @ 24dB/ octave	painted high-acoustic- efficiency MDF	XLR, TRS	S/PDIF in/thru, AES/ EBU	yes (6 controls)	20 lbs. (9.1 kg)	H 15.2" (385 mm) x W 10.6" (270 mm) x D 10.1" (258 mm)	111dB	DSP-controlled digital crossover and EQ
	Studiophile DSM1	6.5" (165 mm) anodized aluminum	1" soft Teteron	100W LF / 80W HF	49Hz to 27kHz	2.7kHz @ 24dB/ octave	painted high-acoustic- efficiency MDF	XLR, TRS	S/PDIF in/thru, AES/ EBU	yes (6 controls)	16.5 lbs. (7.5 kg)	H 12.8" (326 mm) x W 9" (229 mm) x D 10.3" (262 mm)	110dB	DSP-controlled digital crossover and EQ
	EX66	(2) 6" (152 mm) composite	1" titanium	100W LF/ 100W HF	37Hz to 20kHz	2.56kHz	painted high-acoustic- efficiency MDF	XLR, TRS	S/PDIF in/thru, AES/ EBU	yes (4 controls)	24.5 lbs. (11.2 kg)	H 18.9" (483 mm) x W 8.3" (210 mm) x D 10" (254 mm)	109dB	DSP-controlled digital crossover and EQ
	Studiophile BX8a Deluxe	8" (203 mm) Kevlar	1 1/4" silk	70W LF / 60W HF	40Hz to 22kHz	2.2kHz	vinyl-laminated MDF	XLR, TRS	n/a	no	26.5 lbs. (12 kg)	H 12" (305 mm) x W 10" (254 mm) x D 15" (381 mm)	106dB	ferrofluid liquid cooling technology
	Studiophile BX5a Deluxe	5" (127 mm) Kevlar	1" silk	40W LF / 30W HF	56Hz to 22kHz	3kHz	vinyl-laminated MDF	XLR, TRS	n/a	no	11 lbs. (5 kg)	H 9.8" (250 mm) x W 6.9" (176 mm) x D 7.9" (200 mm)	102dB	ferrofluid liquid cooling technology
	Studiophile BX10s	10" (254 mm) composite	n/a	240W	20Hz to 200Hz	50 to 200Hz sweepable	vinyl-laminated MDF	XLR, TRS	n/a	no	54.5 lbs. (24.8 kg)	H 15" (381 mm) x W 15" (381 mm) x D 15" (381 mm)	112dB	subwoofer bypass function with included footwitch
•	Studiophile AV 40	4" (102 mm) poly-coated paper cone	1" silk	20W	85Hz to 20kHz	2.7kHz	vinyl-laminated MDF	TRS, RCA (unbalanced)	n/a	no	14 lbs. (6.3 kg)	H 8.8" (222 mm) x W 6" (152 mm) x D 7.25" (184 mm)	97dB	integrated bass boost switch

	Application	Driver	Frequency response	Input sensitivity	Input impedance	><	Magnet material	Weight	Isolation	Input connector	1/8" to 1/4" adapter	Inline attenuator	Cable length	Additional features
Studiophile Q40	professional monitoring in the studio and on the go	40mm Mylar	10Hz to 20kHz	116dB SPL	64 ohms		neodymium	8.8 oz (.25 kg) w/o cable	n/a	1/8" (3.5 mm) gold plated	yes	n/a	10' (3048 mm)	detachable/replaceable cable
IE-40	critical listening across full spectrum	triple driver: low-, mid- and high-range precision-balanced armatures	10Hz to 17kHz	117dB/mW	32 ohms		n/a	0.6 oz (.017 kg) w/ cable	26dB	1/8" (3.5 mm) gold plated	yes	yes	46" (1168 mm)	custom soft and metal cases, detachable/ replaceable cable
IE-30	critical listening across full spectrum	dual driver: low- and high-range precision-balanced armatures	20Hz to 16kHz	119dB/mW	21 ohms		n/a	0.6 oz (.017 kg) w/ cable	26dB	1/8" (3.5 mm) gold plated	yes	yes	46" (1168 mm)	custom soft and metal cases, detachable/ replaceable cable
IE-20 XB	monitoring bass-enhanced music	dual driver: low- range 13.5mm diaphragm, high- range precision- balanced armature	20Hz to 16kHz	119dB/mW	11 ohms		n/a	0.7 oz (.019 kg) w/ cable	16dB	1/8" (3.5 mm) gold plated	yes	yes	46" (1168 mm)	custom soft and metal cases, detachable/ replaceable cable
IE-10	all mobile monitoring applications	single driver: full- range, precision- balanced armature	20Hz to 13kHz	115dB/mW	13 ohms		n/a	0.5 oz (.014 kg) w/ cable	26dB	1/8" (3.5 mm) gold plated	no	no	46" (1168 mm)	custom soft case, detachable/replaceable cable

Sputnik

Multi-Pattern Large-Diaphragm Vacuum Tube Condenser Microphone

M-Audio's Sputnik® mic signals the end of microphone envy. Based on a classic vacuum tube design and manufactured in modern facilities to exacting standards, this multi-pattern, largediaphragm studio condenser delivers the lush, classic sound normally associated with rare and expensive vintage mics like the U47 and C12. Sputnik's combination of a military-grade vacuum tube, ultra-sensitive evaporated-gold 3-micron Mylar diaphragm, solid brass construction and multiple polar patterns makes it ideal for a wide variety of voices, instruments and applications.

"The Sputnik sounds so close to my U47, it's just astounding." -Buddy Miller (Grammy-nominated singer/songwriter)

"It's nice to find a microphone that sounds clean and pure without severely coloring the signal one way or another. And very few microphones have such an elegant EQ response.

At my studio, I have a pair of Sputnik microphones permanently installed over the piano."

-Patrick Leonard (Grammy-nominated producer; Madonna)







3-micron, evaporated-gold Mylar diaphragm solid brass backplate new old stock 6205M vacuum tube

cardioid, figure-8 and omni polar patterns switchable 10dB (attenuation) pad

switchable 80Hz high-pass filter

solid brass body with polished nickel finish

all Sputnik microphones exhibit no more than +/- 1dB variation from the published curves

professional shockmount

custom briefcase and cloth bag

dedicated power supply with included 7-pin cable



"The Solaris is not only a great buy but also a proud addition to anyone's recording arsenal."



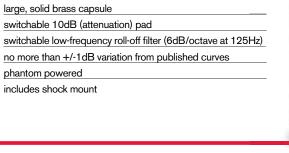
Solaris

Large-Diaphragm Multi-Pattern Condenser Microphone

The Solaris™ large-diaphragm, multi-pattern, studio condenser microphone delivers the best-sounding Class-A solid-state electronics available at any price. Like the rest of the award-winning M-Audio® microphone line, Solaris is designed in the USA and hand-assembled in limited quantities. Vintage design principles merge with incredibly tight state-of-the-art manufacturing standards to yield no more than +/-1dB of variation from published curves across the entire 20Hz-20kHz frequency range. The industry's thinnest evaporated-gold diaphragm mounted in a solid brass capsule delivers exceptional sensitivity and warmth. Multi-pattern architecture provides flexibility that makes it ideal for just about any recording situation-including vocals, guitar, piano, drums and live events. It's like having three mics in one.

large, solid brass capsule

phantom powered





"I've just always assumed that to be good, microphones had to be expensive, and preferably old. After using the Luna on a few sessions. I now know otherwise.

-Jeff Rona (film composer; Black Hawk Down)

Professional Large-Diaphragm Condenser Microphone

The M-Audio® Luna® II professional cardioid studio condenser microphone updates the award-winning Luna mic, acclaimed for superb sound and excellent performance. Designed in the USA and hand-assembled in limited quantities, Luna II delivers vintage look and sound combined with modern Class A, solid-state electronics for the best of both worlds. The large, 1.1" ultra-thin, 3-micron, evaporated-gold diaphragm with brass capsule captures nuances with incredible detail. The addition of a 10dB pad extends the maximum SPL to 140dB to handle just about any sound source. The inclusion of a switchable low-frequency roll-off filter rounds out a professional microphone that's great for recording



large, solid brass capsule

"...the Luna really captured the punch and spittle of a raging Marshall cabinet, as well as the pick attack of a Larrivée iumbo acoustic." –Guitar Plaver

"Pros: Extremely versatile recording mic. Sturdy chassis, Sleek design, Cons; None." -Remix

single-pattern, side-address, cardioid studio condenser microphone
vintage sound, modern technology
Class A FET electronics
ultra-sensitive, 3-micron, 1.1" evaporated-gold diaphragm

switchable 10dB (attenuation) pad switchable low-frequency roll-off filter (6dB/octave at 125Hz) no more than +/-1dB variation from published curves phantom powered

includes shock mount and wooden box



Large-Diaphragm Condenser Microphone

Using manufacturing breakthroughs pioneered by M-Audio's popular Luna[®] and Solaris[™] mics, the Nova[®] cardioid redefines the entry level for studio-quality condenser microphones. The 1.1" evaporated-gold diaphragm mounted in a solid brass capsule ensures recordings that are faithful to the source, and the Class A solid-state electronics are engineered for low noise, distortion and coloration.

large-capsule cardioid condenser 1.1" evaporated-gold diaphragm solid brass backplate Class A solid state electronics

"If you're on a tight budget, the Nova... rules."

includes hard mount and soft case



Aries

Live Vocal Condenser Microphone

The Aries professional condenser microphone is designed to capture high-quality live vocals. Internal shock-mounting allows Aries to deliver studio-quality sound while minimizing handling noise and standing up to the rigors of the road. Why settle for a dynamic mic when you can have the superior quality of a condenser?

"... Aries features a wider frequency response than the majority of vocal stage condensers out there, and, like most M-Audio products, is quite affordable for the quality." -Performing Songwriter

studio-quality condenser design

internal shock mount and pop filter

solid brass capsule with evaporated-gold diaphragm cardioid pattern for clarity and minimal feedback

20Hz-20kHz frequency response



Pulsar II

Small-Diaphragm Studio

The redesigned Pulsar™ II small-diaphragm cardioid condenser microphone takes M-Audio's pencil condenser technology to an even higher level. Designed in the U.S.A. by veteran engineers and handassembled to exacting standards, this precision microphone exhibits quality comparable to that of vintage classics. New features like 10dB pad and 80Hz high-pass filter switches, higher SPL than the original and a redesigned solid brass backplate deliver smooth, accurate sound in the studio and on stage. Pulsar II has many applications, including miking strings, woodwinds, acoustic guitar, percussion and more.

> "The Pulsar II combines a nice thick sound with detail and body-great for recording acoustic instruments." -Joe Barresi (producer; Weezer, Tool, Bad Religion)

MusicTech

"I really like the Pulsars on acoustic guitar. There's a lot of reality in the midrange. And they get some really nice sound off the wood. They're great mics with their own sound. They are not a copy of anything, and they don't have that irritating 4-6k bump so many other small-diaphragm mics have." -Buddy Miller (Grammy-nominated singer/songwriter)

professional small-capsule condenser microphone 3/4"-diameter, 6-micron Mylar evaporated-gold diaphragm

thermal diaphragm treatment yields 2dB SPL boost over original solid brass backplate

solid brass body Class A FET electronics switchable 10dB pad switchable 80Hz high-pass filter (12dB/octave) wooden storage box houses microphone, windscreen and clip

Pulsar II Matched Pair

Small-Diaphragm Studio Condenser Microphones

The Pulsar™ II Matched Pair delivers a boxed set of Pulsar II smallcapsule condenser microphones that are within +/-1dB of each other-at no additional premium. Featuring a deluxe wooden box and stereo mounting bracket, the Pulsar II matched pair will breathe new life into stereo miking applications including piano, drum overheads and room recording.





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M-Audio Microphone Applications							
	Sputnik	Solaris	Pulsar II	Pulsar II pair	Luna II	Aries	Nova
Voice	***	**	*		**	*	*
Voice (live)						***	
Grand Piano	***	**		***	**		
Guitar/Bass Amp	***	**	*		**	*	*
Acoustic Guitar	***	**	***	***	*	*	*
Guitar (smashed on stage)			*			***	*
Drum Overheads	**	**		***	*		
Hi-Hat	**	**	***		**	*	*
Snare	**	*	*		*	**	*
Toms	**	*	*	*	*	*	*
Kick		*			*		
Violin	**		***	***			
Viola/Cello	***	**	***	***	**		
Double Bass	***	**	**	**	**		
French Horn	***	*					
Flute/Piccolo	***	*	***	***	*		
Oboe/English Horn	***	**	***	***	**		
Clarinet	***	*	**	**	*		
Bassoon	***	**	*	*	**		
Contrabassoon/Bass Clarinet	***	***			**		
Tuba	***	**			**		
Soprano Sax	***	**			**	*	*
Alto & Tenor Sax	***						*
Vibraphone	*			***			
Marimba/Xylophone	***			***			
Timpani	***	**			**		
Choir (distant)	***	**		***	*		
Congas/Bongos	***	**	***	***	**	*	*
Tabla	***	**	*	*	**		*
Leslie Lower Rotor	***	***			***	*	
Leslie Upper Rotor	***	**	***	***	**	*	
Tree (falling in the woods)	***	**	***	***	**		

DMP3

"The DMP3 is shockingly transparent for a preamp with a \$200 list price." - Jazz Times

Dual Microphone/Instrument Preamp and Direct Box

Based on the M-Audio® preamplifier technology that won Pro Audio Review's highest accolades, the DMP3 preamp is designed to provide a premium front end for today's high-resolution digital recording systems. It features an amazing 20Hz to 80kHz frequency response, 67dB of gain and an exceptional dynamic range to capture all the nuances of your music. With pristine sound quality and a compact, rugged chassis, DMP3 has become the top choice of numerous audio professionals and a fixture in studios around the world. Whether you use it for performance or production, the DMP3 two-channel microphone/ instrument preamp is destined to be integral in defining your sound.



"One of the quietest preamps available anywhere, regardless of price. For comparison's sake, this tiny difference [between original source and mic pre output] is far less than the degradation I've heard in the most expensive world-class A/D converters." -Pro Audio Review

2-channel preamp with XLR microphone inputs and 1/4" instrument inputs

balanced outputs on TRS jacks low harmonic distortion (THD+N:

.0005% @ 20Hz-20kHz) exceptional dynamic range 20Hz to 80kHz frequency response (+/- 0.5dB) ideal for high-resolution recording

high and low gain range controls offer up to 67dB of gain

clip LED for visual feedback of detrimental levels

low-cut filters remove unwanted rumble and hum

phantom power to accommodate all types of microphones

phase reverse switch on each channel ensures optimal recordings



Audio Buddy

Microphone Preamp and Direct Box

The Audio Buddy is one of the most popular mic preamps in home recording today. The reasons are easy to understand: Performance and value. You get professional microphone support along with instrument inputsimpedance matched for ideal performance with electric guitar and bass. Simply put, the Audio Buddy is a workhorse mic preamp that delivers great sound.

2 channels, each including:

- mic input (balanced XLR) with phantom power switch
- high impedance instrument inputs (1/4")

- signal/clip LEDs
- professional line output (balanced/unbalanced 1/4" TRS)

super-small size/weight: 5.5" x 3.3" x 1.7"; less than .5 lbs.

Mobile Recording. Remastered.



ProFire 610

High-Definition 6-in/10-out FireWire Audio Interface with Octane Preamp Techr

M-Audio revolutionized mobile music production with the • 6 x 10 simultaneous analog/digital I/O FireWire 410—one of the best-selling FireWire interfaces • 2 preamps with award-winning M-Audio Octane technology 6-in/10-out mobile audio interface that incorporates technology from our critically acclaimed ProFire 2626. Premium digital converters ensure pristine 24-bit/192kHz audio throughout the entire signal path. Two preamps with award-winning Octane™ technology offer optimal headroom, superior sound quality and extremely low distortion. Standalone mode lets you use the unit as a self-contained, two-channel microphone preamp and A/D-D/A converter. ProFire 610 also offers dual headphone outputs and an advanced onboard DSP mixer—all in a compact, bus-powered enclosure.

ProFire 610 back panel



- ever. Now the legacy continues with ProFire™ 610, a powerful flexible onboard DSP mixer for creating multiple unique cue mixes
 - up to 24-bit/192kHz for high-definition audio
 - user-assignable master volume knob
 - dual headphone outputs
 - doubles as a 2 mic pre/2-channel A/D-D/A converter



The ProFire 610 control panel features a flexible DSP mixer that allows you to create, save and recall custom configurations for different recording setups. The control panel also facilitates instant access to driver updates, manual downloads, FAQ information and more.

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USB MIDISPORT 4x4

In commemoration of M-Audio's 20th anniversary, the Midiman MIDISPORT 4x4 Anniversary

Edition gives our popular multi-port USB MIDI interface an updated new look. MIDISPORT 4x4

addresses a total of 64 x 64 discrete MIDII/O via USB, allowing you to simultaneously connect a

2-In/2-Out USB Bus-Powered MIDI Interface

Designed in celebration of M-Audio's 20th anniversary, the Midiman MIDISPORT 2x2 Anniversary Edition multi-port MIDI interface delivers an updated new look. The compact interface addresses 32 x 32 discrete channels of MIDI I/O via USB, making it ideal for connecting multiple MIDI devices to your computer simultaneously, as well as eliminating delays associated with long MIDI THRU chains. Standalone functionality also gives you MIDI throughput and merge capabilities. Bus-powered architecture provides the option of totally mobile operation.

64 discrete MIDI input and output channels

MIDI activity indicator for each port



2-in/2-out MIDI interface 32 discrete MIDI input and output channels standalone MIDI merge and THRU functionality connection and power via USB

compact and lightweight for mobile or desktop use MIDI activity indicator for each port class-compliant with Mac OS X. Windows XP and Vista*



1-in/1-out MIDI interface 16 discrete MIDI input and output channels bus-powered-requires no external power supply

class-compliant with Mac OS X, Windows XP and Vista* extremely compact and lightweight for mobile use

USB MIDISPORT 1x1

1-In/1-Out USB Bus-Powered MIDI Interface

The MIDISPORT 1x1 is the perfect solution if you're just starting out or only need basic 16-channel MIDI in/out connectivity for your computer. The unit is class-compliant under Mac OS X, Windows XP and Vista*-no drivers needed. The bus-powered design makes it easy to take anywhere.



USB Uno

1-In/1-Out USB Bus-Powered MIDI Interface

Our smallest and simplest USB MIDI interface, USB Uno offers basic 1 x 1 operation and bus-powered mobility-and even has its own built-in USB and MIDI cables. It's also class-compliant under Mac OS X, Windows XP and Vista*-no drivers needed. What could be simpler?

1-in/1-out MIDI interface 16 MIDI input and

output channels bus-powered-requires no external power supply ultra-compact and lightweight built-in USB and MIDI cables compatible with Windows and Macintosh operating systems class-compliant with Mac OS X, Windows XP and Vista*





MidAir

Wireless MIDI Transmitter and Receiver System

Bring your MIDI gear into the wireless age with the MidAir™ system. Comprised of a portable transmitter and receiver, MidAir allows you to wirelessly transmit data from any device that outputs MIDI data-keyboards, drum machines, electronic drums and more. Just plug the battery-powered transmitter into the MIDI Out port on your controller of choice and connect the receiver to the MIDI In port on the device you'd like to control, and you're set. As a 2.4GHz wireless device, MidAir enables you to interface MIDI hardware at distances up to 30 feet with the same feel and response as a wired unit. The MidAir receiver includes a USB port and functions as a class-compliant 1 x 1 MIDI interface, so you can also wirelessly control soft synths from your favorite MIDI controller.

2.4GHz wireless MIDI system battery-powered transmitter receiver powered by USB bus or 9V power supply

1 x 1 USB MIDI interface including standard MIDI In and Out jacks discrete operation of multiple MidAir systems in close proximity class-compliant with Mac OS X, Windows XP and Vista

Coaxial/Optical Bi-Directional Converter

The CO2 is a compact and rugged digital audio tool that converts S/PDIF signals from optical to coax and from coax to optical. Its 2-in/2-out design allows the CO2 to operate as a full-duplex, bi-directional converter or as a half-duplex converter with THRU port-allowing it to work as both a converter and a repeater. It's easy to use several CO2 units if needed.



converts between optical and coax S/PDIF S/PDIF optical and coaxial digital outputs always active serves as a repeater to extend cable runs transformer-isolated RCA jacks prevent system ground loops S/PDIF optical and coaxial digital inputs

ACCESSORIES



MicroPack

ring Case for the MicroTrack Recorder

Protect your M-Audio MicroTrack recorder with the M-Audio MicroPack. With a full complement of zippered pockets, the MicroPack allows you to keep all of your cables, mics and accessories in one place.

Torq Xponent Gig Bag

Torq Xponent, Laptop and DJ Accessories

The M-Audio® Torg® Xponent® Gig Bag features padded sleeves, high-contrast interior, built-in cable management and multiple compartments-along with rugged nylon construction for durability.



Keyboard Covers

ve Covers for All Keyboards

Protect the tools of your trade with M-Audio's rugged keyboard covers. Made from 100% stretchable nylon with adjustable locking drawstring. Four sizes. Limited lifetime warranty



ProjectMix I/O Studio Bag

Designed to fit one ProjectMix I/O unit, a laptop, and a collection of cables and accessories, the ProjectMix I/O Studio Bag is the perfect way to take a personal recording system on the road. The bag features four padded compartments, a reinforced bottom, shoulder strap and a carrying handle.



Studio Pack

Studio on Your Back

M-Audio's padded ballistic nylon Studio Pack conveniently carries an M-Audio® Ozone®, Oxygen8 or Oxygen 8 v2, a



Mobile Laptop Studio Messenger Bag

Carrier for Your Laptop and Studio Essentia

The M-Audio Mobile Laptop Studio Messenger Bag is the perfect way to take your studio or performance rig with you wherever you go. The bag features rugged construction, padded compartments and a sturdy, adjustable shoulder strap. Capable of holding a 17" laptop, full-size vinyl records, keyboards, audio interfaces and more, the M-Audio Messenger Bag is the ideal solution for the musician on the go.



al Expression Controller Pedal

M-Audio's EX-P expression controller pedal works with all M-Audio® MIDI controllers that have an expression pedal input. A built-in polarity switch insures compatibility with most other brands of controllers and keyboards. Use EXP to control any assignable variable MIDI controller value including volume, modulation, panning, filter sweep and more. Durable molded construction with integral cable.



The SP-1 is a simple, compact and sturdy sustain pedal with polarity switch. Compatible with all electronic keyboards, it is perfect for any application that requires a non-latching momentary switch.



This high-quality damper pedal with expressive half-pedal capabilities and polarity switch is compatible with all electronic keyboards. Designed to work and feel just like an acoustic piano's sustain pedal, it has a specially designed rubber bottom that grips the floor.



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*Driver installation required for multi-client operation

SPECIFICATIONS

Please visit m-audio.com for additional product information. All specifications are measured at 48kHz sampling rate unless

Aries

type:	
handheld small-diaph	ragm condenser
Pattern:	cardioid
Frequency Response:	20Hz to 20kHz
Preamp Topology:	class A FET, transformerless
Sensitivity:	13.8mV/Pa (-37dBV re 1V/Pa)
Max SPL for 0.5% THD:	134dB SPL
Equivalent Noise Level:	17dB A-weighted
Output Impedance:	200 ohms
Recommended Load	
Impedance:	>1,000 ohms
Connector:	3-pin male XLR
Power Requirement:	48V phantom power
Dimensions:	6.60" (168 mm) x 1.57" (40 mm)
Weight:	.58 lbs (.26 kg)
Audio Buddy	

Mic Inputs (analog Max Gain:

requency Response:

Max Input levels: -18dBu @ min gain, -37	'dBu @ max gain
Signal-to-Noise Ratio,	-
Mid Gain:	-95dB, A-weighted
Dynamic Range, Mid Gain:	95dB, A-weighted
THD+N, Mid Gain: 0.021% (-73dB), 1dB b	pelow clipping, 22Hz to 22kHz
EIN Noise Rating: 20kHz BW (unweighter	-116dBu @ max gain, 150 ohm load d)

>60dB

20 to 50kHz, +/-1dB

Instrument Inputs (analog)

Phantom Power:

Max Gain:	40aB
Frequency Response,	
Min Gain:	20Hz to 50kHz, +/-0.1dB
Signal-to-Noise Ratio,	
Min Gain:	-104dB, A-weighted
Dynamic Range, Min Gain:	104dB, A-weighted

THD+N Min Gair 0.007% (-83dB), 1dB below clipping, 22Hz to 22kHz

Audiophile 192 Line Inputs (A/D)

Impedance: 20k ohms balanced, 10k ohms unbalanced Max Input Level: +14.2dBu (4 Vrms), balanced/unbalanced

Channel-to-Channel <-130dB @ 1kHz Signal-to-Noise Ratio: -113dB, A-weighted 113dB, A-weighted namic Range: THD+N: 0.00061% (-104dB) @ -1dBFS, 1kHz

Frequency Response: 20Hz to 22kHz, +/-0.1dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.1dB @ 96kHz sample rate 20Hz to 80kHz, +/-0.5dB @ 192kHz sample rate

Main Outputs (D/A)

Impedance: 300 ohms, balanced; 150 ohms, unbalanced +14dBu (4 Vrms) balanced Channel-to-Channel

Signal-to-Noise Ratio: -108dB. A-weighted Dynamic Range: THD+N. 0.00235% (-92.6dB) @ -1dBFS, 1kHz

20Hz to 22kHz, +/-0.1dB @ 48kHz sample rate

20Hz to 40kHz, +/-0.1dB @ 96kHz sample rate 20Hz to 80kHz, +/-0.5dB @ 192kHz sample rate

Monitor Outputs (D/A) Max Output Level:

+13.9dBu balanced, +5.7dBV unbalanced

Channel-to-Channel <-100dB Crosstalk Signal-to-Noise Ratio: namic Range: 100dB, A-weighted THD+N: 0.00279% (-91.1dB) @ -1dBFS. 1kHz

Frequency Response: 20Hz to 22kHz, +/-0.1dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.1dB @ 96kHz sample

Audiophile 2496

Line Inputs (A/D)

Max Input Level: +2.2dBV (1.3 Vrms) Channel-to-Channel <-120dB @ 1kHz Crosstalk Signal-to-Noise Ratio -100dB, A-weighted Dynamic Range: 100dB, A-weighted 0.00705% (-83dB) @ -1dBFS, 1kHz

Frequency Response: 20Hz to 22kHz, +/-0.15dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.4dB @ 96kHz sample rate

Main Outputs (D/A) Max Output Level

+1.9dBV (1.247 Vrms) Channel-to-Channel Crosstalk: <-115dB Signal-to-Noise Ratio -108dB, A-weighted Ovnamic Range: 108dB, A-weighted

THD+N: 0.00262% (-91.6dB) @ -1dBFS, 1kHz

Frequency Response: 20Hz to 22kHz, +/-0.1dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.2dB @ 96kHz sample rate

Delta 1010

Line Inputs (A/D)

22k ohms balanced, 11k ohms unbalanced Max Input Level: +20dBu (+1.8dBV (1.23 Vrms) unbalanced +20dBu (7.75 Vrms) balanced, Channel-to-Channel <-105dB @ 1kHz Crosstalk: -109dB, A-weighted Signal-to-Noise Ratio: 109dB, A-weighted Dynamic Range: THD+N: 0.00072% (-103dB) @ -1dBFS, 1kHz Frequency Response: 20Hz to 22kHz, +/-0.3dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.9dB @ 96kHz sample rate

Main Outputs (D/A)

2k ohms balanced; 1k ohm unbalanced Max Output Level: +20dBu (+1.8dBV (1.23 Vrms) unbalanced Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: -117dB, A-weighted Dynamic Range: 117dB, A-weighted 0.00200% (-94dB) @ -1dBFS, 1kHz Frequency Response: 20Hz to 22kHz, +/-0.1dB @ 48kHz sample rate

20Hz to 40kHz, +/-0.6dB @ 96kHz sample rate

Delta 1010LT

Mic Inputs Max Input Level, Min Gain: -26dBu, balanced Channel-to-Channel Crosstalk: <-100dB @ 1kHz -93dB, A-weighted Signal-to-Noise Ratio: 93dB, A-weighted Dynamic Range 0.00420% (-87.5dB) @ -1dBFS, 1kHz Frequency Response: 20Hz to 20kHz +/-0 15dB

Line Innuts

Max Input Level, Min Gain: +14.3dBu (4 Vrms), unbalanced Channel-to-Channel <-110dB @ 1kHz -98dB, A-weighted Signal-to-Noise Ratio Dynamic Range: 98dB, A-weighted THD+N: 0.00377% (-88.5dB) @ -1dBFS, 1kHz

Frequency Response: 20Hz to 20kHz, +/-0.1dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.2dB @ 96kHz sample rate

Line Outputs Max Output Leve +14.2dBu (4 Vrms) unbalanced Channel-to-Channel <-110dB Signal-to-Noise Ratio: -110dB, A-weighted 110dB, A-weighted Dynamic Range: THD+N

0.00230% (-92.8dB) @ -1dBFS, 1kHz Frequency Response puency Response. 20Hz to 20kHz, +/-0.1dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.25dB @ 96kHz sample rate

Delta 44

Line Inputs (A/D)

edance: 22k ohms balanced, 11k ohms unbalanced Max Input Level: +20.5dE +12.2dBV (4.064 Vrms), unbalanced +20.5dBu (8.182 Vrms) balanced: Channel-to-Channel <-100dB @ 1kHz Signal-to-Noise Ratio: -100dB, A-weighted 100dB, A-weighted amic Range:

0.00371% (-88.6dB) @ -1dBFS, 1kHz Frequency Response: 20Hz to 20kHz, +/-0.1dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.4dB @ 96kHz sample rate

Line Outputs (D/A)

2k ohms balanced, 1k ohm unbalanced Max Output Level: +13.9dBu (3. +11.7dBV (3.859 Vrms), unbalanced +13.9dBu (3.859 Vrms), balanced; Channel-to-Channel -110dB, A-weighted Signal-to-Noise Ratio: 110dB, A-weighted Dynamic Range: 0.00200% (-94dB) @ -1dBFS, 1kHz

Frequency Response: 20Hz to 20kHz, +/-0.1dB @ 48kHz sample rate

20Hz to 40kHz, +/-0.3dB @ 96kHz sample rate

DMP3

Mic Inputs (analog)

34dB to 67dB, high gain range Frequency Response: 20-80kHz, +/-0.5dB Max Input Levels: 14.6dBu @ min gain, low range Signal-to-Noise Ratio: 120dB, A-weighted, min gain Dynamic Range: -120dB, A-weighted, min gain ThD+N: 20kHz, 13.6dBu input, min gain ElN Noise Rating: -128dBm @ 600 ohms, max gair
Max Input Levels: 14.6dBu @ min gain, low range Signal-to-Noise Ratio: 120dB, A-weighted, min gain Dynamic Range: -120dB, A-weighted, min gain THD+N: <0.0005% (-106dB), 20Hz to 20kHz, 13.6dBu input, min gain
Signal-to-Noise Ratio: -120dB, A-weighted, min gain Dynamic Range: -120dB, A-weighted, min gain THD+N: <0.0005% (-106dB), 20Hz to 20kHz, 13.6dBu input, min gain
Dynamic Range: -120dB, A-weighted, min gain THD+N: <0.0005% (-106dB), 20Hz to 20kHz, 13.6dBu input, min gain
THD+N: <0.0005% (-106dB), 20Hz to 20kHz, 13.6dBu input, min gain
20kHz, 13.6dBu input, min gain
EIN Noise Pating: 109dPm @ 600 abms, may gain
EIN Noise Rating1260bit @ 600 offitis, max gair
Low Cut Filter: -3dB @ 72 (18dB/octave)
Input Impedance: 3k ohms
Phantom Power: yes

Digidesign-qualified Windows XP- or Mac-based Pro Tools system running Pro Tools 7.1 or higher software (Pro Tools 7.3 or greater recommended for optimal performance) Avid Xpress, Avid Xpress DV or Avid DNA system

iLok USB Smart Key (sold separately), Internet access and a free iLok.com account (for retrieving and managing iLok licenses)

EX66	
Type:	MTM studio reference monitor
LF Driver:	two 6" custom linear-piston woofers
HF Driver:	
1" titanium dome	e tweeter with waveguide
Frequency Respons	e:
37Hz to 20kHz	(near-field acoustic response)
Crossover:	
2.56kHz, 4th-ord	der Linkwitz-Riley maximally flat alignment

LF Amplifier Power: 104 watts (IHF dynamic power into 4 ohms) HF Amplifier Power:
104 watts (IHF dynamic power into 4 ohms) <0.1% (@ 50 watts into 4 ohms) Peak SPI @ 1 Meter 109dB (115dB for a stereo pair) peal

Analog Inputs: XLR balanced (20k ohms), 1/4" TRS balanced (20k ohms) Digital Inputs: S/PDIF (75 ohms), AES/EBU (110 ohms), built-in phase lock loop (PLL) with low jitter (<250 psec p-p): S/PDIF receiver can lock to an incoming clock of up to Acoustic Space (full half quarter)

Mid-Range Boost (flat, +2dB); Low Cutoff (37Hz, 80Hz, 100Hz); High Frequency Adjustment (+2dB, flat, -2dB)) Polarity: positive signal @ + input produce outward LF cone displacement

Input Sensitivity: -10dBV pink noise (90dB SLP (c-weighted) @ 1 meter; variable -10dBV pink noise @ input yields tection: RF interference, output current ting, over temperature, turn-on/off transient, subsonic filte

Indicator: power on/off indicator on front panel wer: user selectable for 100V ~50/60Hz, 115V ~50/60Hz, and 230V ~50/60Hz painted high acoustic efficiency MDF

Dimensions: 19" (482.6 mm) x 8.25" (209.6 mm) x 9.5" (241.3 mm) 24.65 lbs (11.18 kg

Fast Track Pro

Mic Inputs (A/D) Input Impedance 2.7k ohms, unbalanced: 5.4k ohms, balanced Max Input Level: from +2 to -40dBu @ max gain (no pad) from +24dBu @ min gain (with pad), Channel-to-Channel Crosstalk: Signal-to-Noise Ratio -101dB, A-weighted 101dB, A-weighted 0.00357% (-89dB) @ -1dBFS, 1kHz Dynamic Range: 20Hz to 20kHz, +/-0.1dB Preamp Gain: >40dB

nstrument Inputs (A/D) Input Impedance >220k ohms, unbalanced rom +22dBV @ min gain (with pad), Max Input Level to -42dBV @ max gain (no pad) -102dB, A-weighted Signal-to-Noise Ratio: Dynamic Range: THD+N: 102dB, A-weighted 0.00386% (-88dB) @ -1dBFS, 1kHz

Frequency Response: 20Hz to 20kHz, +/-0.1dB @ 48kHz sample rate Preamp Gain: -20dB pac

Line Inputs (A/D) 20k ohms, balanced and unbalanced Input Impedance Max Input Level: +4.1dBu, balanced; +1.9dBV, unbalanced Channel-to-Channel <-110dB Crosstalk: Signal-to-Noise Ratio -102dB, A-weighted 102dB, A-weighted Dynamic Range: THD+N: 0.00345% (-89dB) @ -1dBFS, 1kHz requency Response: 20Hz to 20kHz, +/-0.1dB @ 48kHz sample rate

20Hz to 40kHz, +/-0.1dB @ 96kHz sample rate 20dB pad Input Inserts (analog +6dBV. unbalanced

Max Level Return +6dBV, unbalanced 0.00281% (-91dB) @ -1dB from THD+N: max level, 1kHz 20Hz to 40kHz, +/-0.1dB Frequency Response Line Outputs (D/A)

Output Impedance 150 ohms unbalanced, 300 ohms balanced Max Output Level: +1.8dBV. unbalanced

Channel-to-Channe <-1084B Signal-to-Noise Ratio -104dB, A-weighted Dynamic Range: 04dB, A-weighted 0.00446% (-87dB) @ -1dBFS. 1kHz THD+N: 20Hz to 20kHz, +/-0.3dB @ 48kHz sample rate

Headphone Output (D/A)

Max Output:
-2.5dBV @ THD <0.05% into 32 ohms Working Range: 32 to 600 ohms

20Hz to 40kHz, +/-0.5dB @ 96kHz sample rate

Fast Track Ultra Mic Inputs (A/D)

Dynamic Range: 103dB, A-weighted (min gain, no pad)

1.7k ohms, unbalanced: 3.4k ohms, balanced Max Input Level: from +21dBu @ min gain (with pad) to -50dBu @ max gain (no pad) Channel-to-Channel Crosstalk: <-110dB Signal-to-Noise Ratio: -103dB, A-weighted (min gain, no pad)

-20dB Line Inputs (A/D) Input Impedance 28k ohms, balanced: 20k ohms, unbalanced Max Input Level: +3.4dBu, balanced/+1.2dBV, unbalanced Channel-to-Channel Crosstalk: -103dB, A-weighted Signal-to-Noise Ratio: 103dB, A-weighted Dynamic Range: THD+N: 0.002% (-94dB) @ -1dBFS, 1kHz Frequency Response 20Hz to 20kHz, +/-0.1dB Input Inserts (analog) +6dBV, unbalanced Max Level Send: Max Level Return +6dBV. unbalanced THD+N: 0.002% (-94dB) @ -1dB from max level, 1kHz 20Hz to 20kHz, +/-0.1dB Frequency Response: Line Outputs (D/A) Output Impedance: 150 ohms, unbal iced; 300 ohms, balanced Max Output Level: +10dBu, balanced +1.8dBV. unbalanced Channel-to-Channel Crosstalk: <-100dB Signal-to-Noise Ratio: -104dB, A-weighted Dynamic Range: 104dB, A-weighted THD+N 0.002% (-94dB) @ -1dBFS, 1kHz Frequency Response 20Hz to 20kHz. +/-0.1dB Headphone Output (D/A) Max Output: 0dBV @ THD <0.05% into 32 ohms Channel-to-Channel Crosstalk: <-55dB

>60dB

J+N: 0.005% (-86dB) @ -1dBFS, 1kHz (min gain, no pad)

20Hz to 20kHz, +/-0.1dB @ 48kHz sample rate

Frequency Response Working Range: Fast Track Ultra 8R

Frequency Response:

-40dBV @ max gain (no pad)

Signal-to-Noise Ratio: -103dB, A-weighted (min gain, no pad)

Dynamic Range: 103dB, A-weighted (min gain, no pad)

Instrument Inputs (A/D)

Frequency Response

Preamp Gain:

Input Impedance

THD+N:

Preamp Gain:

Mic Inputs (A/D)

Input Impedance: 1.6k ohms unbalanced, 3.2k ohms balanced Max Input Level: from +28dBu, min gain (with pad) to -45dBu, max gain (no pad) Channel-to-Channel Crosstalk <-110dB Signal-to-Noise Ratio

20Hz to 20kHz, +/-0.7dB

24 to 600 ohms

-106dB, A-weighted (min gain, no pad) Dynamic Range: 106dB, A-weighted (min gain, no pad) THD+N: 0.0028% (-91dB) @ -1dBFS, 1kHz (min gain, no pad) Frequency Response:

20Hz to 20kHz, +/-0.1dB >53dB (no pad) Preamp Gain: Instrument Inputs (A/D) Input Impedance: 1M ohm, unbalanced from +12dBV @ min gain (no pad) to -41dBV @ max gain (no pad)

Signal-to-Noise Ratio: -107dB, A-weighted (min gain, no pad) Dynamic Range: 107dB, A-weighted (min gain, no pad)

THD+N: 0.003% (-90dB) @ -1dBFS, 1kHz (min gain, no pad)

Frequency Response:
20Hz to 20kHz, +/-0.1dB @ 48kHz sample rate Preamp Gain: Pad: -20dB pad

Line Inputs (A/D) 18k ohms, balanced: 9k ohms, Input Impedance unbalanced Max Input Leve +20.2dBu balanced/unbalanced Channel-to-Channel Crosstalk: <-110dB Signal-to-Noise Ratio -106dB. A-weighted 106dB, A-weighted Dynamic Range: 0.003% (-90dB) @ -1dBFS. 1kHz THD+N-Frequency Response 20Hz to 20kHz, +/-0.1dB Input Inserts (analog) Max Level Send: +2.2dBV_unbalanced Max Level Return: +2.2dBV, unbalanced

THD+N: level, 1kHz 0.0022% (-93dB) @ -1dB from max Frequency Respons 20Hz to 20kHz, +/-0.1dB Line Outputs (D/A)

470 ohms, unbalanced: 940 ohms, balanced Max Output Level: +11.8dBV, unbalance Channel-to-Channel

Output Impedance:

Crosstalk <-110dB Signal-to-Noise Ratio: 104dB, A-weighted 104dB, A-weighted Dynamic Range: 0.0025% (-92dB) @ -1dBFS, 1kHz THD+N: Frequency Response

Headphone Output (D/A)

Max Output Level: +2.8dBV @ THD <0.003% into 32 ohms 0.002% (-94dB) @ -1dBFS, 1kHz (min gain, no pad) 20Hz to 20kHz, +/-0.1dB Power into Ohms Channel-to-Channe Crosstalk <-85dB 104dB, A-weighted Signal-to-Noise Ratio 104dB, A-weighte Dynamic Range: 1M ohm, unbalanced 20Hz to 20kHz, +/-0.1dE Frequency Res from +14dBV @ min gain (no pad) to Working Range:

Fast Track USB Mic Input (A/D) 1M ohm -2.2dBu (0.6 Vrms), min gair Max Input Level -100dB, A-weighted Signal-to-Noise Ratio Dynamic Range: THD+N: 100dB. A-weigh 0.0053% (-86dB) @ -1dBFS, 1kHz 22Hz to 22kHz, +/-0.1dB Frequency Response Preamp Gain 45dB Instrument Input (A/D) Input Impedance 500k ohms +8.3dBV (2.6 Vrms Max Input Leve Signal-to-Noise Ratio -97dB, A-weighted Dynamic Range: 97dB, A-weighted 0.0049% (-86dB) @ -1dBFS, 1kHz quency Response: 22Hz to 22kHz, +/-0.3dB @ 48kHz sample rate

Line Input (A/D)

Input Impedance 20k ohms, balanced; 10k ohms, unbalanced Max Input Level: Signal-to-Noise Ratio -98dB, A-weighted Dynamic Range: 98dB, A-weighted 0.0045% @ -1dBFS, 1kHz HD+N requency Response: 22Hz to 22kHz, +/-0.2dB @ 48kHz sample rat

Line Outputs (D/A)

Output Impedance 240 ohms, unbalanced +2.0dBV (1.2 Vrms), unbalanced /lax Output Leve Channel-to-Channel <-92dB Signal-to-Noise Ratio 105dB, A-weighted 105dB, A-weighted Dynamic Range: 0.004% (-88dB) @ -1dBFS, 1kHz THD+N: Frequency Response: 22Hz to 22kHz, +/-0.1dB @ 48kHz sample rate

32 to 600 ohms

Headphone Output (D/A) Max Output: -2.3dBV @ THD <0.05% into 32 ohms

Working Range: FireWire Solo

Mic Inputs Max Input Level, Min Gain: -1.7dBu, balanced Signal-to-Noise Ratio: -101dB, A-weighted Dynamic Range: 101dB, A-weighted THD+N-

0.0049% (-86dB) @ -1dBFS, 1kHz Frequency Response: 20Hz to 20kHz, +/-0.25dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.3dB @ 96kHz sample rate Preamp Gain: 40dB

Phantom Power 48V Instrument Input Input Impedano 280k ohms

Max Input Level, Min Gain: +12.2dBV, unbalanced -101dB, A-weighted Signal-to-Noise Ratio: 101dB, A-weigh amic Range: 0.0079 % (-82dB) @ -1dBFS. 1kHz

requency Response: 20Hz to 20kHz, +/-0.25dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.25dB @ 96kHz sample rate Preamp Gain:

Line Inputs +2.2dBV (1.285 Vrms), unbalanced Max Input Leve

Channel-to-Channel Crosstalk: <-110dB@1kHz Signal-to-Noise Ratio -102dB, A-weighted Dynamic Range: 102dB, A-weighted THD+N: 0.005% (-86dB) @ -1dBFS, 1kHz Frequency R

20Hz to 20kHz, +/-0,2dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.3dB @ 96kHz sample rate Line Outputs

Max Output Level +10.1dBu, balanced +1.9dBV, unbalanced -106dB, A-weighter Signal-to-Noise Ratio Dynamic Range: 106dB. A-weighte

0.0024 % (-92.5dB) @ -1dBFS, 1kHz quency Response: 20Hz to 20kHz, +/-0.2dB @ 48kHz sample rate 20Hz to 40kHz, +/-0.3dB @ 96kHz sample rate

Headphone Output Max Output:
-2.0dBV @ THD+N <0.02% into 32 ohms

Working Range: Hvbrid Digidesign-qualified Windows XP- or Mac-based Pro Tools system running Pro Tools 7.0 or higher software DVD-ROM drive for installation (download-only also available)

iLok USB Smart Key (sold separately), Internet access and a free iLok.com account (for retrieving and managing iLok licenses)

IE-10

Frequency Response: 20Hz to 13kHz Input Sensitivity 115dB/mW Input Impedance 13 ohms (@ 1kHz

Cable Length:	46"
Weight:	0.5 oz (.017 kg) w/ cable
IE-20 XB	
Isolation:	16dB
Frequency Response:	20Hz to 16kHz
Input Sensitivity:	119dB/mW
Input Impedance:	11 ohms (@ 1kHz)
Cable Length:	46"
Weight:	0.7 oz (.019 kg) w/ cable
IE-30	
Isolation:	26dB
Frequency Response:	20Hz to 16kHz
Input Sensitivity:	119dB/mW
Input Impedance:	21 ohms (@ 1kHz)
Cable Length:	46"
Weight:	0.6 oz (.017 kg) w/ cable
IE-40	
Isolation:	26dB
Frequency Response:	10Hz to 17kHz
Input Sensitivity:	117dB/mW
Input Impedance:	32 ohms (@ 1kHz)
Cable Length:	46"

impOSCar

Intel Pentium III 300MHz with 64MB RAM Windows 98, 98SE, Me, 2000, XP

Macintosh

PowerMac G4 with 64MB RAM
Mac OS 9 or later (CarbonLib 1.5 or higher required); OS X

0.6 oz (.017 kg) w/ cable

RTAS, VST 2.0, MAS, AU KeyStudio 49i

Line Inputs

+2.1dBV (1.3 Vrms), unbalanced Max Input: 94dB, A-weighted Signal-to-Noise Ra namic Range: 94dB. A-weighted 0.005% (-86dB) @ -1dBFS, 1kHz THD+N Frequency Respons +/-0.5dB, 20Hz to 20kHz 10k ohms, unbalanced Mic Input

Max Input L

-4.4dBV (0.6 Vrms), balanced @ Min Gain: -97dB, A-weighted Signal-to-Noise Ratio Dynamic Range: 97dB, A-weighted 0.005% (-86dB) @ -1dBFS, 1kHz requency Response +/-0.5dB. 20Hz to 20kHz >42dB adjustable range Preamp Gain 3.3k ohms, balanced Impedance:

Instrument Input +11 6dBV (3.8 Vrms) unbalanced Max Input Level Min Gain: Signal-to-Noise Ratio 94dB, A-weighted amic Range: 94dR A-weighted 0.008% (-82dB) @ -1dBFS, 1kHz THD+N: +/-0.6dB, 20Hz to 20kHz requency Response >42dB adjustable range eamp Gai

500k ohms, unbalance

Impedance Line Outputs

+1.9dBV (1.25 Vrms), unbalanced Max Output Signal-to-Noise Ratio -95dB, A-weighted 95dB, A-weighted Dynamic Range: 0.0035% (-89dB) @ -1dBFS, 1kHz HD+N: Frequency Response +/-0.2dB, 20Hz to 20kHz Impedance 240 ohms, unbalanced Headphone Outputs

Max Output: -1.9dBV @ <0.075% THD+N into 32 ohms

Working Headphone 32 to 600 ohms

KikAxxe Windows

Windows XP (SP2), Vista 32 512MB RAM (1GB for Vista) 100MB available disk space

Universal binary for Intel- and PowerPC-based Mac computers Mac OS X 10.3.9 or higher, 1GHz 512MB RAM

200MB available disk Compatibility Luna II

large-diaphragm condense Type: 3-micron thick Mylar with evaporated gold cardioid 20Hz to 20kHz Frequency Response: class A FET, with output transformer Sensitivity: -36dB @ 1kHz, 0dB=1V/Pa (-36dBV) 130dB for 0.5% THD 10dB switchable

Max SPL: F Rolloff Filter 6dB/octave @ 125Hz, switchable 14dB A-weighted quivalent Noise Leve Output Impedance 200 ohms Recommended Load >1.000 ohms Impedance: 3-pin male XLR Power Require >48V phantom por

8.25" (209.55 mm) x 3" (76.2 mm) x 2" (50.8 mm)

1.4 lbs (.64 kg)

Weight: M-Tron Pro

Windows

Intel Pentium IV 1GHz with 512MB RAM 5GB available disk space

Windows XP SP2, Vista

5GB available disk space OS X 10.4

Compatibility RTAS, VST 2.0, MAS, AU

ASIO-compatible sound card for standalone operation

Macintosh 1.25GHz PowerMac G4 with 512MB RAM or Intel-based Mac

Wireless Platform:		Crosstalk:	<-100dB	DVD-ROM or equivalent	optical drive for software installation
	2.4GHz radio frequency	Signal-to-Noise Ratio: Dynamic Range:	-102dB, A-weighted 102dB, A-weighted	1024 x 768 or higher dis	play resolution
Operating Range Under Typical Conditions:	30 feet (10 meters)	Dynamic Range: THD+N:	0.0025% (-92dB) @ -1dBFS, 1kHz	Available USB port for iL for registration, software	ok USB Smart Key, plus Internet acces updates and activations
Receiver:	9VDC, 500mA	Frequency Response:		ProFire 610	
Transmitter:	2 AA batteries, or 9VDC, 500mA		1dB @ 48kHz sample rate	Line Inputs (balanced)	
Battery Life: power (variable based o	exceeds 20 hours of continuous	Variable Gain: Line Inputs 1-8 (A/D, pre	56dB	Frequency Response:	+/-0.1dB, 20Hz to 22kHz
MobilePre USB	on battery brand,	Input Impedance:	20k ohms, balanced		+/-0.2dB, 20Hz to 80kHz (192kHz)
Mic Input (A/D)		Max Input Level:	+20dBu, balanced = 0dBFS	Signal-to-Noise Ratio:	-108dB, A-weighted
Max Input Level:	-5.2dBu		+17.8dBV, unbalanced = 0dBFS	Dynamic Range: THD+N:	108dB, A-weighted 0.002% (-94dB), 1kHz, -1dBFS
Signal-to-Noise Ratio:	-88dB, A-weighted	Channel-to-Channel Crosstalk:	<-82dB	Crosstalk:	<-110dB @ 1kHz
Dynamic Range:	88dB, A-weighted	Signal-to-Noise Ratio:	-102dB, A-weighted	Max Input Level:	+16.1dBu, typical
THD+N:	0.013% (-78dB) @ -1dBFS, 1kHz	Dynamic Range:	102dB, A-weighted	Input Impedance:	>20k ohms, balanced
Frequency Response:	+/-0.8dB, 20Hz to 20kHz	THD+N:	0.0025% (-92dB) @ -1dBFS, 1kHz	Mic Inputs (balanced; @	
ine Outputs (D/A) Max Output Level:	+2.8dBV (1.387 Vrms), unbalanced	Frequency Response:	4 IB G 48111	Frequency Response:	+/-0.2dB, 20Hz to 22kHz
Channel-to-Channel	12.0dBV (1.007 VIIII3), uribalariced		1dB @ 48kHz sample rate 1dB @ 96kHz sample rate	Signal-to-Noise Ratio:	-108dB, A-weighted
Crosstalk:	-94dB @ 1kHz	Mix Line Inputs 9-10 (line	•	Dynamic Range: THD+N:	108dB, A-weighted 0.0022% (-93dB) @ -1dBFS, 1kHz
Signal-to-Noise Ratio:	-84dB, A-weighted	Max Input Level:	+20dBu balanced = 0dBFS	Crosstalk:	<-110dB @ 1kHz
Dynamic Range: THD+N:	82dB, A-weighted 0.01% (-80dB) @ -1dBFS, 1kHz	Channel-to-Channel		Max Input Level:	from +28dBu @ min gain (with pad
Frequency Response:	+/-0.1dB, 20Hz to 20kHz	Crosstalk:	<-80dB	to -45dBu @ max ga	
leadphone Output (D/A)	in direct, zer z te zere z	Signal-to-Noise Ratio:	-103dB, A-weighted	Input Impedance: Adjustable Gain:	3.7k ohms, balanced (no pad) >53dB (no pad)
Output Impedance:	75 ohms	Dynamic Range: THD+N:	103dB, A-weighted 0.004% (-88dB) @ -1dBFS, 1kHz	Pad:	-20dB
Max Output:		Frequency Response:	0.00470 (00dB) @ 1dB1 0, 1k12		lanced; @ min gain, no pad)
-6dBV into 32 ohms @		20Hz to 20kHz, +/-0	2dB @ 48kHz sample rate	Frequency Response:	+/-0.2dB, 20Hz to 22kHz
Working Range:	24 to 600 ohms		4dB @ 96kHz sample rate	Signal-to-Noise Ratio:	107dB, A-weighted
/licroTrack II		Channel Returns 1-8 (D/		Dynamic Range:	107dB, A-weighted
ieneral	06 +- 000 librar @ 44.1 40lil la	Max Output Level:	0dBFS = +20dBu balanced	THD+N:	0.0025% (-92dB), 1kHz, -1dBFS
MP3 Recording: PCM Recording:	96 to 320 kbps @ 44.1 or 48kHz	Channel-to-Channel	0dBFS = +11.8dBV unbalanced	Crosstalk:	<-110dB @ 1kHz
16- or 24-bit @ 44.1, 48	8, 88.2 or 96kHz	Channel-to-Channel Crosstalk:	<-85dB	Max Input Level: to -39dBV @ max ga	from +14.5dBV @ min gain (no pad ain (no pad)
Storage Capacity:		Signal-to-Noise Ratio:	-105dB, A-weighted	Input Impedance:	1M ohm, unbalanced
	rate and storage medium	Dynamic Range:	105dB, A-weighted	Adjustable Gain:	>53dB without pad
/8" Mic Input (A/D)	0.0k ahma	THD+N:	<0.006% (-84dB) @ -1dBFS, 1kHz	Line Outputs (balanced)	
Input Impedance: Max Input Level, Min Gain:	2.2k ohms -3.6dBV	Frequency Response: 22Hz to 22kHz +/-0	2dB @ 48kHz sample rate	Frequency Response:	+/-0.1dB, 20Hz to 22kHz
Channel-to-Channel	5.50D¥		3dB @ 96kHz sample rate	0: 1: 1: - :	+/-0.2dB, 20Hz to 80kHz (192kHz)
Crosstalk:	<-95dB	Mix Return 9-10 (D/A)	<u> </u>	Signal-to-Noise Ratio: Dynamic Range:	-108dB, A-weighted 108dB, A-weighted
Signal-to-Noise Ratio:	-101dB, A-weighted	Max Output Level:	0dBFS = +20dBu balanced	THD+N:	0.0016% (-96dB) @ -1dBFS, 1kHz
Dynamic Range:	101dB, A-weighted		0dBFS = +11.8dBV unbalanced	Crosstalk:	<-105dB @ 1kHz
THD+N:	0.003% (-90dB)	Channel-to-Channel	< 05 ID	Max Output Level:	+15.9dBu, balanced, typical
Frequency Response: Preamp Gain:	20Hz to 20kHz, +/-0.5dB >52dB	Crosstalk: Signal-to-Noise Ratio:	<-85dB -105dB, A-weighted	Output Impedance:	940 ohms, balanced
electret condenser power er		Dynamic Range:	105dB, A-weighted	Headphone Outputs (@	max volume into 32 ohm load)
/4" Mic/Line Inputs (A/D)		THD+N:	<0.004% (-88dB) @ -1dBFS, 1kHz	Frequency Response:	+/-0.1dB, 20Hz to 22kHz
	>5.3k ohms	Frequency Response:		Signal-to-Noise Ratio:	-108dB, A-weighted
Max Input Level @ Min Gain			2dB @ 48kHz sample rate	Dynamic Range:	108dB, A-weighted
+4.3dBu, balanced/unb	palanced		5dB @ 96kHz sample rate	THD+N: Max Output Level	0.003% (-90dB), 1kHz, -1dBFS
Channel-to-Channel Crosstalk:	<-95dB	Analog Mixer Specification		into 32 Ohms:	+2.9dBV, typical
Signal-to-Noise Ratio:	-101dB, A-weighted	Input Impedance:	nput to chan insert output) 4.5k ohms	Power into Ohms:	60mW into 32 ohms
Dynamic Range:	101dB, A-weighted	Max Input Level:	+13dBu to -43dBu, balanced	Output Impedance:	75 ohms
THD+N:	0.0027% (-91dB)	Channel-to-Channel	1 Todas to Todas , Salariosa	Load Impedance:	24 to 600 ohms
Frequency Response:	JD @ 401.1 lelete	Crosstalk:	<-100dB	Crosstalk:	<-75dB @ 1kHz
	dB @ 48kHz sample rate IB @ 96kHz sample rate	Signal-to-Noise Ratio:	-115dB, A-weighted	ProFire 2626	
Preamp Gain:	>52dB	Dynamic Range: THD+N:	115dB, A-weighted	Line Inputs (balanced)	+/-0.1dB, 20Hz to 22kHz (48kHz)
Phantom Power:	7 0245	Frequency Response:	<0.001% (-100dB) @ +4dBu, 1kHz 20Hz to 60kHz, +/-0.1dB	Frequency Response:	+/-0.1dB, 20Hz to 22kHz (46kHz) +/-0.4dB, 20Hz to 80kHz (192kHz)
phantom power enabled	d (switchable on/off)	Variable Gain:	56dB	Dynamic Range:	110dB, A-weighted
ine Outputs (D/A)		Line Inputs (analog - inpu		Signal-to-Noise Ratio:	-110dB, A-weighted
Output Impedance:	465 ohms	Input Impedance:	20k ohm, balanced	THD+N:	0.0008% (-102dB), 1kHz, -1dBFS
Max Output Level: Channel-to-Channel	+6dBV, unbalanced	Max Input Level:	+20dBu, balanced	Crosstalk:	-120dB @ 1kHz
Crosstalk:	<-105dB	+17.8dBV, unbalance	d	Max Input Level:	+19.6dBu, typical
Signal-to-Noise Ratio:	-99dB, A-weighted	Channel-to-Channel	< 90dB	Input Impedance:	>20k ohms, balanced
Dynamic Range:	99dB, A-weighted	Signal-to-Noise Ratio:	<-80dB -115dB, A-weighted	Mic Inputs (balanced; @	
THD+N:	0.0044% (-87dB) @ -1dBFS, 1kHz	Dynamic Range:	115dB, A-weighted	Frequency Response: Dynamic Range:	+/-0.1dB, 20Hz to 22kHz 109dB, A-weighted
Frequency Response:	dB @ 48kHz sample rate	THD+N:	<0.001% (-94dB) @ +4dBu, 1kHz	Signal-to-Noise Ratio:	-109dB, A-weighted
	IB @ 96kHz sample rate	Frequency Response:	20Hz to 60kHz, +/-0.1dB	THD+N:	0.0011% (-99dB), 1kHz, -1dBFS
leadphone Output (D/A)		Line Outputs (analog)		Crosstalk:	<-120dB @ 1kHz
Max Output:		Output Impedance:	360 ohms balanced	Max Input Level:	+6.5dBu (no pad)
0dBV @ THD <0.075%		Max Output Level:	+20dBu balanced	Input Impedance:	3.7k ohms, balanced (no pad)
Working Range:	16 ohms to 600 ohms	Frequency Response:	+17.8dBV unbalanced 20Hz to 60kHz, +/-0.1dB	Adjustable Gain:	>53dB without pad
Vinimonsta:Melohr	man	Headphone Output (anal	· .	Pad:	-20dB lanced; @ min gain, no pad)
Aicrosoft Windows	O MAD DAM	Max Output:	og/	Frequency Response:	+/-0.1dB, 20Hz to 22kHz
Intel Pentium III 1GHz with 6 Windows 98SE, Me, 2000,		+7.2dBu (5 Vpp) @ 1	HD <0.01% into 32 ohms	Dynamic Range:	108dB, A-weighted
Macintosh	, Ai	Impedance:	38 ohms	Signal-to-Noise Ratio:	-108dB, A-weighted
1.25GHz PowerMac G4		Working Range:	24 to 600 ohms	THD+N:	0.0014% (-97dB), 1kHz, -1dBFS
128MB of RAM		Crosstalk:	<-85dB	Crosstalk:	<-110dB @ 1kHz
	Lib 1.5 or higher required); OS X	Inserts (analog) Output Impedance:	150 ohms unbalanced	Max Input Level:	13.7dBV, typical (no pad)
Compatibility		Input Impedance:	20k ohms unbalanced	Input Impedance:	1M ohm unbalanced >53dB (no pad)
RTAS, VST 2.0, MAS, AU		Max Send/Return Level:	+12dBV unbalanced	Adjustable Gain: Pad:	-20dB (no pad)
Nova		Aux Sends (analog)		Line Outputs (balanced)	
Type:	large-diaphragm condenser	Output Impedance:	150 ohms	Frequency Response:	+/-0.1dB, 20Hz to 22kHz
Pattern:	cardioid 20Hz to 18kHz	Nominal Send Level:	+12dBV unbalanced		+/-0.6dB, 20Hz to 80kHz (192kHz
Frequency Response: Preamp Topology:	class A FET preamp	Aux Returns (analog)	001 1 1 1	Dynamic Range:	110dB, A-weighted
Sensitivity:	16 mV/Pa (-36dBV)	Input Impedance:	20k ohms balanced	Signal-to-Noise Ratio:	-110dB, A-weighted
Max SPL for 0.5% THD:	128dB	Max Level:	+20dBu balanced +17.8dBV unbalanced	THD+N:	0.0013% (-98dB), 1kHz, -1dBFS
Equivalent Noise Level:	14dB, A-weighted	Oddity	. 17.5GD v Gribala iled	Crosstalk: Max Output Level:	<-110dB @ 1kHz +20.2dBu, balanced, typical
Output Impedance:	200 ohms	Oddity Windows		Output Impedance:	300 ohms, balanced
Recommended Load	>1,000	Intel Pentium III 300MHz	with 64MB DAM		ax volume into 32 ohm load)
Impode	>1,000 ohms	Windows 98SE, Me, 200		Frequency Response:	+/-0.4dB, 20Hz to 22kHz
Impedance:			-,		
Impedance: Connector:	3-pin male XLR			Dynamic Range:	110dB, A-weighted
Impedance:	3-pin male XLR 48V phantom power wire mesh grille and body, all brass	Macintosh PowerMac G4 with 128M	//B RAM	Dynamic Range: Signal-to-Noise Ratio:	110dB, A-weighted -110dB, A-weighted

7.25" (185 mm) x 1.9" (52 mm)

+13dBu to -43dBu, balanced

1.43 lbs (.65 kg)

<-100dB

Weight:

NRV10

Max Input Level: Channel-to-Channel

Digital Audio Interface Specifications
Mic Inputs 1-4 (A/D, pre-EQ, min gain)
Input Impedance: 4.5k ohms

Compatibility RTAS, VST 2.0, MAS, AU

Minimum 1GB of RAM

Pro Tools M-Powered 8

Supported M-Audio hardware peripheral
Digidesign-qualified Windows Vista- (32-bit Business or
Ultimate), Windows XP-, or Mac OS X 10.5.5-based computer
(visit the Pro Tools M-Powered section of www.digidesign.com/
support for details)

Power into Ohms:	+6.8dBV, typical 150mW into 32 ohms	Max Output: Signal-to-Noise Ratio:	+2.1dBV (1.27 -95dB, A-weigh
Output Impedance:	75 ohms	Dynamic Range:	93dB, A-weigh
Load Impedance:	24 to 600 ohms	THD+N:	0.0045% (-870
ProFire Lightbridge	•	Frequency Response:	+/-0.2dB, 20H
ine Outputs Output Impedance:		Impedance: Headphone Outputs	240 ohms, uni
300 ohms, balanced; 1		Max Output:	0550/:00
Max Output Level:	+10.2dBu (2.5 Vrms), balanced +2.0dBV (1.26 Vrms), unbalanced	-2.2dBV @ THD+N <0 Working Headphone	7.055% Into 32 t
Channel-to-Channel		Impedance:	32 to 600 ohm
Crosstalk: Signal-to-Noise Ratio:	<-110dB @ 1kHz -109dB, A-weighted	ProKeys Sono 88	
Dynamic Range:	109dB, A-weighted	Line Inputs Max Input:	+2.1dBV (1.3 \
THD+N: 1kHz	0.00127 % (-98dB) @ -1dBFS,	Signal-to-Noise Ratio:	-94dB, A-weigl
Frequency Response:		Dynamic Range: THD+N:	94dB, A-weigh 0.005% (-86dB
	dB @ 48kHz sample rate dB @ 96kHz sample rate	Frequency Response:	+/-0.2dB, 20H
eadphone Output	ub @ 90kHz sample rate	Impedance:	10k ohms, unb
Max Output:	. 00 00 000 TUD N 4111	Mic Input Max Input Level, Min Gain:	-2.2dBu (0.6 Vi
-3.0dBV (1.71 Vrms) in Working Range:	to 32 ohms @ <0.02% THD+N, 1kHz 32 to 600 ohms	Signal-to-Noise Ratio:	-94dB, A-weigh
Impedance:	75 ohms	Dynamic Range:	94dB, A-weigh
Crosstalk:	<-85dB	THD+N: Frequency Response:	0.005% (-86dE +/-0.2dB, 20H
Signal-to-Noise Ratio: Dynamic Range:	-108dB, A-weighted 108dB, A-weighted	Preamp Gain:	>38dB adjusta
Frequency Response:		Impedance:	5.4k ohms, bala
	dB @ 48kHz sample rate dB @ 96kHz sample rate	Instrument Input Max Input Level, Min Gain:	+10.0dBV (3.2
ProjectMix I/O	ab @ ook iz sample rate	Signal-to-Noise Ratio:	-94dB, A-weigh
lic Inputs		Dynamic Range:	94dB, A-weigh
Input Impedance:	3.4k ohms	THD+N: Frequency Response:	0.009% (-81dE +/-0.2dB, 20H;
Max Input Level, Min Gain: Channel-to-Channel	-3dBu, balanced	Preamp Gain:	>34dB adjusta
Crosstalk:	<-110dB @ 1kHz	Impedance:	1M ohm, unbal
Signal-to-Noise Ratio: Dynamic Range:	-104dB, A-weighted 104dB, A-weighted	Line Outputs	±0.14DV/4.05
THD+N:		Max Output: Signal-to-Noise Ratio:	+2.1dBV (1.27 -95dB, A-weigh
0.00188% (-94.6dB) (Dynamic Range:	93dB, A-weigh
Frequency Response: Preamp Gain:	20Hz to 20kHz, +/-0.1dB 55dB	THD+N: Frequency Response:	0.0045% (-87d +/-0.2dB, 20H
Phantom Power:	48V DC @ 16mA	Impedance:	240 ohms, unb
nstrument Input		Headphone Outputs	,
Input Impedance: 560k ohms, balanced:	; 280k ohms, unbalanced	Max Output: -2.2dBV @ THD+N <0	0.055% into 32 d
Max Input Level, Min Gain:	1.9dDV uphalanaed	Working Headphone	
+14dBu, balanced; +1 Signal-to-Noise Ratio:	-100dB, A-weighted	Impedance:	32 to 600 ohm
Dynamic Range:	100dB, A-weighted	Transit Line/Mic Input (A/D)	
THD+N: 0.00243% (-92.3dB) (@ -1dBES 1kHz	Max Input Level:	-1.9dBV (0.808
Frequency Response:		Channel-to-Channel	< 400 ID @ 41
	dB @ 48kHz sample rate	Crosstalk: Signal-to-Noise Ratio:	<-100dB @ 1k -99dB, A-weigh
ine Inputs Input Impedance:		Dynamic Range:	99dB, A-weigh
20k ohms, balanced; 1		THD+N:	0.00504% (-86
Max Input Level @ Min Gair +10dBu balanced/+7.8		Frequency Response: 20Hz to 22kHz, +/-0.1	dB @ 48kHz sar
Channel-to-Channel	< 110-ID @ 1U L	Line/Headphone Output (I	
Crosstalk: Signal-to-Noise Ratio:	<-110dB @ 1kHz -104dB, A-weighted	Max Output Level: Channel-to-Channel	-0.4dBV (0.953
Dynamic Range:	104dB, A-weighted	Crosstalk:	<-96dB
		Signal-to-Noise Ratio: Dynamic Range:	-101dB, A-weig
THD+N: 0.00201% (-94.1dB) (@-1dBES 1kHz		
0.00201% (-94.1dB) (Frequency Response:		THD+N:	
0.00201% (-94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1	dB @ 48kHz sample rate	THD+N: 0.00453% (-87.1dB) @	
0.00201% (-94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.16		THD+N:	9-1dBFS, 1kHz
0.00201% (-94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 ine Outputs Output Impedance:	dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response:	9-1dBFS, 1kHz
0.00201% (-94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 ine Outputs Output Impedance: 300 ohms, balanced; 1	dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs	9 -1dBFS, 1kHz dB @ 48kHz sar
0.00201% (-94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1(ine Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1	dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output:	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala
0.00201% (-94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.10 ine Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level:	dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range:	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig
0.00201% (94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1(ine Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio:	dB @ 48kHz sample rate dB @ 96kHz sample rate 150 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.20 ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response:	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig +/-0.75dB, 20l
0.00201% (-94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1(ine Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range:	dB @ 48kHz sample rate dB @ 96kHz sample rate 150 ohms, unbalanced .8dBV, unbalanced <-109dB	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance:	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig
0.00201% (94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1(ine Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio:	dB @ 48kHz sample rate dB @ 96kHz sample rate 150 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted 110dB, A-weighted	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output:	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig +/-0.75dB, 201 1000 ohms
0.00201% (94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1(ine Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) (Frequency Response:	dB @ 48kHz sample rate dB @ 96kHz sample rate 150 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance:	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig +/-0.75dB, 201 1000 ohms
0.00201% (94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 Line Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) (Frequency Response: 20Hz to 22kHz, +/-0.1	dB @ 48kHz sample rate dB @ 96kHz sample rate 150 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted 110dB, A-weighted	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output:	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig +/-0.75dB, 20l 1000 ohms +0.3dBV into 3 32 ohms
0.00201% (94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 Line Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (-93.8dB) (Frequency Response: 20Hz to 22kHz, +/-0.2) 20Hz to 44kHz, +/-0.2)	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig +/-0.75dB, 20l 1000 ohms +0.3dBV into 3 32 ohms
0.00201% (94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1(ine Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2(1eadphone Output Max Output:	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 1/-0.75dB, 20l 1000 ohms +0.3dBV into 3 20 ohms 32 to 600 ohm
0.00201% (94.1dB) (Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 Line Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (-93.8dB) (Frequency Response: 20Hz to 22kHz, +/-0.2) 20Hz to 44kHz, +/-0.2)	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Outputs Max Outputs	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig +/-0.75dB, 201 1000 ohms +0.3dBV into 3 32 ohms 32 to 600 ohm +12.9dBV (4.4
0.00201% (94.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 Line Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel+C-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (-93.8dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2 Headphone Output Max Output: -2.9dBV @ THD+N Working Range:	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced l8dBV, unbalanced <-109dB -110dB, A-weighted l10dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs	###
0.00201% (-94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (-93.8dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2 Headphone Output Max Output: -2.9dBV @ THD+N < Working Range: ProKeys Sono 61 Line Inputs	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate dB @ 46kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Outputs Signal-to-Noise Ratio: Dynamic Range: THD+N:	9-1dBFS, 1kHz and 48kHz sard 47dBV, unbala 4104dB, A-weig 104dB, A-weig 104dB, 201000 ohms 412.9dBV (4.4-104dB, A-weig 104dB, A-
0.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.10 Line Outputs Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (-93.8dB) @ Frequency Response: 20Hz to 42kHz, +/-0.2 UHz to 44kHz, +/-0.2 Headphone Output Max Output: -2.9dBV @ THD+N <0 Working Range: ProKeys Sono 61	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced l8dBV, unbalanced <-109dB -110dB, A-weighted l10dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range:	2 -1dBFS, 1kHz ard B @ 48kHz sard +7dBV, unbala -104dB, A-weig +/-0.75dB, 201 1000 ohms +0.3dBV into 3 2 ohms 32 to 600 ohm +12.9dBV (4.4 -104dB, A-weig 104dB, A-weig 2 -1dBFS, 1kHz
0.00201% (94.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.2 Headphone Output Max Output: -2.9dBV @ THD+N < White Company Compa	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced .8dBV, unbalanced .8dBV, unbalanced -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate -10,03% into 32 ohms 24 to 600 ohms -10,03% into 32 ohms	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance:	### - 1dBFS, 1kHz
0.00201% (94.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 22kHz, +/-0.1 20Hz to 24kHz, +/-0.2 20Hz to 44kHz, +/-0.2 20Hz to 54kHz, +/-0.2 20Hz	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced l8dBV, unbalanced	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs	###
O.00201% (94.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.2 1eadphone Output Max Output: -2.9dBV @ THD+N Working Range: ProKeys Sono 61 ine Inputs Max Input: Signal-to-Noise Ratio: Dynamic Range:</td <td>dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced .8dBV, unbalanced .8dBV, unbalanced -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate -10,03% into 32 ohms 24 to 600 ohms -10,03% into 32 ohms</td> <td>THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance:</td> <td>###</td>	dB @ 48kHz sample rate dB @ 96kHz sample rate l50 ohms, unbalanced .8dBV, unbalanced .8dBV, unbalanced -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate -10,03% into 32 ohms 24 to 600 ohms -10,03% into 32 ohms	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance:	###
0.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 22kHz, +/-0.1 20Hz to 24kHz, +/-0.2 20Hz to 24kHz, +/-0.2 20Hz to 44kHz, +/-0.2 20Hz to 22kHz, +/-0.1 20Hz to 42kHz, +/-0.2 20Hz to 22kHz, +/-0.1 20Hz to 42kHz, +/-0.2 20Hz to 42kHz, +/-0.2 20Hz to 42kHz, +/-0.1 20Hz to 42kHz, +/-0.2 20Hz to 42kHz, +/-0.1 20H	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate 150 ohms, unbalanced .8dBV, unbalanced <-109dB -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate 24 to 600 ohms +2.1dBV (1.3 Vrms), unbalanced -94dB, A-weighted 94dB, A-weighted 94dB, A-weighted 90.005% (-86dB) @ -1dBFS, 1kHz +/-0.2dB, 20Hz to 20kHz 10k ohms, unbalanced	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBW, unbala -104dB, A-weig 104dB, A-weig 1000 ohms -103dBV into 3 32 ohms 32 to 600 ohm -12.9dBV (4.4 -104dB, A-weig 104dB, A-weig 104dB, A-weig 2 -1dBFS, 1kHz -4/-0.3dB, 20H 150 ohms, unb
0.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 300 ohms, balanced; 1 Max Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) @ Frequency Response: 20Hz to 42kHz, +/-0.2 4eadphone Output Max Output: -2.9dBV @ THD+N <0 Working Range: ProKeys Sono 61 ine Inputs Max Input: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response: Impedance: Mic Input Mic Input Mic Input Mic Input Mic Input Mic Input Level, Min Gain:	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate 150 ohms, unbalanced .8dBV, unbalanced -109dB -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate 24 to 600 ohms +2.1dBV (1.3 Vrms), unbalanced -94dB, A-weighted 94dB, A-weighted	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2c ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Uorking Headphone Output Impedance: Headphone Outputs Max Output: Output Impedance: Vorking Headphone	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBW, unbala -104dB, A-weig 104dB, A-weig 1000 ohms -103dBV into 3 32 ohms 32 to 600 ohm -12.9dBV (4.4 -104dB, A-weig 104dB, A-weig 104dB, A-weig 2 -1dBFS, 1kHz -4/-0.3dB, 20H 150 ohms, unb
0.00201% (94.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 300 ohms, balanced; 1 Max Output Impedance; 3 Max Output Level: +10dBu, balanced; +1 Consatalk: Dynamic Range: THD+N: 0.00205% (93.8dB) @ Frequency Response: 20Hz to 24kHz, +/-0.1 20Hz to 44kHz, +/-0.2 Headphone Output Max Output: -2.9dBV @ THD+N <0 Working Range: ProKeys Sono 61 ine Inputs Max Input: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response:	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 48kHz sample rate dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone	###
0.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2 20Hz to 44kHz, +/-0.2 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.2 20Hz to 24kHz, +/-0.2 20Hz to 44kHz, +/-0.2 20Hz to 44kHz, +/-0.2 20Hz to 42kHz, +/-0.2 20Hz to 22kHz, +/-0.2 20Hz to 42kHz, +/-0.2 20Hz to 42kHz, +/-0.2 20Hz to 22kHz, +/-0.2 20Hz to 22kHz, +/-0.2 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.2 20H	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 48kHz sample rate dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2c ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Uorking Headphone Output Impedance: Headphone Outputs Max Output: Output Impedance: Vorking Headphone Impedance: Pulsar II Type: Diaphragm:	9-1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig 1-70,75dB, 201 1000 ohms +0.3dBV into 3 32 ohms 32 to 600 ohm +12.9dBV (4.4 -104dB, A-weig 104dB, A-weig 104dB, A-weig 3-1dBFS, 1kHz +7-0.3dB, 20H 150 ohms, unb +2.4dBV into 3 75 ohms 32 to 600 ohm small-diaphragr 3/4" diameter, 1
0.00201% (94.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 Max Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) @ Frequency Response: 20Hz to 24kHz, +/-0.1 20Hz to 44kHz, +/-0.2 Headphone Output Max Output: -2.9dBV @ THD+N <0 Working Range: ProKeys Sono 61 ine Inputs Max Input: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response: Impedance: Wic Input Max Input Level, Min Gain: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response: Impedance: Wic Input Max Input Level, Min Gain: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response:	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 1dBFS, 1kHz dB @ 10dB @ 1dBFS, 1kHz dB @ 10dB @ 1dBFS, 1kHz dB @ 1dBFS, 1kHz	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2d ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Vorking Headphone Impedance: Headphone Outputs Max Output: Output Impedance: Vorking Headphone Impedance: Vorking Headphone Impedance: Pulsar II Jype: Diaphragm: Pattern:	### - 1dBFS, 1kHz ### - 1dBFS, 1kHz ### - 1dBFS, 1kHz ### - 104dB, A-weig ### - 1dBFS, 1kHz ## - 1dBFS, 1kHz #
O.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2 20Hz to 22kHz, +/-0.1 20Hz to 42kHz, +/-0.2 20Hz to 42kHz, +/-0.2 20Hz to 44kHz, +/-0.2 20Hz to 52kHz, +/-0.2 20Hz to 52kHz, +/-0.2 20Hz to 52kHz, +/-0.3 20H	dB @ 48kHz sample rate dB @ 96kHz sample rate dS @ 400 dB -110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate dB @ 400 ohms 42 to 600 ohms 42.1dBV (1.3 Vrms), unbalanced -94dB, A-weighted 94dB, A-weighted 94dB, A-weighted 10k ohms, unbalanced -2.2dBu (0.6 Vrms), balanced -94dB, A-weighted 94dB, B, Weighted	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2c ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Uorking Headphone Output Impedance: Headphone Outputs Max Output: Output Impedance: Vorking Headphone Impedance: Pulsar II Type: Diaphragm:	9-1dBFS, 1kHz dB @ 48kHz sar +7dBW, unbala -104dB, A-weig 104dB, A-weig 104dB, A-d 1000 ohms -0.3dBV into 3 32 ohms 32 to 600 ohm -12.9dBV (4.4 -104dB, A-weig 104dB, A-weig 104dB, A-weig 104dB, A-weig 2-1dBFS, 1kHz -4/0.3dB, 20H 150 ohms, unb -2.4dBV into 3 75 ohms 32 to 600 ohm small-diaphragr 3/4" diameter, icardioid solid brass
0.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 300 ohms, balanced; 1 Max Output Impedance: 300 ohms, balanced; 1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.2 deadphone Output Max Output: -2.9dBV @ THD+N <0 Working Range: ProKeys Sono 61 ine Inputs Max Input: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response: Impedance: Impedanc	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 1dBFS, 1kHz dB @ 10dB @ 1dBFS, 1kHz dB @ 10dB @ 1dBFS, 1kHz dB @ 1dBFS, 1kHz	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2c ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Vorking Headphone Impedance: Pulsar II Jype: Diaphragm: Pattern: Backplate: Electronics: Frequency Response:	1dBFS, 1kHz 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig 1000 ohms +0.3dBV into 3 32 ohms 33 ohms 4-104dB, A-weig 3-1dBFS, 1kHz 4-10-3dB, 20Hz 150 ohms, unb 4-2.4dBV into 3 75 ohms 32 to 600 ohm small-diaphragr 3/4" diameter, cardioid solid brass class A FET (tra
0.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2 20Hz to 42kHz, +/-0.2 20Hz to 44kHz, +/-0.2 20Hz to 52kHz, +/-0.1 20Hz to 44kHz, +/-0.2 20Hz to 52kHz, +/-0.2 20Hz to 52kHz, +/-0.3 20H	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: Pulsar II Type: Diaphragm: Pattern: Backplate: Electronics: Frequency Response:	2 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbalai -104dB, A-weig 104dB, A-weig 1-70,75dB, 20t 1000 ohms +0.3dBV into 3 32 ohms 32 to 600 ohm +12.9dBV (4.4 -104dB, A-weig 104dB, A-weig 104dB, A-weig 104dB, A-weig 32 to 600 ohm +2.4dBV into 3 75 ohms 32 to 600 ohm small-diaphragr 3/4" diameter, to cardioid solid brass class A FET (tra 20Hz to 20kHz
0.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 20Hz to 44kHz, +/-0.1 300 ohms, balanced; 1 Max Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel-to-Channel Crosstalk: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (93.8dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.2 4eadphone Output Max Output: -2.9dBV @ THD+N <0 Working Range: ProKeys Sono 61 ine Inputs Max Input: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response: Impedance: Mic Input Max Input Level, Min Gain: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response: Frequency Response: Freamp Gain: Impedance: Instrument Input Max Input Level, Min Gain: Signal-to-Noise Ratio:	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 48kHz sample rate dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 11dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Output Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Working Headphone Impedance: Working Headphone Impedance: Working Headphone Impedance: Pulsar II Type: Diaphragm: Pattern: Backplate: Electronics: Frequency Response: Frequency Response: Frequency Response: Sensitivity: 13.8 mV/Pa (-37dB @ Max SPL:	## dB @ 48kHz sar ## rdBV, unbalai ## rdBV into 3 ## rdBV, unbalai ##
0.00201% (94.1dB) @ Frequency Responses: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2 20Hz to 22kHz, +/-0.1 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.2 20Hz to 42kHz, +/-0.2 20Hz to 44kHz, +/-0.2 20Hz to 52kHz, +/-0.1 20Hz to 44kHz, +/-0.2 20Hz to 52kHz, +/-0.2 20Hz to 52kHz, +/-0.3 20H	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 96kHz sample rate	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2c ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: Pulsar II Type: Diaphragm: Pattern: Backplate: Electronics: Frequency Response: Frequency Response: Frequency Response: Sensitivity: 13.8 mV/Pa (-37dB @ Max SPL: 134dB @ 0.5% THD,	@ -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbalai -104dB, A-weig 1/-0.75dB, 20l 1000 ohms +0.3dBV into 3 32 ohms 32 to 600 ohm +12.9dBV (4.4 -104dB, A-weig 0-1dBFS, 1kHz +/-0.3dB, 20H: 150 ohms, unb +2.4dBV into 3 75 ohms 32 to 600 ohm small-diaphragr 3/4" diameter, to ardioid solid brass class A FET (tra 20Hz to 20kHz 1kHz, 0dB=1V/I 14ddB with -10cl
0.00201% (94.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 44kHz, +/-0.15 Max Output Impedance: 300 ohms, balanced; 1 Max Output Level: +10dBu, balanced; +1 Channel+C-Channel Crosstalk: Signal+to-Noise Ratio: Dynamic Range: THD+N: 0.00205% (-93.8dB) @ Frequency Response: 20Hz to 22kHz, +/-0.1 20Hz to 24kHz, +/-0.2 Headphone Output Max Output: -2.9dBV @ THD+N Working Range: ProKeys Sono 61 ine Inputs Max Input: Dynamic Range: THD+N: Frequency Response: Impedance: Wic Input Max Input Level, Min Gain: Signal-to-Noise Ratio: Dynamic Range: THD+N: Frequency Response: Impedance: Frequency Response: Prengenge Response: Prequency Resp	dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 48kHz sample rate dB @ 110dB, A-weighted 110dB, A-weighted @ -1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample rate dB @ 1dBFS, 1kHz dB @ 48kHz sample rate dB @ 96kHz sample	THD+N: 0.00453% (-87.1dB) @ Frequency Response: 20Hz to 22kHz, +/-0.2t ProKeys 88 Line Outputs Max Output: Signal-to-Noise Ratio: Dynamic Range: Frequency Response: Impedance: Headphone Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Output Impedance: Working Headphone Impedance: ProKeys 88sx Line Outputs Max Output: Output Range: THD+N: 0.00200% (-94.3dB) @ Frequency Response: Impedance: Working Headphone Impedance: Working Headphone Impedance: Working Headphone Impedance: Pulsar II Type: Diaphragm: Pattern: Backplate: Electronics: Frequency Response: Frequency Response: Frequency Response: Sensitivity: 13.8 mV/Pa (-37dB @ Max SPL:	9 -1dBFS, 1kHz dB @ 48kHz sar +7dBV, unbala -104dB, A-weig 104dB, A-weig 104dB, A-weig 1-0.75dB, 20l 1000 ohms +0.3dBV into 3 32 ohms 32 to 600 ohm +12.9dBV (4.4 -104dB, A-weig 104dB, A-weig 104dB, A-weig 2 -1dBFS, 1kHz -1/0.3dB, 20H 150 ohms, unb +2.4dBV into 3 75 ohms 32 to 600 ohm small-diaphragr 3/4" diameter, to ardioid solid brass class A FET (tra 20Hz to 20kHz 1kHz, 0dB=1V/l 144dB with -10c 16dBA

put Level 32 Ohms:	+6.8dBV, typical	Line Outputs Max Output:	+2.1dBV (1.27 Vrms), unbalanced
to Ohms:	150mW into 32 ohms	Signal-to-Noise Ratio:	-95dB, A-weighted
mpedance: pedance:	75 ohms 24 to 600 ohms	Dynamic Range: THD+N:	93dB, A-weighted 0.0045% (-87dB) @ -1dBFS, 1kHz
Lightbridge		Frequency Response:	+/-0.2dB, 20Hz to 20kHz
outs		Impedance:	240 ohms, unbalanced
	50 ohms, unbalanced	Headphone Outputs Max Output:	
put Level:	+10.2dBu (2.5 Vrms), balanced	-2.2dBV @ THD+N <	0.055% into 32 ohms
to-Channel	+2.0dBV (1.26 Vrms), unbalanced	Working Headphone Impedance:	32 to 600 ohms
stalk:	<-110dB @ 1kHz	ProKeys Sono 88	
-Noise Ratio: Range:	-109dB, A-weighted 109dB, A-weighted	Line Inputs Max Input:	+2.1dBV (1.3 Vrms), unbalanced
_	0.00127 % (-98dB) @ -1dBFS,	Signal-to-Noise Ratio:	-94dB, A-weighted
z cy Response:		Dynamic Range:	94dB, A-weighted
z to 20kHz, +/-0.1	dB @ 48kHz sample rate	THD+N: Frequency Response:	0.005% (-86dB) @ -1dBFS, 1kHz +/-0.2dB, 20Hz to 20kHz
z to 40kHz, +/-0.2d ne Output	dB @ 96kHz sample rate	Impedance:	10k ohms, unbalanced
put:		Mic Input	0.0 15 (0.0)()
dBV (1.71 Vrms) in Range:	to 32 ohms @ <0.02% THD+N, 1kHz 32 to 600 ohms	Max Input Level, Min Gain: Signal-to-Noise Ratio:	-2.2dBu (0.6 Vrms), balanced -94dB, A-weighted
ce:	75 ohms	Dynamic Range:	94dB, A-weighted
k:	<-85dB	THD+N:	0.005% (-86dB) @ -1dBFS, 1kHz
-Noise Ratio: Range:	-108dB, A-weighted 108dB, A-weighted	Frequency Response: Preamp Gain:	+/-0.2dB, 20Hz to 20kHz >38dB adjustable range
cy Response:	-	Impedance:	5.4k ohms, balanced
	dB @ 48kHz sample rate dB @ 96kHz sample rate	Instrument Input	110 0 d D V (0 0 V)
t Mix I/O	db @ 90kHz sample fate	Max Input Level, Min Gain: Signal-to-Noise Ratio:	+10.0dBV (3.2 Vrms), unbalanced -94dB, A-weighted
s		Dynamic Range:	94dB, A-weighted
pedance:	3.4k ohms	THD+N: Frequency Response:	0.009% (-81dB) @ -1dBFS, 1kHz +/-0.2dB, 20Hz to 20kHz
t Level, Min Gain: to-Channel	-3dBu, balanced	Preamp Gain:	>34dB adjustable range
stalk:	<-110dB @ 1kHz	Impedance:	1M ohm, unbalanced
-Noise Ratio: Range:	-104dB, A-weighted 104dB, A-weighted	Line Outputs Max Output:	+2.1dBV (1.27 Vrms), unbalanced
		Signal-to-Noise Ratio:	-95dB, A-weighted
188% (-94.6dB) (cy Response:	20 -1 dBFS, 1kHz 20Hz to 20kHz, +/-0.1dB	Dynamic Range:	93dB, A-weighted
Gain:	55dB	THD+N: Frequency Response:	0.0045% (-87dB) @ -1dBFS, 1kHz +/-0.2dB, 20Hz to 20kHz
Power:	48V DC @ 16mA	Impedance:	240 ohms, unbalanced
nt Input bedance:		Headphone Outputs	
k ohms, balanced;	280k ohms, unbalanced	Max Output: -2.2dBV @ THD+N <	0.055% into 32 ohms
it Level, Min Gain: dBu. balanced: +1	1.8dBV, unbalanced	Working Headphone	20.4- 600
-Noise Ratio:	-100dB, A-weighted	Impedance: Transit	32 to 600 ohms
Range:	100dB, A-weighted	Line/Mic Input (A/D)	
243% (-92.3dB) @	9 -1dBFS, 1kHz	Max Input Level:	-1.9dBV (0.808 Vrms)
cy Response:	dB @ 48kHz sample rate	Channel-to-Channel Crosstalk:	<-100dB @ 1kHz
ts	GD @ 40Ki iz sample rate	Signal-to-Noise Ratio:	-99dB, A-weighted
pedance:	Ok ahma unhalanaad	Dynamic Range: THD+N:	99dB, A-weighted 0.00504% (-86dB) @ -1dBFS, 1kF
it Level @ Min Gair	0k ohms, unbalanced n:	Frequency Response:	·
dBu balanced/+7.8 to-Channel	3dBV unbalanced		dB @ 48kHz sample rate
stalk:	<-110dB @ 1kHz	Line/Headphone Output (Max Output Level:	-0.4dBV (0.953 Vrms)
-Noise Ratio:	-104dB, A-weighted	Channel-to-Channel	< 0640
Range:	104dB, A-weighted	Crosstalk: Signal-to-Noise Ratio:	<-96dB -101dB, A-weighted
201% (-94.1dB) (9 -1dBFS, 1kHz	Dynamic Range:	101dB, A-weighted
cy Response: z to 22kHz, +/-0.1	dB @ 48kHz sample rate	THD+N: 0.00453% (-87.1dB) @	ି -1dBFS. 1kHz
z to 44kHz, +/-0.1	dB @ 96kHz sample rate	Frequency Response:	
outs mpedance:			dB @ 48kHz sample rate
	50 ohms, unbalanced	ProKeys 88 Line Outputs	
put Level: dBu. balanced: +1	.8dBV, unbalanced	Max Output:	+7dBV, unbalanced
to-Channel	<u> </u>	Signal-to-Noise Ratio: Dynamic Range:	-104dB, A-weighted 104dB, A-weighted
stalk: -Noise Ratio:	<-109dB -110dB, A-weighted	Frequency Response:	+/-0.75dB, 20Hz to 20kHz
Range:	110dB, A-weighted	Impedance:	1000 ohms
205% (-93.8dB) @	2 1dBEC 1kH-	Headphone Outputs	+0.3dBV into 32 ohms
cy Response:	y TUBES, TREE	Max Output: Output Impedance:	32 ohms
	dB @ 48kHz sample rate	Working Headphone	32 to 600 ohms
ne Output	dB @ 96kHz sample rate	Impedance: ProKeys 88sx	32 to 600 orins
put:		Line Outputs	
dBV @ THD+N <0 Range:	0.03% into 32 ohms 24 to 600 ohms	Max Output:	+12.9dBV (4.4 Vrms), unbalanced
range. /s Sono 61	24 to 600 onns	Signal-to-Noise Ratio:	-104dB, A-weighted
ts		Dynamic Range: THD+N:	104dB, A-weighted
rt:	+2.1dBV (1.3 Vrms), unbalanced	0.00200% (-94.3dB)	
-Noise Ratio: Range:	-94dB, A-weighted 94dB, A-weighted	Frequency Response: Impedance:	+/-0.3dB, 20Hz to 20kHz 150 ohms, unbalanced
Tango.	0.005% (-86dB) @ -1dBFS, 1kHz	Headphone Outputs	
cy Response:	+/-0.2dB, 20Hz to 20kHz	Max Output:	+2.4dBV into 32 ohms 75 ohms
ce:	10k ohms, unbalanced	Output Impedance: Working Headphone	75 Onms
t Level, Min Gain:	-2.2dBu (0.6 Vrms), balanced	Impedance:	32 to 600 ohms
-Noise Ratio:	-94dB, A-weighted	Pulsar II	emall-diaphraem cond
Range:	94dB, A-weighted 0.005% (-86dB) @ -1dBFS, 1kHz	Type: Diaphragm:	small-diaphragm condenser 3/4" diameter, 6-micron Mylar
cy Response:	+/-0.2dB, 20Hz to 20kHz	Pattern:	cardioid
Gain:	>38dB adjustable range	Backplate:	solid brass class A FET (transformerless)
ce: nt Input	5.4k ohms, balanced	Electronics: Frequency Response:	20Hz to 20kHz
t Level, Min Gain:	+10.0dBV (3.2 Vrms), unbalanced	Sensitivity:	
-Noise Ratio:	-94dB, A-weighted	13.8 mV/Pa (-37dB @ Max SPL:	IKHZ, UDD=IV/Pa)
Range:	94dB, A-weighted 0.009% (-81dB) @ -1dBFS, 1kHz	134dB @ 0.5% THD,	144dB with -10dB pad
		Equivalent Noise:	16dBA
cy Response:	+/-0.2dB, 20Hz to 20kHz	B. 1	
cy Response: Gain: ce:	+/-0.2dB, 20Hz to 20kHz >34dB adjustable range 1M ohm, unbalanced	Pulsar II Matched I Type:	Pair small-diaphragm condenser

alanced	Diaphragm: Pattern:	3/4" diameter, 6-micron Mylar cardioid
	Backplate:	solid brass
FS, 1kHz	Electronics:	class-A FET (transformerless)
FS, IKHZ	Frequency Response: 20Hz to 20kHz (bo	oth mics matched within 1dB of each other
	Sensitivity:	
	13.8 mV/Pa (-37dt Max SPL:	B @ 1kHz, 0dB=1V/Pa)
		HD, 144dB with -10dB pad
	Equivalent Noise:	16dBA
	RX	
	Windows Windows XP, XP Profe	assignal v64 or Vista
	Mac	SSIONAL XO4 OF VISTA
anced	Mac OS X 10.3.9 or la	ter (Universal Binary)
	Compatibility	•
S, 1kHz	RTAS, VST, AU, MAS, I	DirectX
	Standalone	
	Solaris	large diaphroam condensor
ed	Type: Capsule:	large-diaphragm condenser double-sided 1.1" evaporated-gold
	Pattern:	cardioid, omni, figure-8
	Frequency Response:	20Hz to 20kHz
, 1kHz	Preamp Topology:	class A FET, with output transformer
	Sensitivity: Max SPL:	16mV/Pa (-36dBV) 130dB for 0.5% THD
	Pad:	10dB switchable
	LF Rolloff Filter:	6dB/octave @ 125Hz, switchable
nced	Equivalent Noise Level:	, ,
	Output Impedance:	200 ohms
1kHz	Recommended Load Impedance:	>1,000 ohms
I KITIZ	Connector:	3-pin male XLR
	Power:	>48V phantom power
	Dimensions: 8 25" (209 55 mm) x 3" (76.2 mm) x 2" (50.8 mm)
	Weight: 8.25" (209.55 mm	1.4 lbs (.64 kg)
nced	Spectron	. 3,
	Windows	
S, 1kHz	Windows XP, x64, Vista	a
,	Mac	
	OS X 10.4 or later	
	Compatibility	DNA Dimesty MARC VICT ALL
		DM, DirectX, MAS, VST, AU
	Sputnik Type:	large-diaphragm condenser
	Capsule:	large-diaprilagin condenser
	3-micron thick Myla	ar with evaporated gold, double sided
	Patterns:	cardioid, omni, figure-8
	Transconductance Am vacuum tube, wired as	plifier: 6205M pentode a triode, military grade selected
	Frequency Response:	20Hz to 20kHz ±1.5dB
	Sensitivity:	30mV/Pa (-30.5dBV)
	Max SPL for 0.5% THE	
S, 1kHz	Equivalent Noise Level: Output Impedance:	: 18dB A-weighted 200 ohms, transformer isolated
	Recommended Load	200 Oning, narisionner isolated
	Impedance:	>1k ohm
	Connectors:	7-pin male XLR for mic output to ale XLR for power supply output
	Attenuation and Rolloff	i: switchable 10dB pad; switchable 3/octave) rolloff
		3/octave) rolloff
	Dimensions: 8.25" (209.55 mm) x 3" (76.2 mm) x 2" (50.8 mm)
	Weight:	1.6 lbs (.72 kg)
	Strike	-
	Digidesign-gualified W	findows XP- or Mac-based Pro Tools
		ols 7.0 or higher software or better recommended)
	DVD-ROM drive for ins	
	iLok USB Smart Key (s	sold separately), Internet access and a free
	iLok.com account (for r	retrieving and managing iLok licenses)
	Structure	
	Digidesign-qualified W system running Pro Too	findows XP- or Mac-based Pro Tools ols 7.0 or higher software (Pro Tools 7.3 o
	greater recommended	for optimal performance)
	1GB RAM (2GB RAM	
	DVD-ROM drive for ins	stallation sold separately), Internet access and a fre
	iLok.com account (for r	retrieving and managing iLok licenses)
	Studiophile AV 4	10
	Type:	two-way desktop reference monitor
nced	LF Driver:	4" diameter, magnetically shielded th high-temperature voice coil
	HF Driver:	gir temperature voice coll
	3/4" magnetically s	shielded silk dome tweeter
	Frequency Response:	85Hz to 20kHz
	Crossover Frequency:	2.7kHz
	RMS SPL: Signal-to-Noise Ratio:	101.5dB @ 1 meter >90dB (typical, A-weighted)
_	Input Connectors:	left and right RCA line input, left and
	right 1/4" TRS input ar	nd 1/8" aux input
	Polarity:	positive signal @ + input produces
	outward LF cone d Dynamic Power:	пориссентен
		us, per channel into 4 ohms
	Input Impedance:	and 20k ahma balaasa
r	10k ohms unbaland Input Sensitivity:	ced, 20k ohms balanced
		input produces 90dBA output SPL @ 1
	meter with volume cont	
	Protection:	tout ourront limiting and a
	DL:	
	RF interference, ou	
	RF interference, ou on/off transient, subsor Indicator:	
	on/off transient, subsor Indicator: blue power LED rir	
	on/off transient, subsor Indicator: blue power LED rin Power:	nic filter ng around volume knob on front panel
	on/off transient, subsor Indicator: blue power LED rin Power:	nic filter ng around volume knob on front panel OHz, 220-240V/~50/60Hz; powered via

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83

vinyl-laminated MDF

	14 (6 04)
Weight: Studiophile BX5a	14 lbs (6.34 kg)
Frequency Response:	56Hz to 22kHz
Crossover Frequency:	3kHz
LF Amplifier Power: HF Amplifier Power:	40 watts 30 watts
Signal-to-Noise Ratio:	>100dB (typical A-weighted
Polarity: positive signal @ + in	put produces outward LF cone
displacement Input Impedance:	•
20k ohms, balanced;	10k ohms, unbalanced
Input Sensitivity: 90dBA output SPL @	85mV pink noise input produces one meter with volume control @ max
Power:	ither 115V ~50/60Hz, 230V~50/60Hz
Protection:	RF interference, output current
external mains fuse	, turn-on/off transient, subsonic filter,
Cabinet: Dimensions:	vinyl-laminated MDF
9.84" (250 mm) x 6.9	93" (176 mm) x 7.87" (200 mm)
Weight: Studiophile BX8a	11.0 lbs (5.0 kg)
Frequency Response:	30Hz to 24kHz
Crossover Frequency:	2.2kHz
LF Amplifier Power: HF Amplifier Power:	70 watts 60 watts
Signal-to-Noise Ratio:	>100dB (typical A-weighted)
Polarity: outward LF cone disp	positive signal @ + input produce placement
Input Impedance:	
Input Sensitivity:	10k ohms, unbalanced 85mV pink noise input produces
Power:	e meter with volume control @ max factory programmed for either 115V
~50/60Hz or 230V ~50/ Protection:	RF interference, output current
limiting, over temperature	, turn-on/off transient, subsonic filter,
Cabinet:	vinyl-laminated MDF
Dimensions:	" (254 mm) x 15" (381 mm)
Weight:	26.4 lbs (11.97 kg)
Studiophile BX10s	5
Type: bass-reflex active pov	vered subwoofer
Woofer Driver:	10-inch composite (treated paper/ n-temperature voice coil and damped
rubber surround	
Frequency Response: Crossover:	20Hz to 200Hz (-3dB points) HP/LP linked filter sweepable from
50 to 200Hz; 24dB/octa	ve 4th-order alignment
Input Sensitivity: 100dBA output SPL @ c	85mV pink noise input produces ne meter with volume control @ max
Amplifier: ohms, discrete transistor	240 watts average power into 8 design
Gain:	36dB
Signal-to-Noise Ratio: THD @ Rated Power, Mo	86dB A-weighted no Mode: 0.019% (175 watts)
Inputs: balanced/unbalanced inp	two XLR balanced inputs; two TRS
Outputs:	two XLR balanced outputs for
satellites; two TRS balance Controls:	volume/gain control, 10dB bass
boost switch, sleep mode	switch, phase inverse switch, variable ofer bypass footswitch jack
Input Impedance:	oler Bypaco rectornici yacı
20k ohms balanced, Max Input Level:	10k ohms unbalanced +20dBU
Auto Turn Off Time:	15 minutes
Auto Turn On Input Sensitivity:	7.3mV
Protection:	over temperature, turn-on/off external mains fuse, DC protection, over
current protection	Accinal mains ruse, DO protection, over
Indicator: power on (blue) or sta	andby/bypass (red)
Power: switch) for either 100-120	dual-voltage (selectable by rear-pane 0V/~50/60Hz or 220-240V/~50/60Hz
powered via detachable	5., 56,661 ½ 01 220-240V/~30/60H2
grounded IEC cable Cabinet:	vinyl-laminated MDF
15" (381 mm) x 15" (Weight:	(381 mm) x 15" (381 mm) 54.6 lbs (24.75 kg)
Studiophile DSM1	- 1.0 1.00 (2 1.7 0 hg)
General	
Frequency Response: Peak SPL @ 1 Meter:	49Hz to 27kHz 110dB
LF Driver:	
6.5" (165.1 mm) dom HF Driver:	neless anodized aluminum cone 1" (25.4 mm) soft Teteron dome with
Neodymium magnet a	and ferrofluid cooling
Crossover Frequency: HF shelf:	2.7kHz @ 24dB/octave +1.5dB, -1.5dB, -3.0dB
Mid EQ:	+1.5dB, -1.5dB, -3.0dB
High-Pass Filter: 40Hz, 60Hz, 80Hz, 1	00Hz @ 12dB/octave
Acoustic Space	
Control Filter: Desk EQ Filter 1:	-1.5dB, -3.0dB, -4.5dB -1.0dB, -2.0dB, -3.0dB @ 220Hz
Desk EQ Filter 2:	-1.0dB, -2.0dB, -3.0dB @ 175Hz
Desk EQ Filter 3:	-1.0dB, -2.0dB, -3.0dB @ 200Hz XLR balanced and 1/4" balanced
Analog Inputs: Digital Inputs:	S/PDIF and AES/EBU
	20k ohms
Input Impedance:	
Input Sensitivity:	@ 1 meter (trim level set to 0)
Input Sensitivity:	

Analog-to-Digital Specifica Max Input Level:	t ions 18dBu
Signal-to-Noise Ratio:	102dB A-weighted
THD+N:	92dB
Conversion Rate:	96kHz
Digital Power Amplifier Output Power (1% THD):	LF: 100 watts, HF: 80 watts
THD+N (10 watts, 1kHz):	<0.05%
Signal-to-Noise Ratio:	100dB A-weighted
Frequency Response	001 - +- 4011 -
(+0, -3dB): Efficiency:	20Hz to 40kHz >90%
Studiophile DSM2	20070
General	
Frequency Response:	42Hz to 27kHz
Peak SPL @ 1 Meter:	111dB
LF Driver: 8" (203.2 mm) domele	ss anodized aluminum cone
HF Driver:	1" (25.4 mm) soft Teteron dome wit
Neodymium magnet and fe	2.7kHz @ 24dB/octave
Crossover Frequency: HF Shelf:	+1.5dB, -1.5dB, -3.0dB
Mid EQ:	+1.5dB, -1.5dB, -3.0dB
High-Pass Filter:	
40Hz, 60Hz, 80Hz, 10	UHZ @ 12dB/octave
Acoustic Space Control Filter:	-1.5dB, -3.0dB, -4.5dB
Desk EQ Filter 1:	-1.0dB, -2.0dB, -3.0dB @ 220Hz
Desk EQ Filter 2:	-1.0dB, -2.0dB, -3.0dB @ 175Hz
Desk EQ Filter 3: Analog Inputs:	-1.0dB, -2.0dB, -3.0dB @ 200Hz XLR balanced and 1/4" balanced
Digital Inputs:	S/PDIF and AES/EBU
Input Impedance:	20k ohms
Input Sensitivity:	1 mater (trim level+ +- 0)
Gain Trim:	1 meter (trim level set to 0) volume trim from -22dB to +10dB
Dimensions:	Volume transform 22dB to 110dB
	" (270.3 mm) x 10.1" (257.5 mm)
Weight:	20 lbs (9.1 kg)
Analog-to-Digital Specifica Max Input Level:	18dBu
Signal-to-Noise Ratio:	102dB A-weighted
THD+N:	92dB
Conversion Rate:	96kHz
Digital Power Amplifier	LE: 100 watto HE: 80 watto
Output Power (1% THD): THD+N (10 watts, 1kHz):	LF: 100 watts, HF: 80 watts <0.05%
Signal-to-Noise Ratio:	100dB A-weighted
Frequency Response	001 - 4- 4011 -
(+0, -3dB): Efficiency:	20Hz to 40kHz >90%
Studiophile Q40	20070
Frequency Response:	10Hz to 20kHz
Sensitivity:	116dB SPL (1mW IEC318)
Rated Impedance:	64 ohms
Weight Without Cable: Magnet Material:	8.8 oz (.25 kg)
Driver:	neodymium 40mm Mylar
TimewARP 2600	· · · · · · · · · · · · · · · · · · ·
Windows	
Windows XP	
1.5GHz	
256MB RAM, 1024 x 768	screen resolution
Mac OS X 10.3 or greater	
1GHz	
Compatibility	
RTAS, VST, AU	
RTAS, VST, AU Torq	
RTAS, VST, AU Torq Minimum System Requirem	nents*
RTAS, VST, AU Torq Minimum System Requirem Windows XP	nents*
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI	B RAM
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fc	
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fo hardware device)	B RAM
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fc	B RAM
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fo hardware device) Windows Vista 32-Bit Windows Vista 32-Bit Pentium IV 1.8GHz, 1GB F	B RAM r use by approved M-Audio
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fo hardware device) Windows Vista 32-Bit Windows Vista 32-Bit Pentium IV 1.8GHz, 1GB F One available USB port (fo	B RAM r use by approved M-Audio
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fo hardware device) Windows Vista 32-Bit Windows Vista 32-Bit Vindows Vista 32-Bit Pentium IV 1.8GHz, 1GB F One available USB port (fo hardware device)	B RAM r use by approved M-Audio RAM r use by approved M-Audio
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fo hardware device) Windows Vista 32-Bit Pentium IV 1.8GHz, 1GB F One available USB port (fo hardware device) Windows Experience Index	B RAM r use by approved M-Audio RAM r use by approved M-Audio
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fo hardware device) Windows Vista 32-Bit Windows Vista 32-Bit Vindows Vista 32-Bit Pentium IV 1.8GHz, 1GB F One available USB port (fo hardware device)	B RAM r use by approved M-Audio RAM r use by approved M-Audio
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fc hardware device) Windows Vista 32-Bit Windows Vista 32-Bit Windows Vista 32-Bit One available USB port (fc hardware device) Windows Experience Index Windows Vista 64-Bit Windows Vista 64-Bit Pentium IV 1.8GHz, 1GB F	B RAM Ir use by approved M-Audio RAM Ir use by approved M-Audio of 3.0 or better
RTAS, VST, AU Torq Minimum System Requirem Windows XP Windows XP (SP3) Pentium IV 1.4GHz, 512MI One available USB port (fc hardware device) Windows Vista 32-Bit Windows Vista 32-Bit Windows Vista 32-Bit One available USB port (fc hardware device) Windows Experience Index Windows Vista 64-Bit Windows Vista 64-Bit Pentium IV 1.8GHz, 1GB F	B RAM Ir use by approved M-Audio RAM Ir use by approved M-Audio of 3.0 or better

Mac G4 users are advised to not open the Timecode Preferences Dialog while streaming more than one timecode source into Torq, as this may degrade G4 performance. The user is advised to calibrate before performing, one timecode source at a time.

340k ohms +4.1dBu (1.243 Vrms), min gain

-98dB, A-weighted

98dB, A-weighted

*Minimum System Requirements reflect the ability to run Torq using a basic feature set. Some advanced or intensive features will require increased system specifications.

THD+N: 0.00686% (-83.3dB) @ -1dBFS, 1kHz

Torq Conectiv

Mic Inputs (A/D)

Input Impedance

Dynamic Range:

Max Input Level Signal-to-Noise Ratio:

Desk EQ Filter 3:	-1.0dB, -2.0dB, -3.0dB @ 200Hz	Headphone Output	
Analog Inputs:	XLR balanced and 1/4" balanced	Max Output:	\ aa .
Digital Inputs:	S/PDIF and AES/EBU		op) into 30 ohms per channe
Input Impedance:	20k ohms	Working Range:	16 to 80 ohms
Input Sensitivity:		Torq Xponent	
+4dBu yields 90dB @	1 meter (trim level set to 0)	Line Outputs (D/A)	
Gain Trim:	volume trim from -22dB to +10dB	Output Impedance:	100 ohms min, unbalanc
Dimensions:	" (OFO O) 40 4" (OFFF)	Max Output Level:	+10dBV, unbalanced
	" (270.3 mm) x 10.1" (257.5 mm)	Channel-to-Channel	< 400 ID @ 4111
Weight:	20 lbs (9.1 kg)	Crosstalk: Signal-to-Noise Ratio:	<-100dB @ 1kHz
Analog-to-Digital Specifica			-100dB, A-weighted
Max Input Level:	18dBu	Dynamic Range: THD+N:	94dB, A-weighted
Signal-to-Noise Ratio:	102dB A-weighted	Frequency Response:	<0.005% (-86dB) @ -1c
THD+N:	92dB		.3dB @ 48kHz sample rate
Conversion Rate:	96kHz	Headphone Output	.002 @ 1010 iz 0011pio 1010
Digital Power Amplifier		Max Output:	
Output Power (1% THD):	LF: 100 watts, HF: 80 watts	+8dBV (2.6 Vrms) int	to 30 ohms per channel
THD+N (10 watts, 1kHz):	<0.05%	Output Impedance:	75 ohms max
Signal-to-Noise Ratio:	100dB A-weighted	Working Range:	32 to 600 ohms
Frequency Response	001 - +- 401-1 -	Channel-to-Channel	
(+0, -3dB):	20Hz to 40kHz	Crosstalk:	<-80dB @ 1kHz
Efficiency:	>90%	Signal-to-Noise Ratio:	-99dB, A-weighted
Studiophile Q40		Dynamic Range:	93dB, A-weighted
Frequency Response:	10Hz to 20kHz	THD+N:	0.02% (-74dB) @ -1dBF
Sensitivity:	116dB SPL (1mW IEC318)	Frequency Response:	
Rated Impedance:	64 ohms	20Hz to 20kHz, +/-0	.4dB @ 48kHz sample rate
Weight Without Cable:	8.8 oz (.25 kg)	Transfuser	
Magnet Material:	neodymium	Digidesign-qualified Wind	dows XP- or Mac-based Pro 7.0 or higher software
Driver:	40mm Mylar		
TimewARP 2600		iLok USB Smart Key (sol	d separately), Internet acces
Windows		iLok.com account (for ret	rieving and managing iLok lic
Windows XP		Trash	
1.5GHz		Windows	
256MB RAM, 1024 x 768	screen resolution	Windows XP, x64, Vista	
Mac		Mac	
Mac OS X 10.3 or greater		OS X 10.4 or later	
1GHz		Compatibility	
Compatibility		RTAS, AudioSuite, HTDM	1. DirectX. MAS. VST. AU
RTAS, VST, AU		Velvet	,,
Torq			dows XP- or Mac-based Pro
Minimum System Requirem	nente*	system running Pro Tools	dows XP- or Mac-based Pro 7.0 or higher software
Windows XP	ichts		llation (download-only also a
Windows XP (SP3)		iLok USB Smart Key (sol	d separately), Internet acces
Pentium IV 1.4GHz, 512M	B RAM	iLok.com account (for ret	rieving and managing iLok lic
	or use by approved M-Audio	Virtual String Mac	hine
hardware device)		Windows	
Windows Vista 32-Bit		Intel Pentium IV 1GHz wi	th 512MB RAM
Windows Vista 32-Bit		5GB available disk space	
Pentium IV 1.8GHz, 1GB F	RAM	Windows XP with SP2	
One available USB port (fo	or use by approved M-Audio		ard for standalone operation
hardware device)		Macintosh	a s. standard operation
Windows Experience Index	of 3.0 or better	1.25GHz PowerMac G4	with 510MB of DAM
Windows Vista 64-Bit		5GB available disk space	
Windows Vista 64-Bit		Mac OS X 10.4.x	-
Pentium IV 1.8GHz, 1GB F	MAS		
One available USB port (fo	or use by approved M-Audio	Compatibility	
hardware device)		RTAS, VST, AU	
Windows Experience Index	of 3.0 or better	CURRORTER CO.	DATING CYCTERS
Mac			ERATING SYSTEMS
Mac OS X 10.4.11		(HARDWARE)^	
G4 1.67GHz, 512MB RAM), Vista (32-Bit SP1, 64-Bit S
	or use by approved M-Audio	Macintosh OS X 10.4.9,	10.5.4
hardware device)	iond to pro applyzo all audio filo	Con an audio com (lisiana I amasana mandina .
to use in Torq.	ised to pre-analyze all audio files prior	information about suppor	litional system requirements
	to not open the Timecode Preferences		that you also check the minin
Di-l	to not opon the fillicoode i leichtelees	ivi-Audio reconninends i	inac you also driedk life illillill

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specifications, system requirements and availability are subject to change without notice. All prices are USMSRP for U.S. only and are subject to change without notice. Avid, M-Audio and

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property of their respective owners.

Frequency Response: 20Hz to 20kHz, +/-0.20	dB @ 48kHz sample rate
Preamp Variable Gain:	>25dB
Phono Inputs (A/D) Input Impedance:	48k ohms
EQ:	RIAA curve
Gain:	43dB
Line Inputs (A/D) Input Impedance:	48k ohms
Max Input Level, Min Gain:	+10dBV (3.162 Vrms), unbalanced
Channel-to-Channel Crosstalk:	<-100dB @ 1kHz
Signal-to-Noise Ratio:	-97dB, A-weighted
Dynamic Range:	97dB, A-weighted
THD+N: 0.00557% (-85.1dB) @	-1dBFS, 1kHz
Frequency Response: 20Hz to 20kHz, +/-0.20	dB @ 48kHz sample rate
Line Outputs (D/A)	35 @ Tota E campio rato
Output Impedance:	100 ohms, unbalanced
Max Output Level: Channel-to-Channel	+11.4dBV (3.711 Vrms), unbalanced
Crosstalk:	<-109dB @ 1kHz
Signal-to-Noise Ratio: Dynamic Range:	-100dB, A-weighted
THD+N:	95dB, A-weighted
0.00390% (-88.2dB) @	9 -1dBFS, 1kHz
Frequency Response: 20Hz to 20kHz, +/-0.10 Headphone Output	dB @ 48kHz sample rate
Max Output:	\ int= 00 -b
OdBV (1 Vrms, 2.8 Vpp Working Range:) into 30 ohms per channel 16 to 80 ohms
Torq Xponent	
Line Outputs (D/A)	
Output Impedance:	100 ohms min, unbalanced
Max Output Level: Channel-to-Channel	+10dBV, unbalanced
Crosstalk:	<-100dB @ 1kHz
Signal-to-Noise Ratio:	-100dB, A-weighted
Dynamic Range: THD+N:	94dB, A-weighted <0.005% (-86dB) @ -1dBFS, 1kHz
Frequency Response:	
Headphone Output	dB @ 48kHz sample rate
Max Output:	
+8dBV (2.6 Vrms) into Output Impedance:	30 ohms per channel 75 ohms max
Working Range:	32 to 600 ohms
Channel-to-Channel	
Crosstalk: Signal-to-Noise Ratio:	<-80dB @ 1kHz -99dB, A-weighted
Dynamic Range:	93dB, A-weighted
THD+N:	0.02% (-74dB) @ -1dBFS, 1kHz
Frequency Response: 20Hz to 20kHz, +/-0.4d	dB @ 48kHz sample rate
Transfuser	
Digidesign-qualified Windowsystem running Pro Tools 7.	ws XP- or Mac-based Pro Tools 0 or higher software
iLok USB Smart Key (sold :	separately), Internet access and a free
	ving and managing iLok licenses)
Trash Windows	
Windows XP, x64, Vista	
Mac	
OS X 10.4 or later	
Compatibility RTAS, AudioSuite, HTDM, I	DirectX, MAS, VST, AU
Velvet	
Digidesign-qualified Windowsystem running Pro Tools 7	ws XP- or Mac-based Pro Tools 0 or higher software
	tion (download-only also available)
iLok USB Smart Key (sold s	separately), Internet access and a free ving and managing iLok licenses)
Virtual String Mach Windows	ine
Intel Pentium IV 1GHz with	512MB RAM
5GB available disk space	
Windows XP with SP2 ASIO-compatible soundcar	d for standalans aparation
Macintosh	d for standatone operation
1.25GHz PowerMac G4 w	ith 512MB of RAM
5GB available disk space	
Mac OS X 10.4.x Compatibility	
RTAS, VST, AU	
	NATINIC CVCTCAC
SUPPORTED OPER (HARDWARE)^	MING STSTEMS
	/ista (32-Bit SP1, 64-Bit SP1)
Macintosh OS X 10.4.9, 10	
	onal system requirements and
information about supported	d operating systems.
requirements for your softw	t you also check the minimum system are, as they may be higher than
the above.	· • •

