

## **DirectOut RAV2 Module Audio Network Module User Manual**

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**DirectOut RAV2 Module Audio Network Module** 



#### **RAV2 Module**

### **Specifications:**

- Software Manual Version: 2.8
- Audio network module for RAVENNA / AES67
- Browser-based interface (HTML5 / JavaScript)
- · Resizable window and zoom level
- · Organized in tabs, pulldown menus, and hyperlinks
- Supports input fields for parameter values (e.g., IP address)
- Two independent network interfaces (NICs)
- Port 1 is fixed assigned to NIC 1

### **Product Usage Instructions**

### **Connecting Audio Network:**

Before connecting the audio network, ensure that NIC 1 and NIC 2 are configured to different subnets. Follow the steps below:

- 1. Access the "Network Settings" on page 7 of the user manual
- 2. Configure NIC 1 and NIC 2 with different subnets

#### Status - Overview:

The "STATUS" tab provides an overview of various sections:

- Monitoring sync state, clock selection, links to I/O settings
- Display network info, link to network settings
- Monitoring device info, link to device settings, phones level control
- · Links to input stream settings and output stream settings

Hyperlinks open a popup window for adjusting related settings. Most settings are updated immediately without further notification.

To exit a popup window, click the button in the top right corner.

Mouse overs display additional information, such as the connection speed of the network link.

#### Status - Sync:

The "Sync" section on the "STATUS" tab displays the following information:

- · Clock source and state for the main frame
- Pulldown menu to select clock source of the main frame (PTP, extern)
- Pulldown menu to adjust sample rate of the main frame (44.1 / 48 / 88.2 / 96 / 176.4 / 192 kHz)
- State of PTP (Master / Slave)
- PTP-clock jitter per second
- Offset relative to PTP-clock master
- Status of packet processing (OK, Error\*)
- State of module's audio engine receiving (ON / blinking)
- State of module's audio engine sending (ON / blinking)

\*Error: packet time stamps are out of bounds. Possible reasons: stream offset may be too small or transmitter or receiver are not synced properly to the Grandmaster.

#### **PTP Settings:**

The "PTP Settings" section allows you to configure PTP input:

- NIC selection for PTP clock input. "NIC 1 & 2" means input redundancy.
- PTP via multicast, unicast, or in hybrid mode\*
- PTP-clock master / slave configuration is auto-negotiated between devices in the network. The module's master / slave state may change automatically.
- PTP profile selection (default E2E, default P2P, media E2E, media P2P, customized)
- Edit opens the "ADVANCED" tab to adjust the custom profile.

### **FAQs**

#### Q: What is RAV2 Module?

A: RAV2 Module is an audio network module for RAVENNA / AES67.

#### Q: How can I access the device settings?

A: Access the "STATUS" tab and click on the corresponding links to access device settings.

### Q: How can I adjust the clock source and sample rate?

A: On the "STATUS" tab, use the pulldown menus to select the desired clock source and adjust the sample rate.

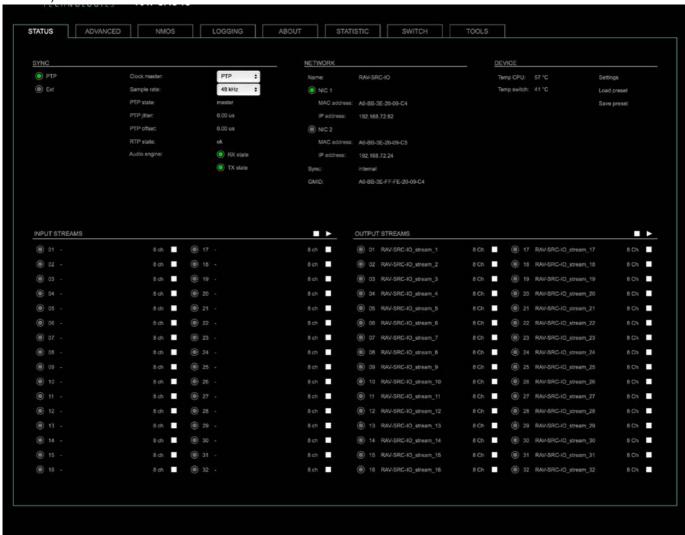
Q: What does the blinking state indicate for the audio engine?

A: The blinking state indicates that not all received packets can be processed or not all packets can be sent to the network.

#### Introduction

RAV2 is an audio network module for RAVENNA / AES67.

All functions of the device are accessible through a browser based interface (hmtl5 / javascript). The size of the window and the zoom level can be varied. The page is organized in tabs, pulldown menus or hyperlinks offer access to the values of a parameter. Some values use an input field (e.g. IP address).



To access the control page:

- · connect the network with one port
- enter http://<IP Address> (default IP @ PORT 1: 192.168.0.1) in the navigation bar of your browser

Two independent network interfaces (NICs) can be configured in the switch configuration. Port 1 is fixed assigned to NIC 1.

#### NOTE

If NIC 1 and NIC 2 are connected to the same switch, they must be configured to different subnets – see "Network Settings" on page 7.

#### Status - Overview

The tab 'STATUS' is divided into several sections:

- SYNC monitoring sync state, clock selection, links to I/O settings
- NETWORK display network info, link to network settings
- DEVICE monitoring device info, link to device settings, phones level control
- INPUT STREAMS monitoring and control input streams, link to input stream settings
- OUTPUT STREAMS monitoring and control output streams, link to output stream settings

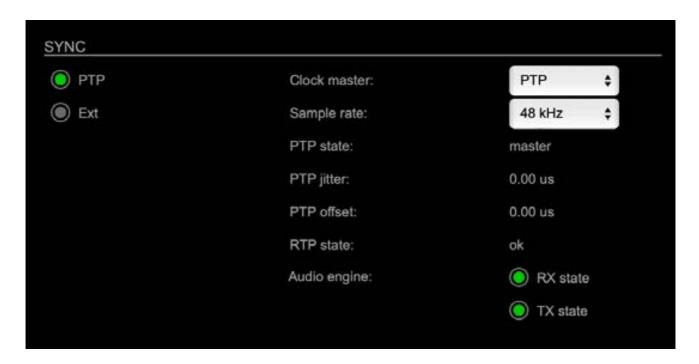
Hyperlinks open a popup window to adjust related settings. Most settings are updated immediately without further notification. To exit a popup window click the button in the top right corner.

Mouse overs are used to display further information (e.g. connection speed of network link).

#### **NOTE**

The web user interface updates itself when changes are applied by other instances (other browsers, external control commands).

#### Status - Sync



	Displays clock source and state for the main frame:				
	• (OFF) = not locked				
PTP, Ext	• (ON) = locked and in sync with clock master (				
	blinking) = locked but not in sync with clock master				
Clock master	Pulldown menu to select clock source of the main frame (PTP, extern)				
Sample rate	Pulldown menu to adjust sample rate of the main frame (44.1 / 48 / 88.2 / 9 6 / 176.4 / 192 kHz).				
PTP state	State of PTP (Master / Slave).				
PTP jitter	PTP-clock jitter per second				
PTP offset	Offet relative to PTP-clock master				
RTP state	Status of packet processing (OK, Error*)				
	State of module's audio engine- receiving				
Audio engine RX state	• (ON) = ok, receiving data				
	• (blinking) = not all received packets can be processed				
	State of module's audio engine- sending				
Audio engine TX state	• (ON) = ok, sending data				
	• (blinking) = not all packets can be sent to the network				

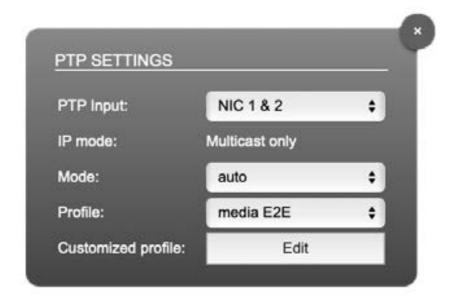
<sup>\*</sup> Error: packet time stamps are out of bounds.

Possible reasons: stream offset may be too small or transmitter or receiver are not synced properly to the Grandmaster.

## Hyperlinks:

PTP / PTP state (p 5)

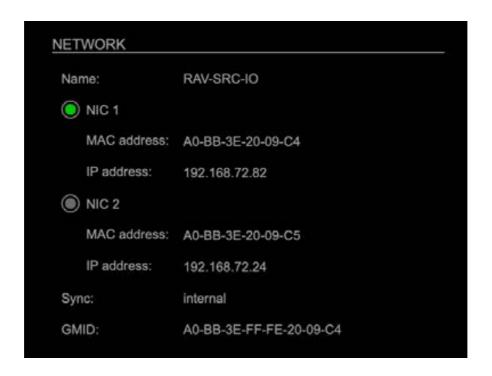
## **PTP Settings**



PTP Input	NIC selection for PTP clock input. 'NIC 1 & 2' means input redundancy.
IP Mode	PTP via multicast, unicast or in hybrid mode. *
Mode	PTP-clock master/slave configuration is auto-negotiated between devices in the network. Module's master/slave state may change automatically.
Profile	PTP profile selection (default E2E, default P2P, media E2E, media P2P, cu stomized)
Customized profile	Edit opens the tab 'ADVANCED' to adjust the custom profile.

See "Advanced – PTP Clock Setting" on page 31 for more details.

### Status - Network



Name	Module's name in the network. Used e.g. for mDNS service. The name nee ds to be unique throughout the network.		
NIC 1 / NIC 2	Monitoring state of network interface controller  • (OFF) = not connected  • (ON) = connected with the network		
MAC address	Hardware identification of network interface controller.		
IP address	IP address of device		
Sync	Selected NIC for PTP sync		
GMID	Grand Master ID (PTP)		

### Hyperlinks

Name / IP address (p 7)

#### Mouse over:

- LED NIC 1 indicating link state and connection speed
- LED NIC 2 indicating link state and connection speed

#### **NOTE**

If NIC 1 and NIC 2 are connected to the same switch, they must be configured to different subnets – see "Network Settings" on page 7.

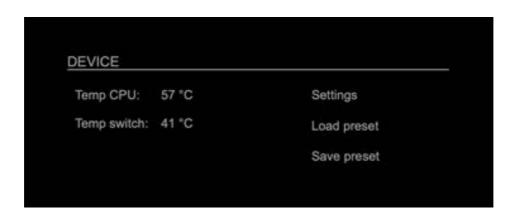
### **Network Settings**

The two network interface controllers (NIC 1 / NIC 2) are configured individually.

IC 1		NIC 2	
Dynamic IP address (IPv4):	•	Dynamic IP address (IPv4):	
Static IP address (IPv4):	•	Static IP address (IPv4):	•
IP address (IPv4):	192.168.72.82	IP address (IPv4):	169.254.61.155
Subnet mask (IPv4):	255.255.255.0	Subnet mask (IPv4):	255.255.0.0
Gateway (IPv4):	0.0.0.0	Gateway (IPv4):	0.0.0.0
DNS server (IPv4):	0.0.0.0	DNS server (IPv4):	0.0.0.0
Direct routing (multicast transfer on	ly):	Direct routing (multicast transfer on	y):
Host IP address 1:	172.44.56.12	Host IP address 1:	0.0.0.0
Host IP address 2:	0.0.0.0	Host IP address 2:	0.0.0.0
Host IP address 3:	0.0.0.0	Host IP address 3:	0.0.0.0
Host IP address 4:	0.0.0.0	Host IP address 4:	0.0.0.0
Host IP address 5:	0.0.0.0	Host IP address 5:	0.0.0.0
Host IP address 6:	0.0.0.0	Host IP address 6:	0.0.0.0
☐ Host IP address 7:	0.0.0.0	Host IP address 7:	0.0.0.0
Host IP address 8:	0.0.0.0	☐ Host IP address 8:	0.0.0.0
☐ Host IP address 9:	0.0.0.0	☐ Host IP address 9:	0.0.0.0
Host IP address 10:	0.0.0.0	☐ Host IP address 10:	0.0.0.0

Device name	Input field – Module's name in the network. Used e.g. for mDNS service. The name needs to be unique throughout the netwo rk.		
Dynamic IP address (IPv4)	Switch to enable the device's DHCP client.  IP address is assigned by DHCP server. If no DHCP is available the IP address is determined via Zeroconf.		
Static IP address (IPv4)	Switch to disable the device's DHCP client. Manual configuration of networ k parameters.		
IP address (IPv4)	Module's IP Address		
Subnet mask (IPv4)	Module's subnet mask		
Gateway (IPv4)	IP address of gateway		
DNS server (IPv4)	IP address of DNS server		
Apply	Button to confirm changes. Another popup window will appear to confirm a reboot of the module.		
Direct routing	IP addresses of devices outside the subnet, to enable multicast traffic; e.g. Grandmaster or IGMP querier.  Mark checkbox to activate.		

## Status - Device



Temp CPU	Display temperature of CPU core in degree Celsius. It may reach 95 °C wit hout effecting the performance of the device.			
Temp switch	Display temperature of network switch in degree Celsius			
Settings	Opens a popup window to configure the device.			
Load preset	Opens a dialog to store the device settings to a file. Filetype: .rps			
Save preset	Opens a dialog to restore the device settings from a file.  Filetype: .rps			

## Hyperlinks:

- Settings (p 8)
- Load preset (p 9)
- Save preset

## Settings



AoIP Module SW	Module's software version. It is updated together with hardware version via network.
AoIP Module HW	Module's bitstream version. It is updated together with software version via network.
AoIP Module Update	Opens a dialog for selection of the update file – see <u>"RAV2- Firmware Update" on page 43.</u>
AoIP Module Reboot	Restart of the AoIP module. Confirmation required. Audio transmission will be interrupted.
Language	Menu language (english, german).
Manufacturer Settings Reset	Restore device settings to factory defaults. Confirmation required.

### **Load Preset**



The device configuration can be stored to a single file (.rps).

Restoring the configuration a dialog prompts for selection of individual settings. This enhances flexibility at setup changes when a particular adjustment shall be preserved or just a single adjustment shall be restored.

### Status - Input Streams

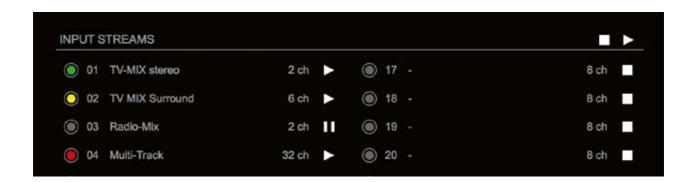
0	01	TV-MIX stereo	2 ch	•	17 -	B ch
0	02	TV MIX Surround	6 ch	•	18 -	8 ch
•	03	Radio-Mix	2 ch	П	19 -	8 ch
0	04	Multi-Track	32 ch	•		8 ch
0	05		8 ch		21 -	8 ch
0	06		8 ch		22 -	8 ch
•	07		8 ch		23 -	8 ch
0	08		8 ch		24 -	8 ch
•	09		8 ch		25 -	8 ch
•	10		8 ch			8 ch
•	11		8 ch		27 -	8 ch
•	12		8 ch		⊚ 28 -	8 ch
0	13		8 ch		29 -	8 ch
•	14		8 ch		⊚ 30 -	8 ch
0	15		8 ch		31 -	8 ch

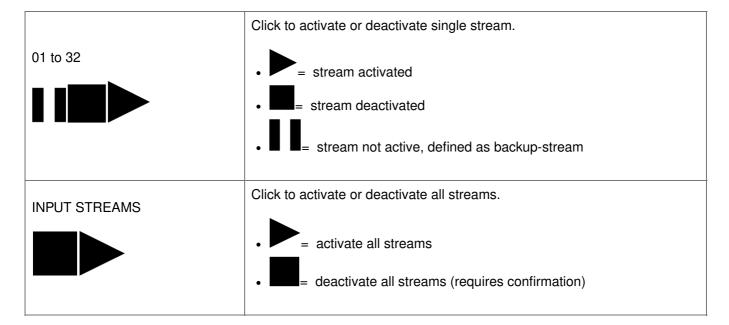
The module can subscribe up to 32 streams. The overview displays the basic information of each stream. The input stream name can be set manually

(discovery protocol: manually, see page p 19) overriding the SDP's stream name information.

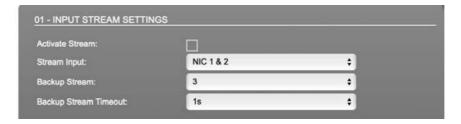
A backup stream can be defined as source after an adjustable timeout. A central active / inactive switch allows to toggle the stream state of all input streams at once.

	State of incoming streams		
01 to 32	O(OFF) = stream not activated  o(ON) = stream activated, receiving data  o(ON) = stream activated, receiving data via one NIC only (in put redundancy)  o(blinking) = stream activated, not receiving data  (unicast, connection not established)		
01 to 32 Name	Name of stream gathered from SDP or set manually in the stream settings dialog.		
01 to 32 xx ch	Number of audio channels transported by the stream		



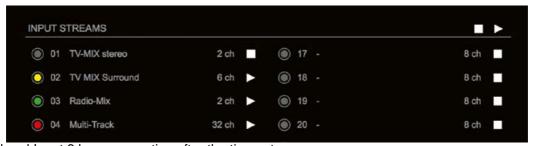


#### **Backup Streams**



#### **Example:**

Backup stream (input 3) that will act as source in the audio matrix if the current session (input 1) fails. Switch-over occurs after the defined timeout (1s). Stream 3 is marked accordingly in the status view



Input 1 failed and Input 3 becomes active after the timeout.

#### **NOTE**

In case the main input fails the main stream is stopped (IGMP LEAVE) before the backup stream is being activated. This behaviour ensures that the required network bandwidth does not increase in case of a failure.

NPI	JT S	STREAMS				
0	01	TV-MIX stereo	2 ch	•	⊚ 17 -	8 ch
0	02	TV MIX Surround	6 ch	•	18 -	8 ch
•	03	Radio-Mix	2 ch	н	19 -	8 ch
0	04	Multi-Track	32 ch		20 -	8 ch

### Hyperlinks:

• Name (p 14)

#### Mouse over:

• LED – indicating stream state

#### **NOTE**

Source-Specific Multicast (SSM) support for IGMP v3, v2 and v1 (SSM via protocol only in IGMP v3, SSM via internal filtering is applied for IGMP v2 and v1) – see "Source Specific Multicast" on page 19.

### **Input Stream Settings**



Up to 32 input streams can be subscribed. Each stream is organized in a 'RAVENNA session' (SDP = Session Description Protocol) that describes the stream parameters (audio channels, audio format, etc.).

The stream settings allow to adjust the processing of the received audio data (offset, signal routing). The receiving of stream data starts once the stream has been enabled.

The settings displayed vary depending on the selected discovery protocol.

### **TIP**

A sample offset of at least doubled packet time (samples per frame) is recommended

Example: Samples per frame = 16 (0.333 ms) 

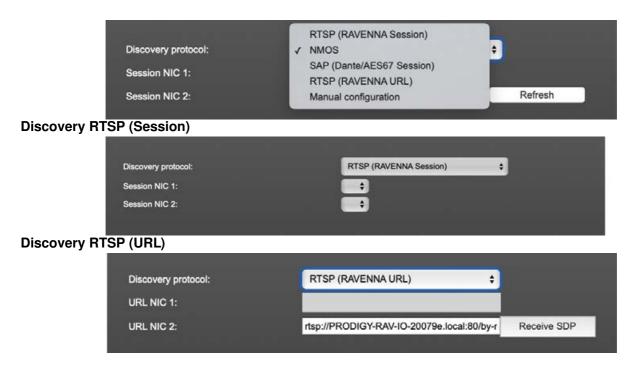
○ Offset ≥ 32 (0.667 ms)

It may be helpful to alter the stream discovery protocol if an expected stream can't be discovered by the device.

tores parameters and activates or deactivates the receiving of audio data.			
Inicast: additionally the negotiation of the connection)			
elects one or both NICs used for stream input. Both NICs means input red ndancy.			
elects a backup stream that will act as source in the audio matrix if the curent session fails. Switch-over occurs after the defined timeout.			
efines timeout [1 s to 120 s] before switch-over to backup stream.			
ame of stream gathered from SDP			
formation about stream state: connected			
ot connected receiving data read successfull			
ror			
tatus info related to stream state.			
easured value (maximum). A high value indicates that the media offset of e source might not match the adjusted media offset of the device.			
easured value (minimum). The offset should not become negative.			
ulticast address of input stream subscribed at NIC 1 / NIC 2.			
nicast transmission: IP address of sender.			
ounter indicates the number of incidents where the network connection was lost (link down)			
ounter indicates the number of lost RTP packets			
ounter indicates the number of packets with invalid timestamp			
counter marcales the number of packets with invalid timestamp			
nables adjustment of offset in increments of one sample.			
odules output delay of received audio data (input buffer).			

Start channel	Assignment of first stream channel in the audio matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix.			
Discovery protocol	Connection protocol or manual setup. RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol			
Session NIC 1	Selection of discovered streams at NIC 1			
Session NIC 2	Selection of discovered streams at NIC 2			

Stream Discovery in AoIP environments is a colorful mixture of different mechanisms. To serve a successful stream management RAV2 provides a bunch of options, not making operation easier but effective.

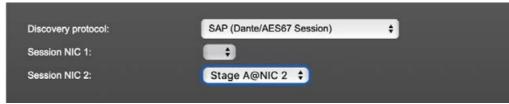


	URL (Uniform Ressource Locator) of the session of the device that is serving streams.
URL	Examples: rtsp://192.168.74.44/by-id/1 or rtsp://PRODIGY-RAV-IO.local:80/by-name/Stage_A
Receive SDP	Recalls the stream configuration of the defined session(s).

#### NOTE

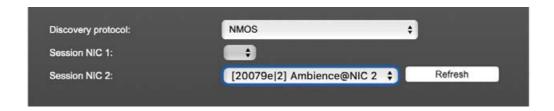
In case the automatic stream announcement and discovery of RAVENNA streams fails or cannot be used in a given network, the stream's SDP file can also be obtained via an RTSP URL.

#### **Discovery SAP**



SAP is used in Dante environments.

### **Discovery NMOS**



Session	[MAC Address of sender] stream name @NIC
Refresh	Initiates a scan for available streams.

NMOS is suited for use in SMPTE ST 2110 environments.

## **Manual Setup**



Stream name (manual)	Stream name for display in status view and matrix. Can be specified individ ually, different than the name gathered from the SDP.
Number of channels	Number of audio channels in the stream
RTP-Payload-ID	RTP-Payload-ID of the audio stream (Real-Time Transport Protocol). Descr ibes the format of the transported content.
Audio Format	Stream's audio format (L16 / L24 / L32 / AM824)
Media Offset	Offset between stream's timestamp and PTP-clock
Dst IP address	Multicast IP address of audio stream
SSM	Activate Source Specific Multicast filter for this stream.*
Src IP address	IP address of sending device.*
RTP dst port	Stream's destination port for RTP
RTCP dst port	Stream's destination port for RTCP (Real-Time Control Protocol)

<sup>\*</sup> An RTP packet contains the IP address of the sender (source IP) and the stream's multicast address (destination IP). With SSM activated the receiver only accepts RTP packets of a certain destination IP that are originated by a sender with the specified source IP.

#### **NOTE**

RTP Payload ID must match between sender and receiver.

### Status - Output Streams

OUTPUT	TSTREAMS						•
01	Stage-A	8 Ch		0	17	8 Ch	
02	Stage-B	8 Ch	•	0	18	8 Ch	
03	Stage-C	8 Ch	•	$\odot$	19	8 Ch	
04		8 Ch		$\odot$	20	8 Ch	
05		8 Ch		$\odot$	21	8 Ch	
06		8 Ch		$\odot$	22	8 Ch	
07		8 Ch		$\odot$	23	8 Ch	
08		8 Ch		$\odot$	24	8 Ch	
O9		8 Ch		$\odot$	25	8 Ch	
10		8 Ch		$\odot$	26	8 Ch	
11		8 Ch		$\odot$	27	8 Ch	
12		8 Ch		$\odot$	28	8 Ch	
13		8 Ch		0	29	8 Ch	
14		8 Ch		$\odot$	30	8 Ch	
15		8 Ch		$\odot$	31	8 Ch	
16		8 Ch		$\odot$	32	8 Ch	

The device can send up to 32 streams. The overview displays the basic information of each stream.

01 to 32	State of outgoing streams  • (OFF) = stream not activated  • (ON) = stream activated, sending data  • (ON) = stream activated, stream output via both NICs selected, but one NIC is not linked to the network.
01 to 32 Name	Name of stream defined in the settings
01 to 32 xx ch	Number of audio channels transported by the stream
01 to 32	Activate or deactivate stream.  = stream activated  = stream deactivated
OUTPUT STREAMS	Click to activate or deactivate all streams.  = activate all streams  = deactivate all streams (requires confirmation)

### Hyperlinks:

• Name (p 22)

Mouse over:

• LED - indicating stream state

### TIP

AES67 Streams

To create output streams for interoperability in AES67 environments please consult the information document Info – AES67 Streams.

### TIP

SMPTE 2110-30 / -31 Streams

To create output streams for interoperability in SMPTE ST 2110 environments please consult the information document Info – ST2110-30 Streams.

Both documents are available at http://academy.directout.eu.

### **Output Stream Settings**

02 - OUTPUT STREAM SETTINGS		
Activate Stream:		
Stream Output:	NIC 2 \$	
Stream name (ASCII):	Ambience	
RTSP URL (HTTP tunnel) (by-name):	rtsp://PRODIGY-RAV-IO-20079e.local:80/by-name/Ambience	
RTSP URL (HTTP tunnel) (by-id): RTSP URL (by-name):	rtsp://PRODIGY-RAV-IO-20079e.local:80/by-id/2	
RTSP URL (by-name): RTSP URL (by-id):	rtsp://PRODIGY-RAV-IO-20079e.local/by-name/Ambience rtsp://PRODIGY-RAV-IO-20079e.local/by-id/2	
SDP:	v=0 o=- 2 2482002937 IN IP4 192.168.74.44 s=Ambience t=0 0 m=audio 5004 RTP/AVP 98	0
Unicast:		
RTP payload ID:	98	
Samples per Frame (packet time):	45 (0.94 ms) \$	
Audio format:	AM824 \$	
Start channel:	9 💠	
Number of channels:	8 \$	
NIC 1	NIC 2	
RTP dst port: 5004	RTP dst port: 5004	
RTCP dst port: 5005	RTCP dst port: 5005	
The state of the s		

Up to 32 output streams can be sent to the network. Each stream is organized in a session (SDP = Session Description Protocol) that describes the stream parameters (audio channels, audio format, etc.). Each stream may be labelled with an individual stream name (ASCII) which is useful for enhanced comfort at organizing the setup.

The stream settings allow to adjust the processing of the sent audio data (blocks per frame, format, signal routing, ...). The sending of stream data starts once the stream has been enabled.

Once the stream is active, the SDP data is displayed and may be copied from the window or downloaded via http://<rav-io>/sdp.html?ID=<streamno.>.

Activate stream	Stores parameters and activates or deactivates the receiving of audio dat a. (Unicast: additionally the negotiation of the connection)
Stream Output	Selects one or both NICs used for stream output. Both NICs means output redundancy.
Stream name (ASCII)	Individually defined name of an output stream. It is used in the URL which is indicated in different ways below.*
RTSP URL (HTTP tunnel) (by-nam e) / (by id)	Current used RTSP-URL of stream with HTTP port used for RTSP, stream name or stream id.
RTSP URL (by-name) / (by id)	Current used RTSP-URL of stream with stream name or stream id.
SDP	SDP data of the active stream.
Unicast	If activated, the stream is sent in unicast mode.**
RTP payload ID	Stream's payload id
Samples per Frame	Number of blocks containing payload (audio) per ethernet frame – see pa cket time on p 14.
Audio format	Stream's audio format (L16 / L24 / L32 / AM824) ***
Start channel	Assignment of first stream channel from the audio matrix. E.g. stream with eight channels, starting at channel 3 is fed from channel 3 to 10 of the rou ting matrix.
Number of channels	Number of audio channels in the stream.
RTP dst port	Stream's destination port for RTP
RTCP dst port	Stream's destination port for RTCP (Real-Time Control Protocol)
Dst IP address (IPv4)	Stream's IP address for multicast (should be unique for each stream).

- 1. Only ASCII characters are allowed.
- 2. A unicast stream can only be received by one device. If a device is already receiving the stream, further connection calls by other clients are answered with ,service unavailable (503). The release time after disconnect or interruption of the client's connection amounts to about 2 minutes.
- 3. L16 = 16 bit audio / L24 = 24 bit audio / L32 = 32 bit audio / AM824 = standardized according to IEC 61883, allows AES3 transparent transmission (SMPTE ST 2110-31).

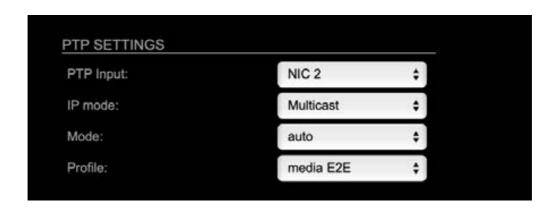
#### Advanced - Overview



The tab 'ADVANCED' is divided into several sections:

- PTP SETTINGS definition of PTP source, mode and profile
- PTP PROFILE CURRENT SETTINGS definition of a customized PTP profile
- CURRENT PTP MASTER monitoring PTP characteristics
- PTP STATISTIC monitoring device's PTP state, jitter and delay
- PTP CLOCK SETTINGS definition of adaption algorithms to reduce jitter
- NETWORK ADVANCED SETTINGS definition of network and QoS characteristics
- PTP JITTER graphical display of measured PTP jitter

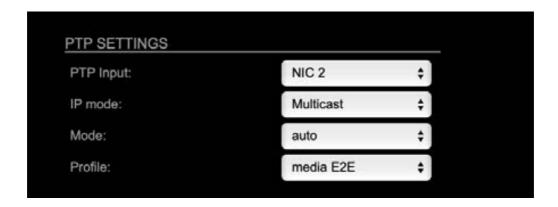
### Advanced - PTP Settings



PTP Input	Selects one or both network ports used for PTP input. Both ports means input redunda ncy. *
IP Mode	Multicast = Sync messages and delay request are sent as multicast message to ever y node within the network.  Hybrid = Sync messages are sent as multicast, delay requests are sent as u nicast messages directly to the Grandmaster or Boundary Clock.**  Unicast = Sync messages are sent as unicast,  delay requests are sent as unicast messages directly to the Grandmaster or Boundary Clock.***

<sup>\*</sup> Using redundant PTP-operation a switch-over is triggered not only at signal loss of the Grandmaster but depends on the quality of the PTP clock. Changes (e.g. clock class) are observed permanently and the algorithm decides for the best signal present.

- \*\* Hybrid Mode reduces the workload for all nodes in the network as they do not receive the (unnecessary) delay requests from other devices anymore.
- \*\*\* Unicast Mode may help when multicast routing is not possible within the network. As an opposite to the Hybrid Mode it increases the workload of the grandmaster since sync messages must be sent to each single slave individually.



Mode	auto = PTP-clock master / slave configuration is auto negotiated betw een devices in the network. Module's master / slave state may change automatically.  slave only = PTP-clock slave configuration is  preferred. Module clocks to another device in the network  preferred master = PTP-clock master configuration is  preferred. Module acts as network grandmaster. Priority values are adjusted automatically to ensure Grandmaster status. *  master only = PTP-clock master is forced. **
Profile	Selects predefined PTP profile (default E2E, default P2P, media E2E, media P2P) or a ctivates customized PTP profile.

<sup>\*</sup> If more than one device announces as PTP-clock master the network Grandmaster is determined following the Best Master Clock Algorithm (BMCA).

<sup>\*\* &#</sup>x27;Master only' configures the device to act as Unicast Grandmaster. This setting is available only with PTP Mode set to 'unicast'

#### **NOTE**

PTP profile ,customized' allows for individual adjustment of the PTP parameters. If profile is set to ,media' or ,default' the PTP parameters cannot be altered and are displayed only. Factory default setting is PTP Media Profile E2E.

### Advanced - PTP Unicast



Auto Detect GM	on = enables the automatic detection of the grandmaster * off = IP address of grandmaster needs to be defined manually
Grant duration (sec)	Time period during which the slave receives sync messages from the grandmaster.**
Grandmaster IP	IP address of the grandmaster. ***

<sup>\* &#</sup>x27;Auto Detect GM' is a proprietary function and might not be supported by 3rd party GMs.

### **About PTP Unicast**

Since the BMCA is not available with PTP unicast, the PTP properties of the devices require some extra configuration.

### **Example:**

Grandmaster	IP Mode Unicast, Mode Master only
	IP Mode Unicast, Mode Slave Only,
Slave(s)	Auto Detect GM ON, Grant Duration 30 sec

### Advanced – PTP Profile Customized Settings

<sup>\*\*</sup> Depending on the temporary workload of the grandmaster the negotiation may fail.

<sup>\*\*\*</sup> This value is used only with 'Auto Detect GM' set to <off>.

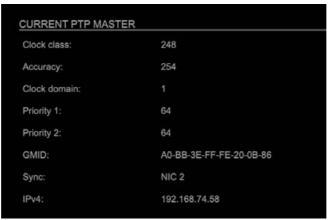
Clock class:	248
Accuracy:	254
Clock domain NIC 1:	0
Clock domain NIC 2:	1
Priority 1:	128
Priority 2:	128
Announce:	2 s (1)
Sync:	125 ms (-3)
Min delay request:	1 s (0)
Min pdelay request:	1 s (0)
Announce receipt timeout:	3
One step clock:	no
Slave only:	no
Delay mechanism:	E2E

The settings become available with PTP profile set to 'customized'.

Clock class	PTP-clock's class according to IEEE 1588 [read only]
Accuracy	PTP-clock's accuracy according to IEEE 1588 [read only]
Clock domain NIC 1	PTP-clock's domain at NIC 1
Clock domain NIC 2	PTP-clock's domain at NIC 2
Priority 1	Priority setting for master announcement (the smaller the value the higher t he priority)
Priority 2	If value 'Priority1' (and other PTP-clock parameters) of more than one devic e in the network match:  Priority setting for master announcement (the smaller
	the value the higher the priority)
Announce	Intervall of sending announce-packets for auto- negotiation.
Sync	Intervall of sending sync-packets to the PTP-clock slaves in the network.
Min delay request	Intervall of sending End-To-End packets of PTP-clock slave to PTP-clock m aster. To determine the offset slave-to-master.
Min pdelay request	Intervall of sending Peer-To-Peer packets between two PTP-clocks. To determine the offset master-to- slave and slave-to-master.
Announce receipt timeout	Number of missed announce-packets (threshold) to reinitialize the negotiati on of PTP-clock master.

One step clock	Timestamp of PTP-clock is integrated in PTP-sync- packets. No follow-up p ackets are sent.  No = Two step clock is used
Slave only	Yes = PTP-clock is always slave.
Delay mechanism	E2E – Offset slave-to-master is determined by End-To- End packets.  P2P – Offset master-to-slave and slave-to-master is determined by Peer-To-Peer packets.

### **Advanced – Current PTP Master**



## Monitoring display only.

Clock class	PTP-clock's class according to IEEE 1588
Accuracy	PTP-clock's accuracy according to IEEE 1588
Clock domain	PTP-clock's domain at selected NIC
Priority 1	Priority setting for master announcement (the smaller the value the higher t he priority)
Priority 2	If value 'Priority1' (and other PTP-clock parameters) of more than one devic e in the network match:  Priority setting for master announcement (the smaller the value the higher the priority)
GMID	ID of current Grandmaster
Sync	Selected NIC for PTP clock
IPv4	IP address of Grandmaster

### Advanced - PTP Statistic



### Monitoring display only.

PTP state	Information about current PTP-clock state: intialize error deactivated receiving data pre master master passive not calibrated slave
PTP jitter	PTP-clock jitter in microseconds (μs)
PTP offset	Offset relative to PTP-clock master
PTP master to slave	Absolute offset master-to-slave in nanoseconds
PTP slave to master	Absolute offset slave-to-master in nanoseconds
Current PTP time (TAI):	Date and time information from GPS source*
Current PTP time (TAI) (RAW):	RAW TAI from GPS source*

 $<sup>^{\</sup>star}$  Temps Atomique International – if no GPS source is available for PTP time- stamping, the date / time display starts at 1970-01-01 / 00:00:00 after every reboot of the device.

### Advanced - PTP Clock Setting



No PTP Switch 1 Gbit/s	Adapted PTP-clock algorithm to reduce clock jitter using 1 GB network switches witho ut PTP support.  Max. number of 1 Gbit/s switches: less than 10
No PTP Switch 100 Mbi t/s	Adapted PTP-clock algorithm to reduce clock jitter using 100 MB network switches wit hout PTP support.  Max. number of 100 Mbit/s switches: 1

IGMP NIC 1:	auto	<b>*</b>
IGMP NIC 2:	auto	<b>\$</b>
TCP port HTTP:	80	
TCP port RTSP:	554	
TTL RTP packets:	128	
DSCP RTP packets:	AF41 (0x22)	<b>‡</b>
DSCP PTP packets:	CS6 (0x30)	<b>*</b>
Multi stream rx:	yes	\$
MDNS announcement:	RX/TX	<b>‡</b>
SAP announcement:	RX/TX	<b>\$</b>
Network settings:	Apply	

IGMP NIC 1	Definition or auto-select of IGMP version used to connect to a multicast router at NIC 1.
IGMP NIC 2	Definition or auto-select of IGMP version used to connect to a multicast router at NIC 2
TCP port HTTP	TCP port for HTTP
TCP port RTSP	TCP port for RTSP
TTL RTP packets	Time-To-Live of RTP packets – default: 128
DSCP RTP packets	DSCP marking of QoS of RTP packets – default: AF41
DSCP PTP packets	DSCP marking for QoS of PTP packets – default: CS6*
Multi stream rx	If activated, the device allows to subscribe to the same multicast stream more than one time – default: off
MDNS	Announcement of streams via MDNS can be controlled to optimize network traffic or CPU load.
announcement	Values: Off, RX, TX or RX/TX **
SAP announcement	Announcement of streams via SAP can be controlled to optimize network traffic o r CPU load.
SAP announcement	Values: Off, RX , TX or RX/TX **
Network settings Apply	Confirms and saves changes being made. Reboot required.

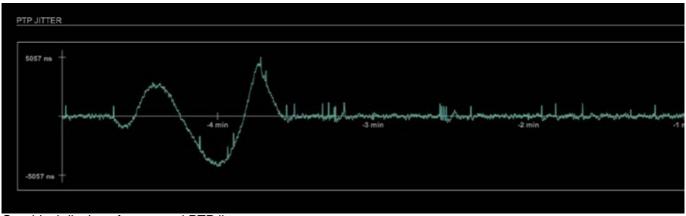
<sup>\*</sup> AES67 specifies EF, but some implementations use EF for Audio streaming. To avoid overlapping of RTP and PTP packets in the same queue CS6 has been chosen as default.

<sup>\*\*</sup> RX = receive, TX = transmit, RX/TX = receive and transmit

#### **NOTE**

Source-Specific Multicast (SSM) support for IGMP v3, v2 and v1 (SSM via protocol only in IGMP v3, SSM via internal filtering is applied for IGMP v2 and v1) – see "Source Specific Multicast" on page 19.

#### Advanced - PTP Jitter



Graphical display of measured PTP jitter.

#### **NOTE**

An error message next to Jitter measurement is displayed if delay requests are not being answered by Grandmaster.

#### **NMOS – Overview**



NMOS provides a family of specifications related to networked media for professional applications. It is produced by the Advanced Media Workflow Association (AMWA).

Support for NMOS is introduced with the AoIP Module version SW 0.17 / HW 0.46 according to the specifications:

- IS-04 Discovery & Registration
- IS-05 Device Connection Management

IS-04 allows control and monitoring applications to find the resources on a network. Resources include Nodes, Devices, Senders, Receivers, Sources, Flows...

IS-05 provides a transport-independent way of connecting Media Nodes.

More information: <a href="https://specs.amwa.tv/nmos/">https://specs.amwa.tv/nmos/</a>

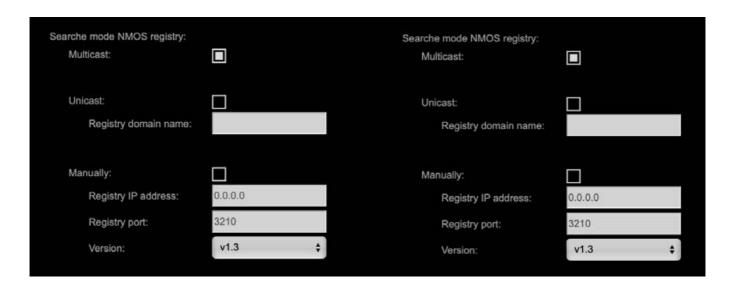
### NMOS port - NIC1 & NIC2

The port entries for NIC1 and NIC2 are pre-configured by default. Modifications are possible but not necessary.



NMOS port (NIC1 + NIC2)	Port address. Reboot required after modification.
-------------------------	---

### Search mode NMOS registry



Multicast	use mDNS to determine and connect to the registry server
Unicast	use DNS-SD to connect to the registry server
Registry domain name	DNS resolvable domain name of the registry server
Manually	
Registry IP address	
Registry port	
Version	Support of NMOS API version

### **NMOS – Additional Settings**

Disable stream during config:	П
Seed id:	d6368ef8-5b74-4508-ad07-652dac4f7796
Generate new seed id:	Generate

Disable stream during config	Automatically disable and re-enable streams when settings are changed via NMOS (recommended)
Seed id	Unique identifier, subordered entities are derived from the seed id.
Generate new seed id Generate	Generates a new unique identifier. Reboot required.

NMOS uses a logical data model based on the JT-NM Reference Architecture to add identity, relationships and time-based information to content and broadcast equipment. Hierarchical relationships group related entities, with each entity having its own identifier.

The identifiers are persistent across restarts of the device in order to make them useful over a period of time longer than a single production deployment.

New identifiers may be generated manually if required.

### Logging



The tab 'LOGGING' displays logging depending on the 'Log Settings'. The logging can be enabled individually for different protocols, each of with an adjustable filter. An adjustable log level specifies the information detail of each entry.

To save a log the content of the view can be copied and pasted to a text document.

### Log Level

0	log data
1	level and log data
2	protocol, level and log data
3	protocol, process-id of requesting process, process-id of running process, level and log data
4	protocol, process-id of requesting process, process-id of running process, level, process or time in ticks and log data
5	protocol, process-id of requesting process, process-id of running process, level, process or time in ticks, file name and line and log data

## **Protocol Types**

ARP	Address Resolution Protocol
BASE	Basic operation of module
DHCP	Dynamic Host Configuration Protocol
DNS	Domain Name System
FLASH	Process for updating the module
IGMP	Internet Group Management Protocol
MDNS	Multicast Domain Name System
NMOS	Network Media Open Specification
PTP	Precision Time Protocol
RS232	Serial Protocol
RTCP	Real Time Control Protocol
SAP	Session Announcement Protocol
TCP	Transmission Control Protocol
Zeroconf	Zero Configuration Protocol

# Log Filter

NONE	logging disabled
ERROR	error occurred
WARNING	warnings- condition that may lead to unwanted behavior or an error
INFO 1	log info* + warning + error
INFO 2	log info* + warning + error
INFO 3	log info* + warning + error
INFO 4	log info* + warning + error

### **Log Operation**

Save log	Downloads the current log entries to a text-file (log.txt).
Clear log	Deletes all log entries without further prompt.
Scroll lock	Interrupts automatic scrolling of the list view to allow copying the content to a text file via copy & paste. If scrolling is stopped for a longer period of time the display may not list all entries.

### **Statistic**

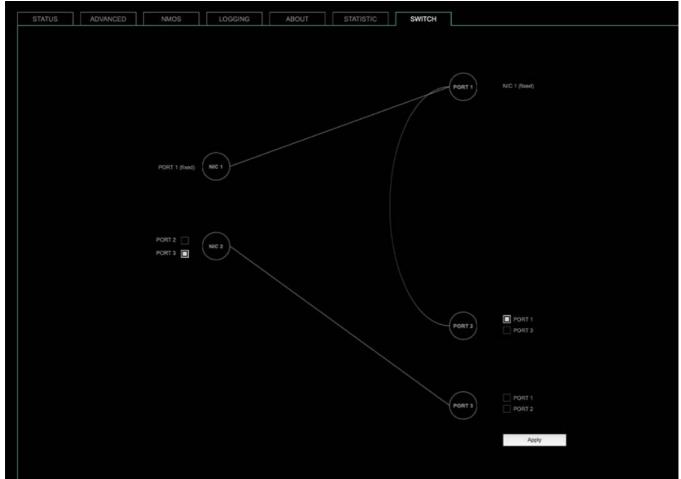


The tab 'STATISTIC' displays an overview of the CPU load of the particular processes, an error counter and a monitor display to indicate the incoming (RX) and outgoing (TX) network traffic on both network ports individually.

Details	Displays a list of input streams and related events (connection lost, packet I ost, wrong timestamp) of received audio packets.
Reset	Resets the packet statistic

See "Protocol Types"

#### **Switch**



Two independent network interfaces (NICs) can be configured in the switch configuration.

Port 1 is fixed assigned to NIC 1.

The other ports can be assigned to either NIC 1 or NIC 2

### **NOTE**

If you want to use a port that is not assigned to a NIC e.g. to patch the device's management port (MGMT) into the audio network, you can link it to one of the audio ports.

#### **NOTE**

To access the module's control page it is required to connect the management network to one of the ports that is directly attached to a NIC – see next page.

To give the very best PTP synchronisation performance, the switch incorporates advanced timestamping between the external PORTS and the internal NICs. As a consequence, the on-board switch cannot be used to connect other PTP devices via a single shared connection to the wider network.

Please connect all other PTP devices directly to your system's network switch.

#### **Tools**



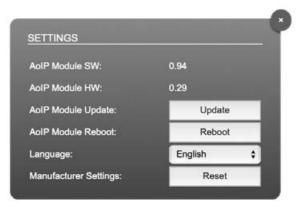
The tab 'TOOLS' offers a generator to ping any IP address (IPv4) from either NIC 1 or NIC 2. The result is displayed at the 'Output'.

IP address (IPv4)	Enter IP address (IPv4) to be pinged
Interface	Select NIC 1 or NIC 2
Start	Sends ping to the specified IP address from selected NIC.

### **RAV2 – Firmware Update**

The RAV2 module is updated via network.

Open the control page of the module and navigate to the tab STATUS and click SETTINGS in the top right corner (p 8).

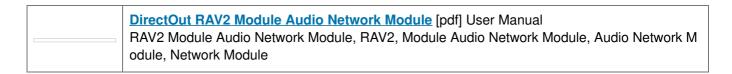


Click 'Update' and browse to the update file after unzipping first. Example: rav\_io\_hw\_0\_29\_sw\_0\_94.update Folllow the instructions displayed.

#### **WARNING!**

It is strongly recommended to backup the device configuration (Save Preset) before running any update.

#### **Documents / Resources**



- OkonsoleH :: Login
- Onmos | Networked Media Open Specifications: Introduction
- User Manual

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