

# zoom F6 MultiTrack Field Recorder Operation Manual

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## zoom F6 MultiTrack Field Recorder Operation Manual



## **Notes about this Operation Manual**

You might need this manual in the future. Always keep it in a place where you can access it easily. The contents of this document and the specifications of the product could be changed without notice.

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#### Introduction

Thank you very much for purchasing a ZOOM F6 multitrack field recorder

The F6 provides the following features in a compact form.

## · Record the quietest and loudest sounds at high quality with 32-bit float WAV format

The high-quality analog input circuits can handle signals ranging from the most delicate to a professional maximum level of +24 dBu.

In addition to 16/24-bit WAV recording, 32-bit float WAV recording, which does not require input level adjustment, is also supported.

With 32-bit float WAV format, the recording resolution can be retained even when changing levels greatly after recording.

### · Simultaneously record 6 channels and 14 tracks

Up to 14 tracks can be recorded simultaneously, including 16/24-bit WAV and 32-bit float WAV for Inputs 1–6 along with left and right tracks of a stereo mix.

### · Support for three types of batteries

A USB mobile battery, L battery or AA batteries can be used for power.

#### Two remote control options

Wireless control is possible by installing a ZOOM wireless adapter (e.g. BTA-1) and using the Control iOS app.

Moreover by connecting an Control, which is a mixer-style controller designed especially for F Series recorders, with a USB cable, 60mm track faders, LED level meters and various transport buttons can be

used for intuitive sound control. Combined with the Control iOS app, iPhones and iPads can also be used as large meters with excellent visibility.

#### · Support for SMPTE timecode input and output along with wireless timecode input

The uses a high-precision oscillator that enables it to independently generate accurate timecode with a discrepancy of less than 0.5 frames per 24 hours.

If a BTA-1 dedicated wireless adapter is installed, wireless timecode can be received from a Timecode Systems UltraSync BLUE and written to recorded files.

## Headphone jack with 100mW+100mW maximum output

Clear headphone monitoring is possible using the digital boost function while sending audio signals to a video camera or other device from the LINE OUT jack.

#### · Flexible signal routing also makes mixer use possible

Pre-fader and post-fader signals from inputs 1–6 can be routed to outputs freely.

Phantom power supply (+24 V or +48 V)

This can be set for each input separately.

• USB audio interface use with up to 6 ins and 4 outs possible

Use as a 2-in/2-out or 6-in/4-out audio interface (driver required for Windows).

· Output multitrack audio by USB while recording

While recording to the installed SD card, multitrack audio can be sent to and from a computer by USB with up to 8 inputs (6 inputs + L/R stereo mix) and 4 outputs.

This enables simultaneous backup recording and Internet live streaming.

360º audio

Ambisonic mode enables 360° spatial audio recording using VR mics. Decoding from Ambisonic format A to format B is supported along with gainand setting link functions.

## Achieving high audio quality throughout recording and editing

With the dual A/D converter circuits and support for 32-bit float WAV files, the audio quality from recording to post-production.

**F6** can maintain the highes

#### Recording



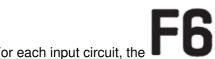
Dual A/D converter circuit enables recording both loud and quiet sounds without making gain adjustments

## **Post-production**



32-bit float WAV file format maintains audio quality from recording when editing

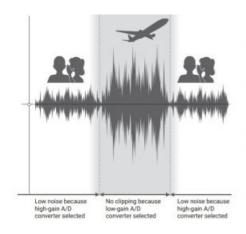
### Dual A/D converter circuit overview



quality audio recording without the need to adjust gain settings, a step that is normally indispensable.

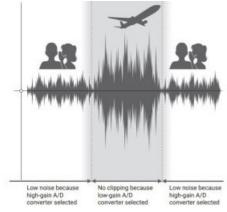
### Providing amazing dynamic range

By combining two A/D converters, a wide dynamic range not possible with a single A/D converter has been realized.



## Switching between two A/D converters

The constantly monitors data from the two A/D converters, and automatically selects the one that provides the best recording results.



#### 32-bit float WAV file overview

32-bit float WAV files have the following advantages over conventional 16/24-bit linear WAV files. These features enable the quality of the sound during recording to be maintained even during post production...

## Resolution advantage

32-bit float WAV files have the advantage of being able to maintain high resolution even at low volumes. As a result, quiet sounds can be made louder when editing after recording without degrading their quality.

#### 16/24-bit linear WAV



#### 32-bit float WAV



## Clipping advantage

If a waveform sounds clipped when output from the or in a DAW,

it can be edited after recording to lower its volume and restore an unclipped waveform because the data in the 32-bit float WAV file itself is not clipped.

#### 16/24-bit linear

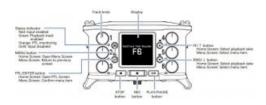


#### 32-bit float

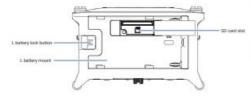


## Names of parts

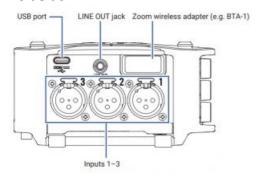
#### **Front**



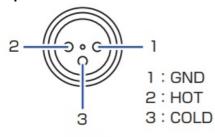
#### **Back**



#### Left side

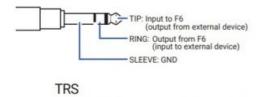


## Inputs 1-6

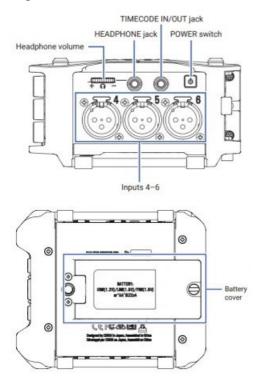


## XLR

**TIMECODE IN/OUT** 



## Right side



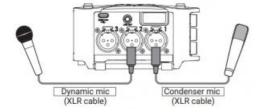
## Connecting mics/other devices to Inputs 1-6

The can record 6 individual tracks that correspond to Inputs 1–6 and a stereo mix of these inputs with left and right tracks.

Mics and the outputs of instruments and audiovisual equipment, for example, can be connected to Inputs 1–6 and recorded to tracks 1–6.

### **Connecting mics**

Connect dynamic and condenser mics with XLR plugs to Inputs 1–6. Phantom power (+24 V/+48 V) can be supplied to condenser mics. (  $\rightarrow$  P. 81)



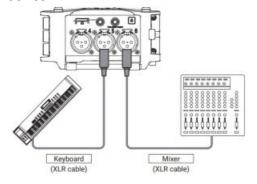
### NOTE:

When disconnecting an XLR cable, pull the XLR plug while pushing the connector lock release button.

## Connecting line level equipment

Connect XLR cables from keyboards and mixers directly to Inputs 1-6.

Direct input of passive guitars and basses is not supported. Connect these instruments through a mixer or effects device.



## **Equipment connection examples**

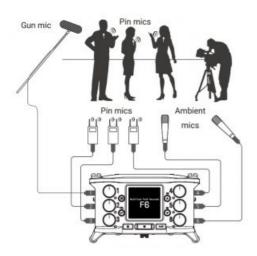
Recording is possible in a variety of situations like these.

## While filming

• Input 1: gun mic for main subject sound

• Inputs 2-4: lapel mics for performers

• Inputs 5-6: mics for ambient sound

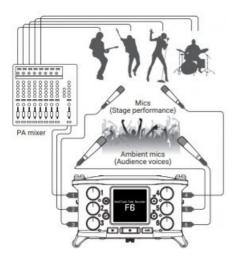


## **Concert recording**

• Inputs 1–2: line inputs for outputs from mixer

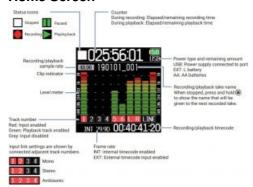
• Inputs 3–4: mics for stage performance

• Inputs 5–6: ambient mics for audience sound



## **Display overview**

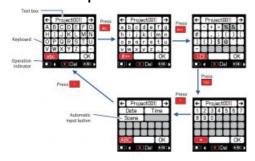
#### Home Screen



## HINT

- When the Home Screen is not open, press and hold to return to the Home Screen.
- Some of the screen will appear differently when the recording mode is Float (32 bit).

## · Character input screen



#### **NOTE**

- The following characters can be used in project names.
- (space) ! # \$ ' () + , 0 1 2 3 4 5 6 7 8 9 ; = @ A B C D E F G H I J K L M N O P Q R S T U V W X Y Z [] ^ \_ ` a b c d e f g h i j k l m n o p q r s t u v w x y z {}

## Editing operations

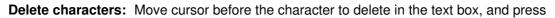


**Move cursor in text box:** Use "  $\leftarrow$ " and "  $\rightarrow$ " to move and press



Confirm characters: Move the cursor to the character to input, and press







Complete editing: Move cursor to "OK" and press



Cancel editing: Press

· Automatic input keys

(Date): This automatically inputs the date. Example: 190210 (Time): This automatically inputs the time. Example: 180950 (Scene): This automatically inputs the current scene name.

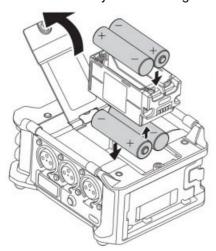
## **Preparations**

## Supplying power

Power can be supplied three ways using AA batteries, an L battery or USB.

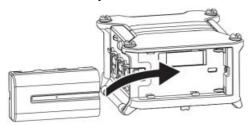
## **Using AA batteries**

- 1. Loosen the screw in the battery cover on the bottom.
- 2. Open the battery compartment cover on the bottom, remove the battery case, and insert 4 AA batteries.
- 3. Put the case into the compartment.
- 4. Close the battery cover and tighten the screw.



### · Using an L battery

1. Slide the battery in the direction of the arrow while pressing it toward the recorder.

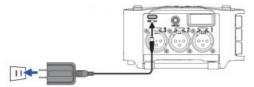


#### **NOTE**

- Be careful because the battery case could become loose unexpectedly if the battery compartment cover screw is not tightened firmly.
- Use only one type of batteries (alkaline, NiMH or lithium) at a time.
- After loading AA batteries, set "Power Source" to the correct type of battery. (→ P. 23)
- If the remaining battery power indicator becomes red, turn the power off immediately and install new batteries.

## • Using a USB Type-C cable

- 1. Connect the cable of the dedicated ZOOM AD-17 AC adapter to the USB port.
- 2. Plug the dedicated AC adapter into an outlet.

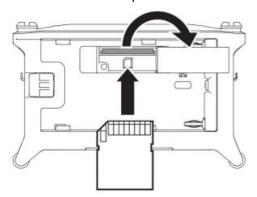


#### **NOTE**

- A 5V mobile battery (commercially-available) can also be connected.
- When connected to a computer, power can be supplied by USB.

## Loading SD cards

- 1. Open the SD card slot cover, and insert an SD card.
- 2. To remove the card: push it further into the slot and then pull it out.



#### **NOTE**

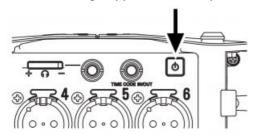
Before using SD cards that have just been purchased or that have been formatted on a computer, they must be formatted. To format an SD card, use Menu > SYSTEM > SD Card > Format.

#### Turning the power on and off

### · Turning the power on



The ZOOM logo appears and the power turns on.



#### **NOTE**

- The first time the power is turned on after purchase, the date/time must be set ( → P. 21). This setting can also be changed later.
- If "No Card!" appears on the display, confirm that an SD card is inserted properly.
- If "Card Protected!" appears on the display, the SD card write-protection is enabled. Slide the lock switch on the SD card to disable write-protection.
- If "Invalid Card!" appears on the display, the card is not formatted correctly. Format the card or use a different card. Formatting SD cards ( $\rightarrow$  P. 178)
- Turning the power off



1. Press and hold

NOTE:

Keep pressing it until the ZOOM logo appears on the LCD.

## Setting the language

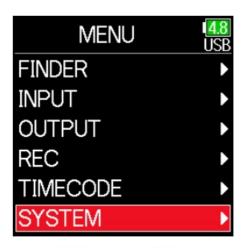




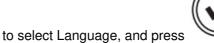
1. Press











SYSTEM USB
SD Card
USB
Bluetooth
Settings
Firmware Version
Language English





Jse and to select the desired language, and press



## **NOTE**

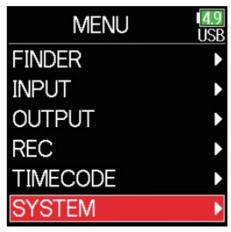
The first time the power is turned on after purchase, the language must be set.

## Setting the date and time

The date and time set on the are used when recording files, for example. The date format (order of year, month and day) can also be set.



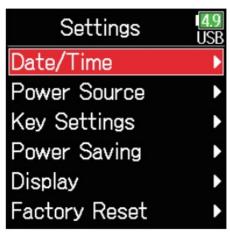




to select Settings, and press



to select Date/Time, and press

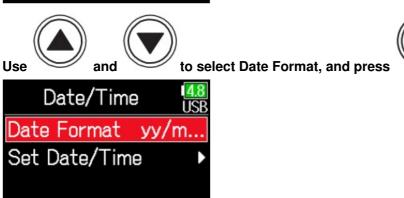


## **NOTE**

The first time the power is turned on after purchase, the date/time must be set.







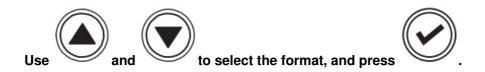
6. Set the date and time

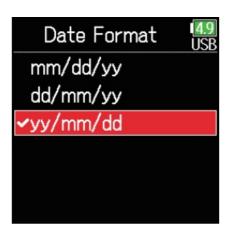


to select the item, and press

Change item value: Use



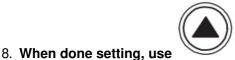




7. The item selected to be changed appears red.



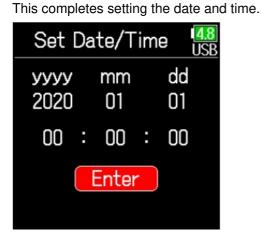






to select Enter, and press



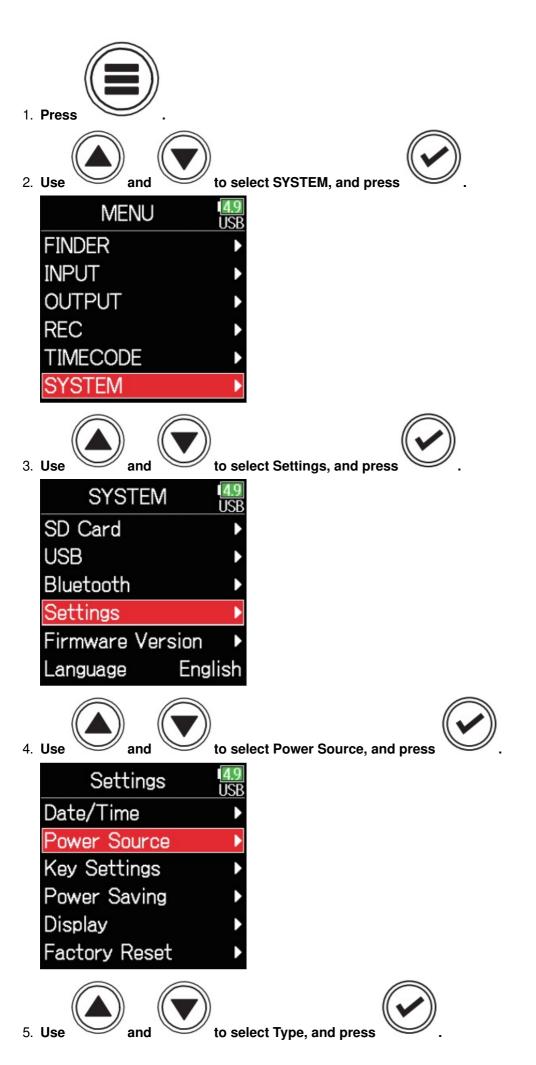


**Setting:** Explanation

mm/dd/yy: Month, day, year order dd/mm/yy: Day, month, year order yy/mm/dd: Year, month, day order

## Setting the power supply used

When using AA batteries, set the battery type so that the amount of remaining power can be shown accurately. The voltage of each power supply and the remaining battery charge can be checked on this menu page.









### **NOTE**

- When multiple power supplies are connected, they will be used in the following order of priority.
  - 1. USB (Power supply connected to USB port)
  - 2. EXT (L battery)
  - 3. AA (Installed AA batteries)
- The voltages of each power supply are shown on the display.

## Recording

## **Recording process**

Recording with the follows the process shown below.

The data created for each recording occurrence is called a "take".



**Connecting:** Connect mics, instruments, audiovisual devices, and other equipment to Inputs 1–6. ( $\rightarrow$  P. 8)

## Preparing to record:

- 1. . Set the recording mode (bit depth) (  $\rightarrow$  P. 32).
  - Select one of the recording modes: 16/24-bit WAV, 32-bit Float WAV, simultaneous 16/24-bit WAV and

### 2. Set the recording file

- Set the recording file format ( → P. 26).
- Set the sampling rate (  $\rightarrow$  P. 30).
- 3. Select tracks to record (  $\rightarrow$  P. 51).

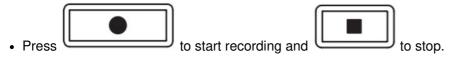


- Turn the left until it clicks to disable the input. Input is enabled at all other positions.
- This can be set to a stereo track (→ P. 99).

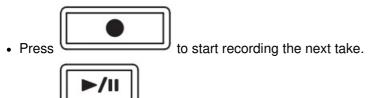
## 4. Make various input and recording settings

- Settings, including metadata (→ P. 67), pre-recording (→ P. 38), low-cut filter (→ P. 85) and limiter (→ P. 87) can be made.
- 5. Adjust input levels ( $\rightarrow$  P. 28).
  - Setting input levels is necessary in some operation modes.

## Recording (→ P.29):



• Marks (for cueing) can also be set.



to pause.

## Playing and checking ( → P. 54)

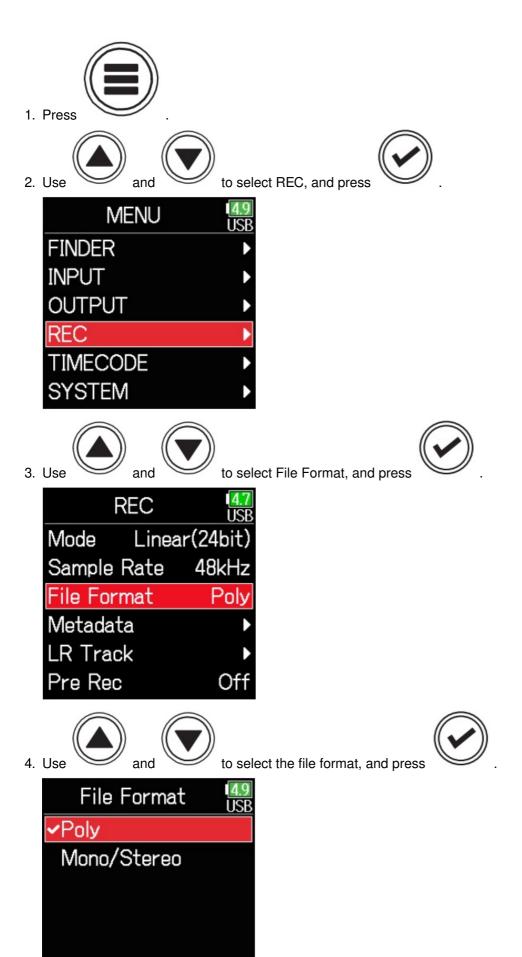


• Marks (for cueing), for example, can also be set.

## Checking take information ( $\rightarrow$ P. 67)

· Check and edit metadata.

## Setting the recording file format



## Selecting inputs and adjusting levels

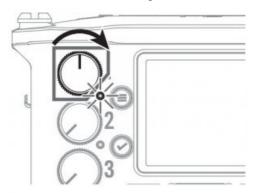
Select which among Inputs 1-6 to use.

Inputs will be recorded on tracks with the same numbers. For example, Input 1 will be recorded on track 1 and Input 2 will be recorded on track 2.

## Selecting inputs



right for the number of an input to record, making the track status indicator light.



#### **HINT**



left until it clicks to disable the input. Input is enabled at all other positions.

Track indicator Track number background color: Explanation

Lit red: **Red:** The input is enabled. Unlit: **Gray:** The input is disabled.

## **NOTE**

• The signals from the inputs selected this way will also be sent to the L/R tracks.

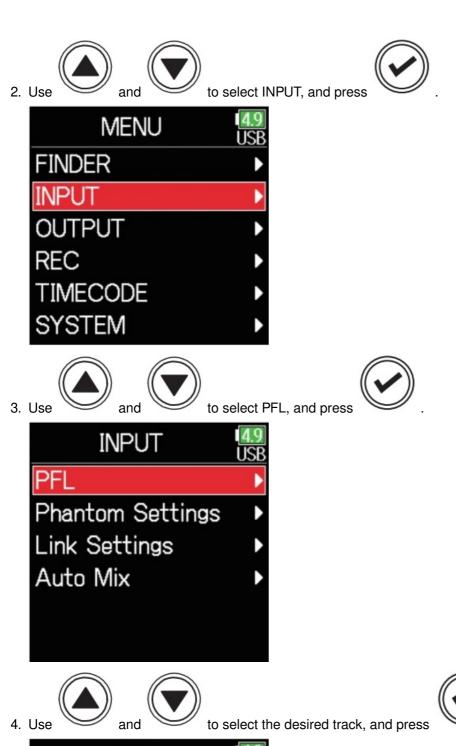


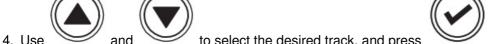
• The levels sent to the L/R tracks are adjusted with

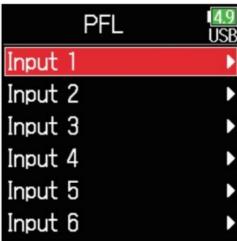
## Adjusting input levels



1. Press





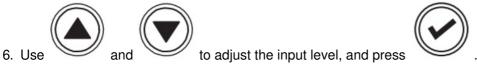






#### NOTE:

Trim cannot be used when the recording mode is set to Float. When set to Float, the setting is shown as "-".





## **HINT**

- This can be set in a range from +12 to +75 dB when the input source is set to Mic, from -8 to +55 dB when set to Line, and from -35 to +30 dB when set to USB.
- If the sound distorts even after lowering the input level, try changing mic positions and adjusting the output levels of connected devices.
- Using the limiter ( → P. 87)
- Using the high pass filter (→ P. 85)

## Recording





This starts recording.

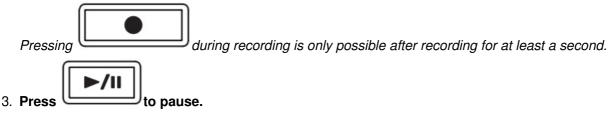
#### **HINT**

If the timecode function is enabled, recording will start from frame 00 (00 or 02 when using drop frame) and the file length will always be a full second value. This makes synchronization easy when editing later.



This will end the current take and start a new take while continuing to record without interruption.

#### NOTE



#### **NOTE**

- · Pausing occurs at whole second increments.
- When recording is paused, a mark is added at that point.
   Press to resume recording.
- A maximum of 99 marks can be added to a take.

#### HINT

- During playback, and can be pressed to jump to places where marks have been added.
- Marks can be added without pausing. (→ P. 170)



### NOTE

If the file size exceeds 2GB during recording, a new take will be created automatically and recording will continue without interruption.

No gap in sound will occur between the two takes when this happens.

#### HINT

- Press and hold when the Home Screen is open to check the name of the next take recorded.
- Files are automatically saved at regular intervals during recording. If the power is interrupted or another problem occurs during recording, an affected file can be restored to normal by playing it with the .

#### Setting the sampling rate

The sampling rate used to record files can be set.

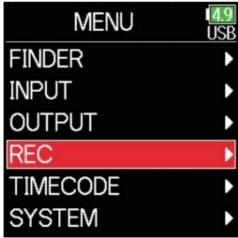


1. Press



to select REC, and press

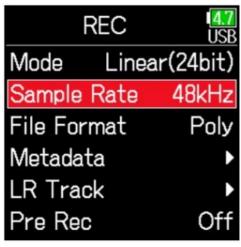




3. Use and

to select Sample Rate, and press

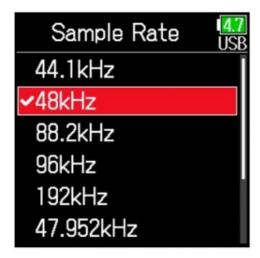






to select the sampling rate, and press





### **Setting** Explanation

44.1 kHz, 48 kHz, 88.2 kHz, 96 kHz, 192 kHz: These are standard sampling rates.

**47.952 kHz:** Select this when recording video at 23.976 frames per second in order to edit later at 24 frames per second.

**48.048 kHz:** Select this when recording video at 24 frames per second in order to edit later at NTSC 29.97 or 23.98 HD.

**47.952 kHz(F), 48.048 kHz(F):** These function the same as the two above, but the sampling rate metadata will be recorded as 48 kHz for . This enables playback and editing with devices and software that do not support 47.952 kHz and 48.048 kHz WAV files. Playback, however, will occur at the ±0.1% speed at which the file was recorded.

#### **NOTE**

- 192 kHz cannot be selected when the recording mode is Float (32bit) and the LR track is on.
- When 192 kHz is selected, Dual (16+32bit) and Dual (24+32bit) cannot be set.
- When the recording mode is MP3, only 44.1 kHz and 48 kHz can be selected.
- When 192 kHz is selected, L/R tracks will not be recorded. Input and output delay are also disabled.
- The Limiter cannot be set to On (Advanced) if Auto Mix is On or the Ambisonic format is not set to Off.
- AIF with Rec cannot be used when values other than 44.1 kHz or 48 kHz are selected.

#### Setting the recording mode (bit depth)

Set the recording mode.

The bit depth of WAV files recorded by the bit depth of WAV files recorded by the bit depth of will change according to the mode setting.

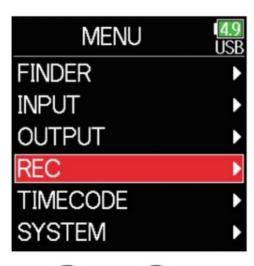




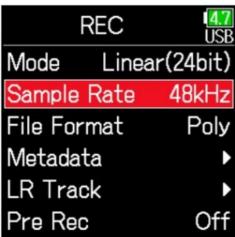


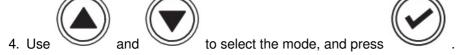


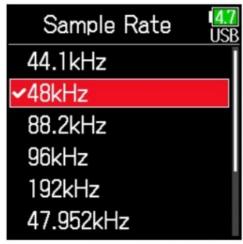
to select REC, and press











#### HINT

The setting options are Linear (16bit), Linear (24bit), Float (32bit), Dual (16+32bit), Dual (24+32bit) and MP3.

## Setting MP3 file bit rate (MP3)

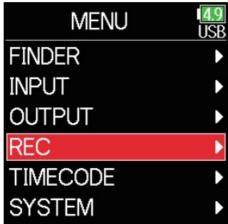
The bit rate used for recording MP3 files can be set.





to select REC, and press

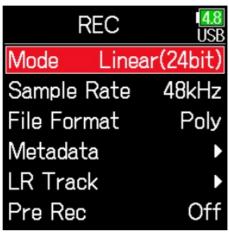






to select Mode, and press

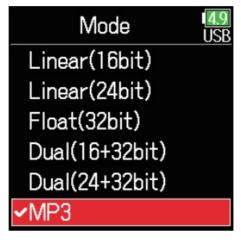






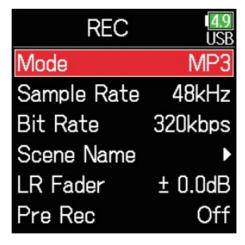
to select MP3, and press







5. Press to return to the REC screen.

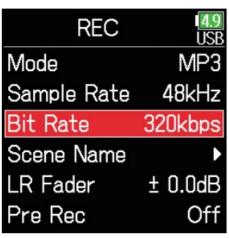




6. Confirm that the Mode is set to MP3. Then, use

nd to select Bit Rate, and pres

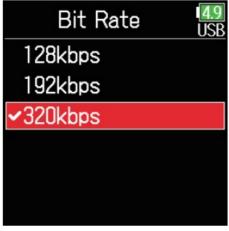






to select the bit rate, and press

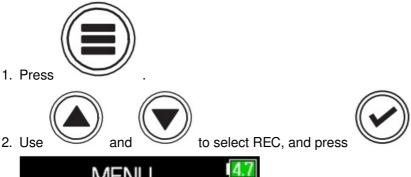


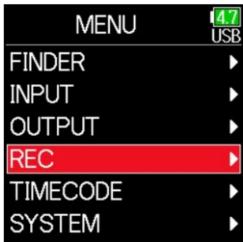


This can be set to 128 kbps, 192 kbps or 320 kbps.

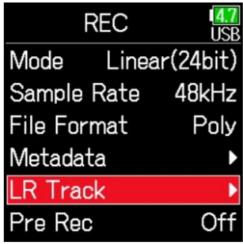
## **Setting the LR Track**

## **Enabling the LR track**





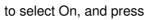
















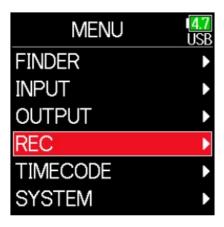
## NOTE

- Off: This disables the LR Track.
- On: This enables the LR Track. All selected tracks and the LR Track will be recorded.
- On (LR only): This enables the LR Track. Only the LR Track will be recorded.
- On cannot be selected if the sample rate is 192 kHz and the recording mode is Float (32bit).

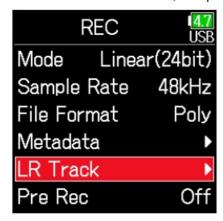
## Adjusting the L/R track volume



2. Use and to select REC, and press .



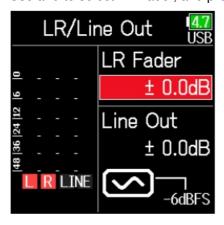
3. Use and to select LR Track, and press.



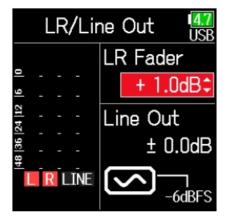
4. Use and to select LR Fader, and press .



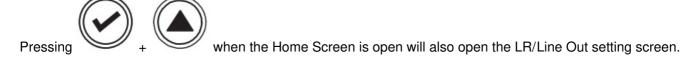
5. Use and to select LR Fader, and press .



6. Use and to change the LR fader value, adjusting the LR track volume.



## **NOTE**



## Capturing audio before recording starts

During recording, either the elapsed recording time or the remaining possible recording time can be shown.







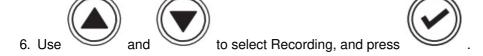


















7. Use and to select the time to show, and press



#### **NOTE**

When recording for a long time, if the file size exceeds 2 GB, recording will continue in a new file and the recording time will reset. This can be changed, however, so that it is not reset and the total recording time is shown.

Set Rec Time Reset on the Time Display screen to On/Off to set whether or not recording time resets when a new file is created.

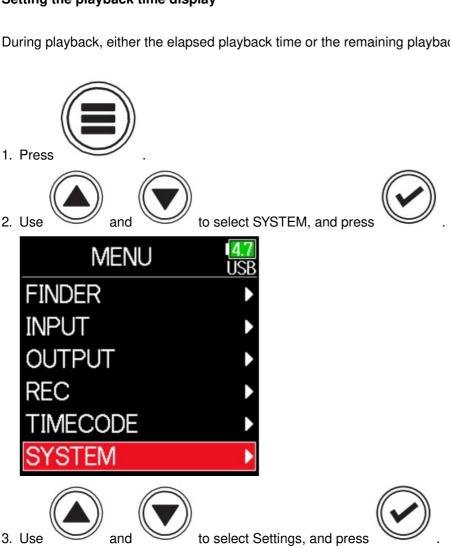


Off: When recording, even if the file size reaches 2GB, the counter shown on the Home Screen will not reset. On (reset): When recording, if the file size reaches 2GB, the counter shown on the Home Screen will be reset to 000:00:00.



## Setting the playback time display

During playback, either the elapsed playback time or the remaining playback time can be shown.

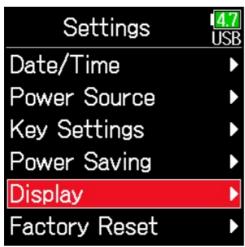






to select Display, and press







to select Time Display, and press







to select Playing, and press









Use and to select the time to show, and press



## Folder and file structure

When recording with the **F6**, folders and files are created on the SD card in the following manner.

F6 folders and files are used to manage scenes and takes as a rule.

## Folder and file structure

The folder and file structure differs according to the recording file format. In addition, the names of folders and files depend on how scenes are named.

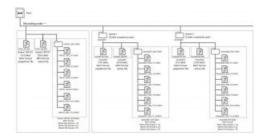
### **NOTE**

- Setting the recording file format ( → P. 26)
- Setting how scenes are named (mode) ( → P. 48)

#### **HINT**

**Take:** This is a unit of data created for a single recording.

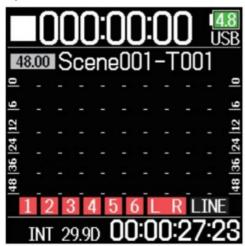
**Scene:** This is a unit containing multiple files and takes that comprise a single scene.



# Move the previously recorded take to the FALSE TAKE folder.

If the just recorded take was a failure, a shortcut can be used to move the recording to the FALSE TAKE folder.

1. Open the Home Screen.







2. While pressing

#### HINT

- Moving a take to the FALSE TAKE folder reduces the take number by one.
- Even during recording, the previously recorded take can be moved to the FALSE TAKE folder.

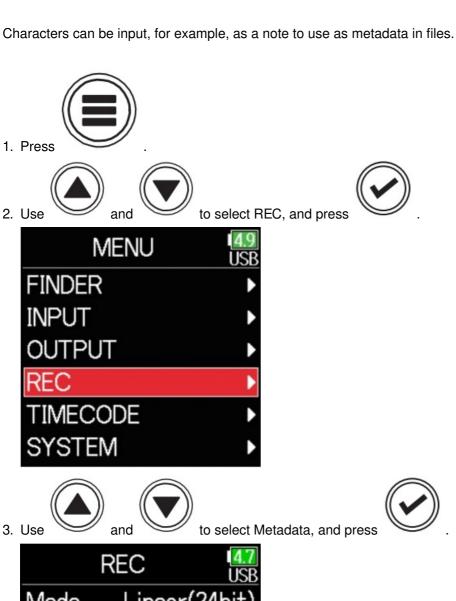


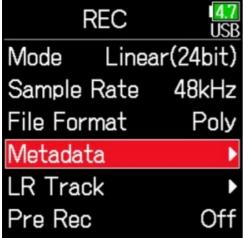
to select Execute, and press





## Changing the note for the next take recorded





## **Editing notes**











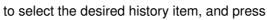


6. Edit the note.

See "Character input screen" (  $\rightarrow$  P. 14) for how to input characters.









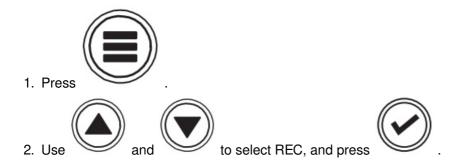


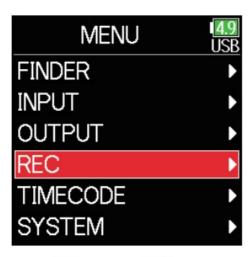
#### NOTE

The history list will be erased if the Factory Reset function is used.

## Setting and managing recorded scene names

The way scenes are named (name mode) can be set.

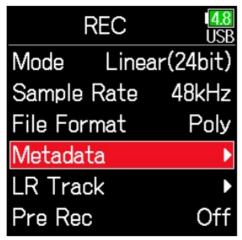






to select Metadata, and press



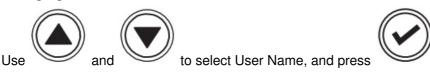








#### Changing scene names





# Selecting a scene name from the history list



to select User Name, and press.







## Changing scene names





# Selecting a scene name from the history list





6. Edit the scene name.

See "Character input screen" ( $\rightarrow$  P. 14) for how to input characters.



# Selecting a scene name from the history list



to select the desired history item, and press

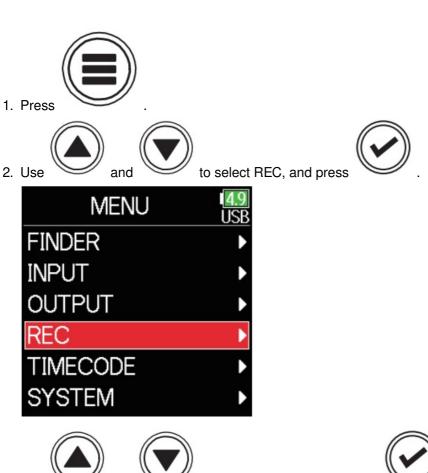


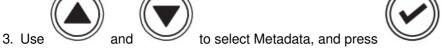
## **NOTE:**

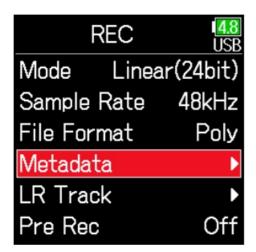
The history list will be erased if the Factory Reset function is used.

# Changing the track name of the next take recorded (Track Name)

The track name set with the following procedure will be given to the next recorded track.



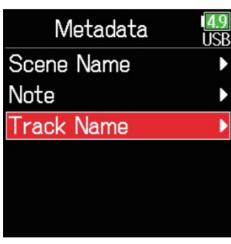






to select Track Name, and press





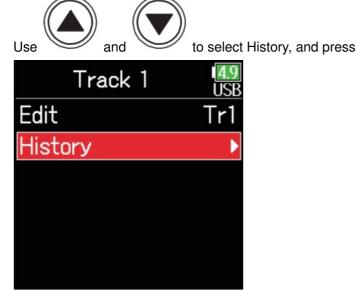
5. Use and to select a track, and press







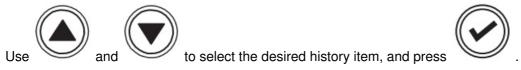
# Selecting a track name from the history list



7. Edit the track name. See "Character input screen" (  $\rightarrow$  P. 14) for how to input characters.



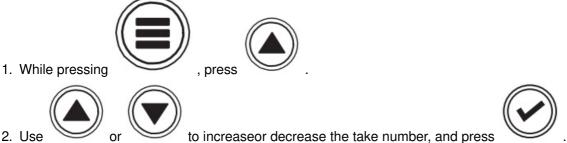
# Selecting a track name from the history list





# Changing the number of the next take recorded

The number given to the next recorded take can be changed when the Home Screen is open.





# **Playback**

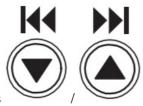
2. Use

# **Playing recordings**

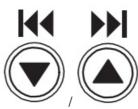




## ■ Playback operations



Select take/Jump to mark: Press



Search backward/forward: Press and hold



Pause/resume playback: Press

## **NOTE:**

Track backgrounds will appear black.



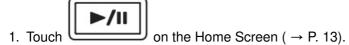
## HINT



- The longer / is pressed and held, the faster the speed of searching backward/forward.
- An "Invalid Take!" message will appear if the selected take is not valid.
- A "No Take!" message will appear if no playable take exists.
- During playback, press to add marks that can be used for skipping.
   ( → P. 170)
- 2. Press to return to the Home Screen.

The volume and panning of each track during playback can be changed.

# **Setting faders**





to adjust the input signal level. 2. Turn





left until it clicks to mute the input.

# Setting the panning











to selectPFL, and press



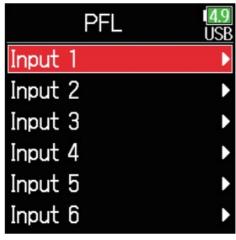


4.



to select the desired track, and press











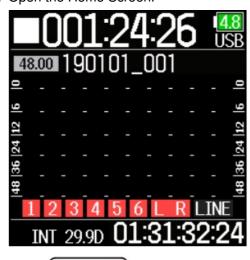
7. Adjust the panning.



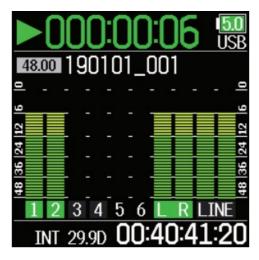
# Monitoring the playback signals of specific tracks during playback

The playback signals of specific tracks can be monitored using SOLO mode.

1. Open the Home Screen.



2. Press to start playback.



3. Press during playback.

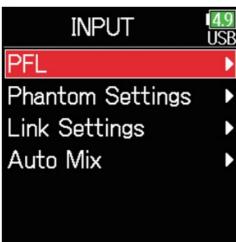


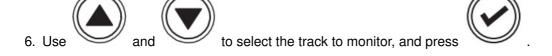


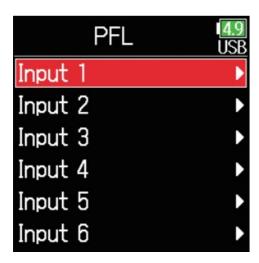
NOTE:

SOLO mode can only be used with tracks that can be played back (indicators lit green).

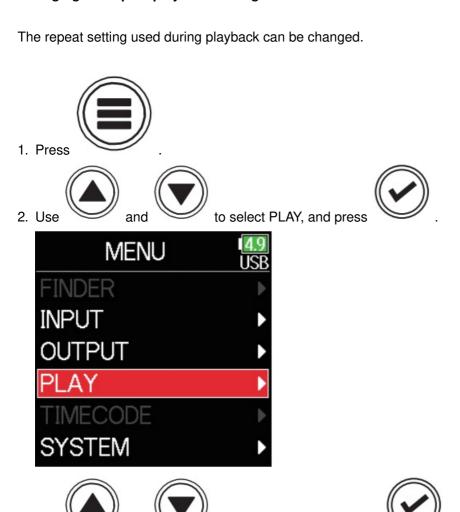








# Changing the repeat playback setting



to select Repeat, and press







I. Use and to select the repeat mode, and press



**Setting:** Explanation

Play One (single playback): Only the selected take will be played.

Play All (all playback): Takes will be played back continuously from the selected one until the last one.

Repeat One (single repeat playback): The selected take will be played repeatedly.

Repeat All (all repeat playback): All takes in the selected folder will be played repeatedly.

#### HINT:

The PLAY menu only appears during playback.

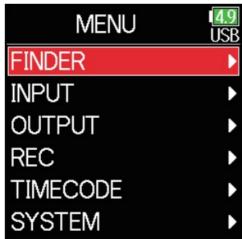
#### Take and folder operations

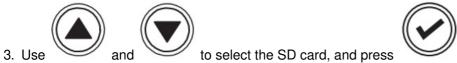
## Working with takes and folders

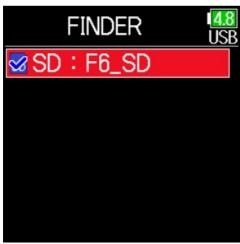
The Finder allows the viewing of the contents of SD cards, takes and folders and the creation of project/scene folders. It also allows the setting and deletion of recording/playback folders along with viewing their information, for example.











4. Use and to select New Folder, and press .

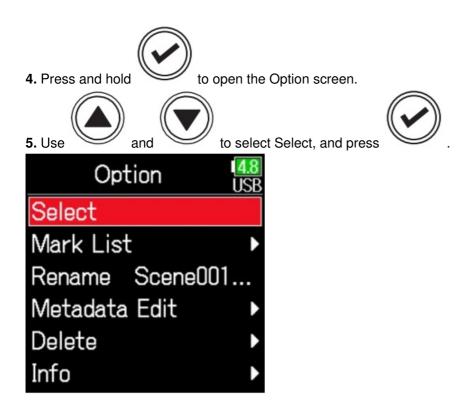


5. Edit the folder name. See "Character input screen" (  $\rightarrow$  P. 14) for how to input characters.



## Selecting the take recording/playback folder

Use this procedure to select the folder that contains the take to be played back or the folder to use for recording takes and return to the Home Screen.



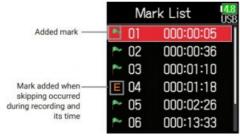
## Checking take marks and using them for playback

A list of the marks in a recorded take can be shown.









## Changing folder and take names

- 4. Press and hold to open the Option screen.
- 5. Use and to select Rename, and press .



**6.** Edit the folder/take name.

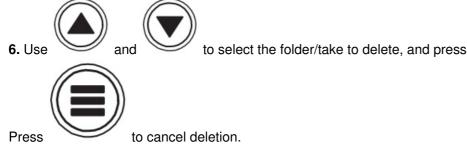
See "Character input screen" (  $\rightarrow$  P. 14) for how to input characters.



## **NOTE**

- The edited name of the folder/take is written to the or metadata.
- Spaces and @ marks cannot be input at name beginnings.







## **NOTE**

Execute

• Deleted folders and takes are not immediately erased from the SD card. They are moved to the TRASH folder.

Cancel

• Deleting folders and takes in the TRASH folder will completely erase their data.

## Checking folder and take information

- 4. Press and hold to open the Option screen.
- 5. Use and to select Info, and press .



## **■ SD card selected**

Free: Open space Size: Card capacity

Remain: Remaining recording time



#### **■** Folder selected

Date: Date Time: Time



# **■** Take selected

TC: Timecode

FPS: Timecode frame rate Len: Take recording length Fmt: Take sample format

Date: Date Time: Time Size: Take size



## **■** Emptying the TRASH/FALSE TAKE folders

4. Use and to select TRASH or FALSE TAKE.



## **TRASH folder**



#### **FALSE TAKE folder**

- 5. Press and hold .
- 6. Use and to select Empty, and press .



7. Use and to select Execute, and press.



#### Overview of metadata (take information) stored in files

The writes a variety of information (metadata) to files during recording.

When these files are read by an application that supports metadata, the saved information can be checked and used.

## WAV file metadata

The metadata saved in files recorded by the him WAV format is collected in BEXT (Broadcast Audio Extension) and iXML chunks.

For details about the metadata saved in these chunks, see "Metadata contained in BEXT chunks in WAV files" ( $\rightarrow$  P. 188), "Metadata contained in iXML chunks in WAV files" ( $\rightarrow$  P. 189).

#### MP3 file metadata

The metadata saved in files recorded by the  $\square$  in MP3 format is written as ID3v1 tags. For information about the ID3 fields and formats saved as metadata, see "Metadata and ID3 fields contained in MP3 files" ( $\rightarrow$  P. 191).

#### HINT

. F6
MP3 files conform to the MPEG-1 Layer III standard.

• MP3 metadata cannot be edited.

## Checking and editing take metadata

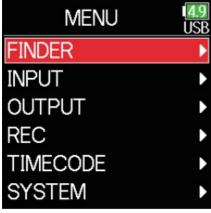


1. Press



to select FINDER, and press









3. Use and to select an SD card, and press

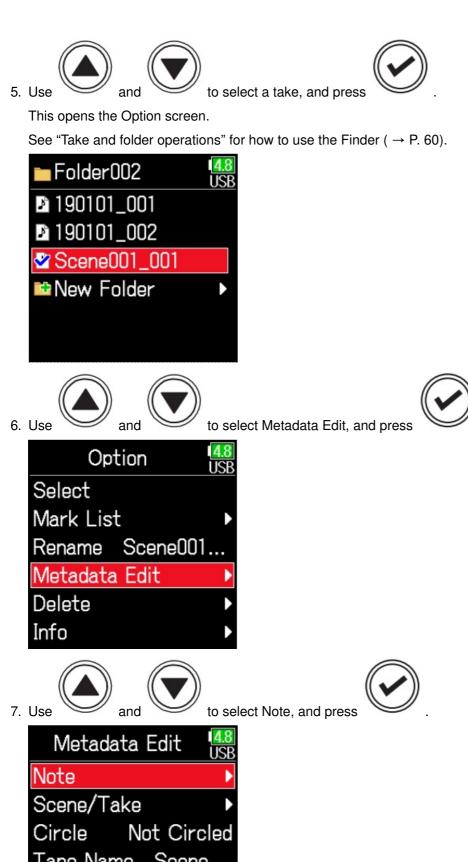




to select a folder, and press









to select Edit, and press



9. Edit the note. See "Character input screen" (  $\rightarrow$  P. 14) for how to input characters.



## Selecting notes from the history list

7. Use and to select Note, and press .



8. Use and to select History, and press .



9. Use and to select the desired history item, and press .



## Checking and editing scene names

7. Use and to select Scene/Take, and press .



8. Use and to select Scene, and press.



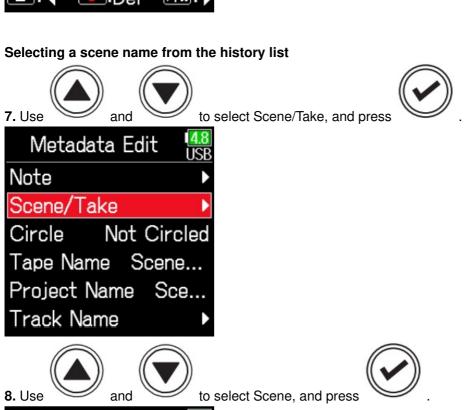
9. Use and to select Edit, and press.



#### 10.Edit the scene name.

See "Character input screen" (  $\rightarrow$  P. 14) for how to input characters.











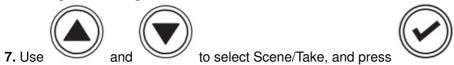


to select the History item to use, and press





## Checking and editing take numbers









9. Change the take number.



## **Editing operations**



Move cursor or change value: Press

Select parameter to change: Press *HINT This can be set from 1 to 999.* 

**NOTE** The take number is written to the metadata.





# **Circling takes**

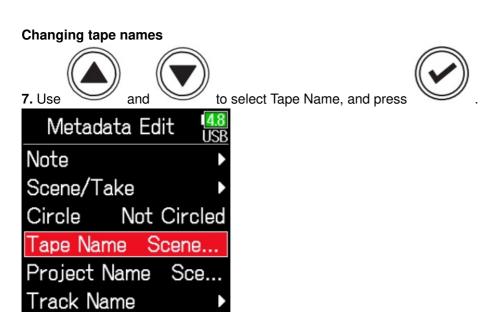
An @ mark can be added to the beginning of the name of the best take to make it stand out. This is called a "circled take".











8. Edit the folder (tape) name.

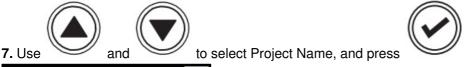
See "Character input screen" ( $\rightarrow$  P. 14) for how to input characters.



#### NOTE

- The folder (tape) name is written to the metadata.
- The folder (tape) name used immediately after recording is the name of the folder in which the take was recorded.

## **Changing project names**





8. Edit the project name.

See "Character input screen" (  $\rightarrow$  P. 14) for how to input characters.



## Checking and editing the track names

7. Use and to select Track Name, and press.



8. Use and to select a track, and press .



9. Use and to select Edit, and press .



10. Edit the track name.

See "Character input screen" ( $\rightarrow$  P. 14) for how to input characters.

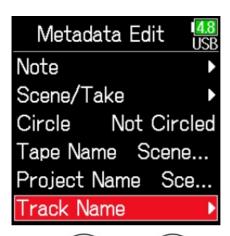


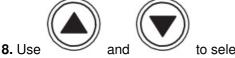
Selecting a track name from the history list











to select a track, and press



 Track Name
 4.8 USB

 Track 1
 Tr1

 Track 2
 Tr2

 Track 3
 Tr3

 Track 4
 Tr4

 Track 5
 Tr5

 Track 6
 Tr6

9. Use and

to select History, and press







to select the desired history, and press

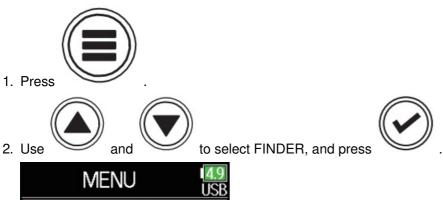


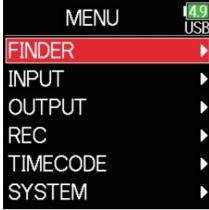
#### NOTE

The history list will be erased if the Factory Reset function is used.

# Writing a sound report

A sound report includes information about recording times and takes. Reports can be written as CSV format files (F6\_[folder name].CSV). Comments written in sound reports can also be edited.





3. Use and to select the folder or SD card desired for sound report creation, and press and







to select Sound Report, and press





## Writing sound reports

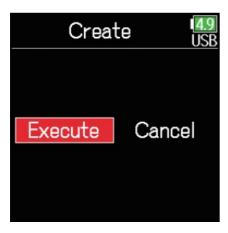




6. Use and to select Execute, and press selected SD card or folder.



. This writes the sound report inside the



# **Editing comments**









9. Edit the comment. See "Character input screen" for how to input characters.



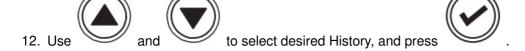
# Selecting comments from the history list











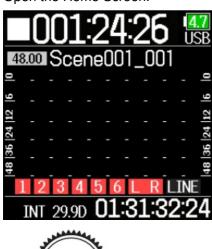


## Input settings

#### Adjusting the input signal monitoring balance

The volume of each track can be adjusted when monitoring input signals

1. Open the Home Screen.





**HINT:** The fader setting range depends on the recording mode. In Float mode, it is muted and –60.0 to +60.0 dB. In Linear mode, it is muted and –48.0 to +24.0 dB.

#### **NOTE**

- Mix settings are saved separately for each recorded take and can be changed during playback.
- Mix settings are not saved with the take when the recorded file format is MP3.

## Monitoring the input signals of specified tracks

The input signals of specified tracks can be monitored. Even tracks that have not been set to record can be input to the PFL screen and their input sounds monitored. This is convenient when using tracks as return inputs. Carious settings can be made for selected tracks.

1. Press when the Home Screen is open.

The PFL screen for the track that was last opened opens, and the status indicator lights orange. Only the input sound of the track show can be monitored through headphones.





This opens the Home Screen.

Parameter	Explanation
Source	This sets the input source.
Trim	This sets the input level.
HPF/Limiter	This sets the high pass filter and limiter.
Phase/Delay	This sets the phase reversal and delay.
Pan	This sets the panning.
Monitor	This sets the monitoring volume on the PFL screen.

### **HINT**



to select parameters and change setting values.

• When the cursor is on the topmost track number, press next track.

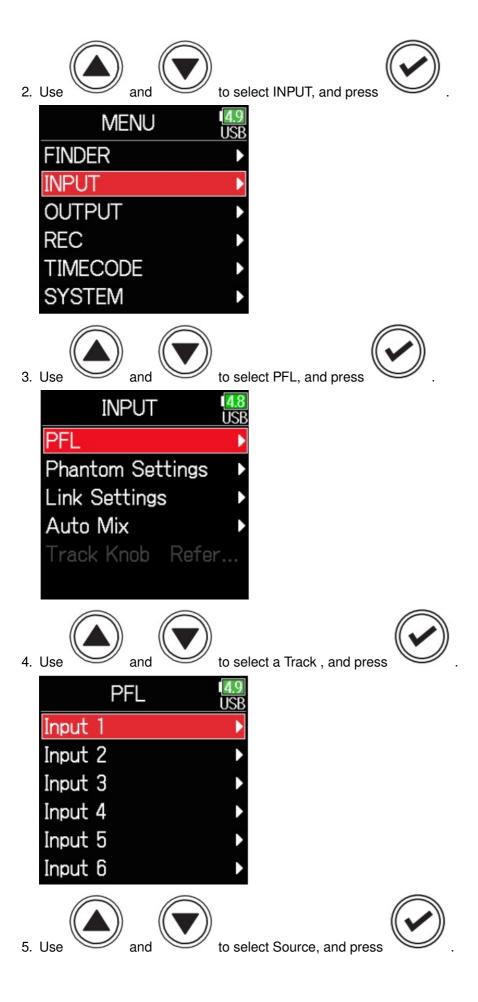


## Setting the input source

The input source and phantom power on/off status can be set for each track.



1. Press







to select Input Source, and press





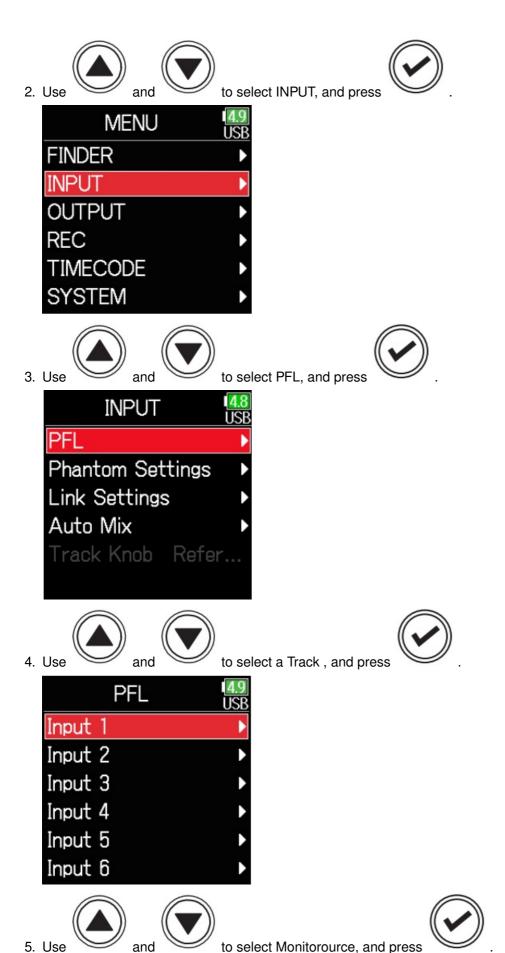
Setting	Explanation
Mic	Use when connecting a Mic or other equipment with a low input level.
Mic (PH	Use for Mic level with phantom power.
Line	Use when connecting line level equipment. The input level will be reduced 20 dB compared to when Mic is selected.
Line (P H)	Use this setting for line level with phantom power.
USB 1–	When AIF with Rec is set to On, computer output signals are treated as input signals

# Setting the monitoring volume on the PFL screen

On the PFL screen, the monitoring sound can be set to be either pre-fader listening (PFL) or fader solo (SOLO).



1. Press







to select the mode, and press

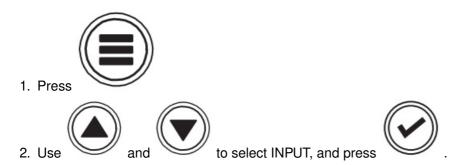




Setting	Explanation
PFL	On the PFL screen, monitor the pre-fader sound.
SOLO	On the PFL screen, monitor the post-fader sound.

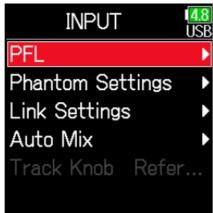
# **Cutting low-frequency noise**

The high pass filter can cut low frequencies to reduce the sound of wind, vocal pops and other noise.

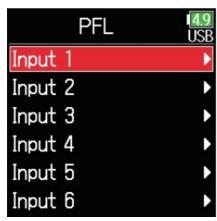


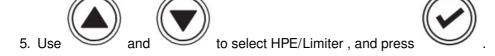






4. Use and to select a Track, and press











7. Use and to select the desired cutoff frequency, and press

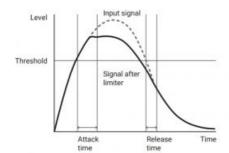




HINT: This can be set to Off or between 10 and 240 Hz.

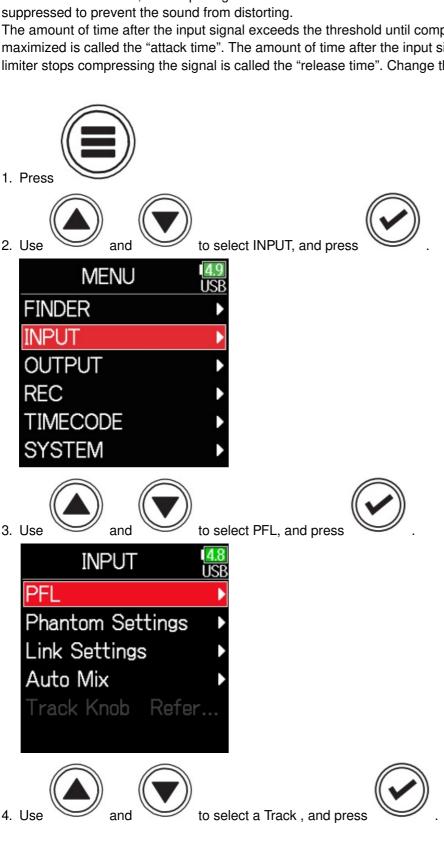
# **Input limiter**

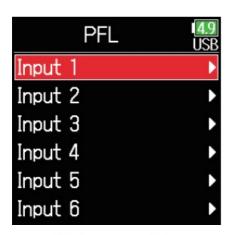
The limiter can prevent distortion by reducing input signals that have excessively high levels.

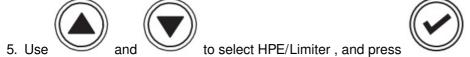


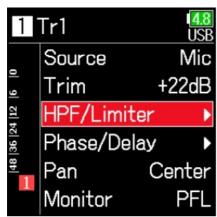
When the limiter is ON, if the input signal level exceeds the set threshold value, the input signal level will be suppressed to prevent the sound from distorting.

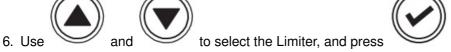
The amount of time after the input signal exceeds the threshold until compression of the output signal is maximized is called the "attack time". The amount of time after the input signal goes below the threshold until the limiter stops compressing the signal is called the "release time". Change these two to adjust the audio quality











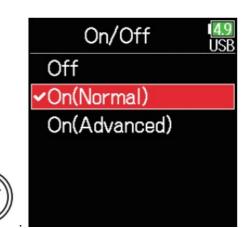


Using the limiter



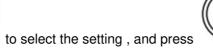


to select ON/OFF, and press





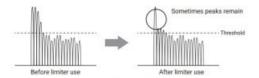




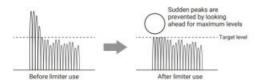
# Continue to one of the following procedures.

Using the limiter	P. 88
Setting the type	P. 90
Setting the threshold	P. 90
Setting the attack time	P. 91
Setting the release time	P. 91
Setting the target level	P. 92

# On (Normal)



# On (Normal)

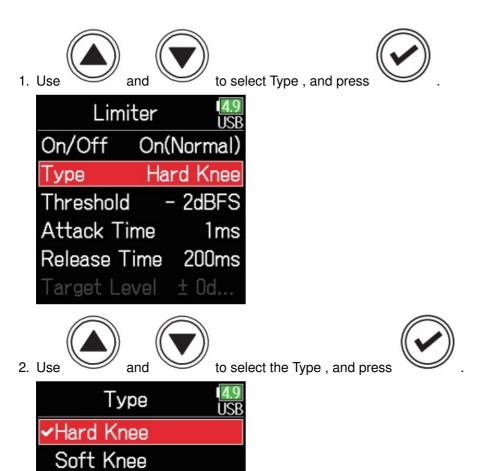


Settin g	Explanation
Off	This disables the limiter.
On (No rmal)	This applies an ordinary limiter. The ratio is 20:1.
On (Ad vanced )	By detecting the maximum level in advance, this optimized limiter prevents distortion even more than ordinary limiter operation. The ratio is ∞:1, providing increased internal headroom.

# **NOTE**

- When set to On (Advanced), the input latency of the increases 1 ms. When monitoring sounds being recorded with a mic in real-time, increased latency can cause interference between the sound being recorded that is transmitted through the air and the delayed monitored sound, possibly making accurate monitoring difficult.
- When set to On (Advanced), the Sample Rate cannot be set to 192 kHz.
- Moreover, when the Sample Rate is set to 192 kHz, the On (Advanced) setting cannot be selected.

#### Setting the type



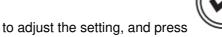
#### Setting the threshold

This sets the base level from which the limiter operates









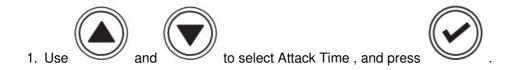


Setting	Explanation	
Hard Knee	Only peaks that exceed the threshold are attenuated. There is no effect below the threshold.	
Soft Knee	The limiter gradually affects the signal about 6 dB below the threshold for a gentler effect	

HINT: This can be set from -16 to -2 dBFS

#### Setting the attack time

This sets the amount of time until compression starts after the input signal exceeds the threshold.







to adjust the time, and press



Limiter USB
On/Off On(Normal)
Type Hard Knee
Threshold - 2dBFS
Attack Time Imst
Release Time 200ms
Target Level ± 0d...

HINT: This can be set from 1 to 4 ms

#### Setting the release time

This sets the amount of time until compression stops after the input signal goes below the threshold.



to select Release Time , and press







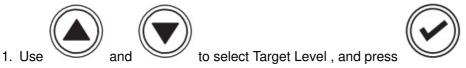
to adjust the time, and press



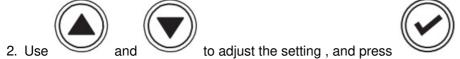
**HINT:** Limiter operation is linked for tracks that have stereo link or MS stereo link enabled. If the signal for either linked channel reaches the threshold, the limiter will operate on both tracks.

#### Setting the target level

When the limiter On/Off setting is set to On (Advanced), use this to set the target output level for the signal.







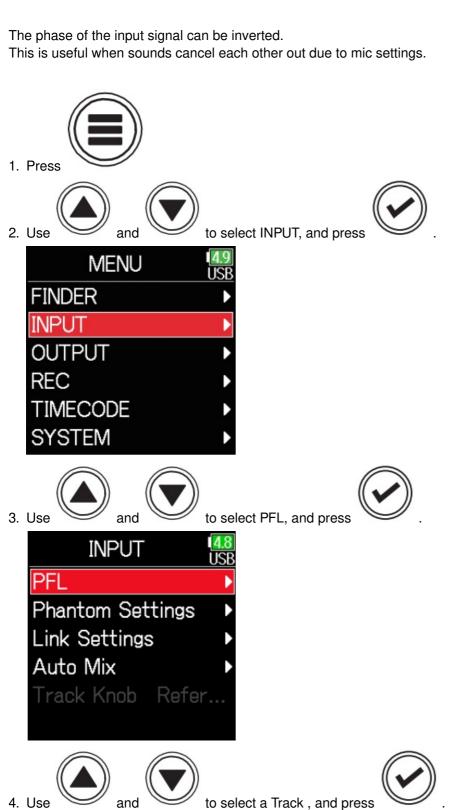


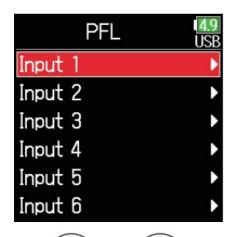
#### HINT:

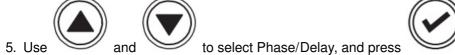
• This can be set from -16 to 0 dBFS.

 After a signal passes through the limiter, it will not exceed the set target level value.

## Inverting the input phase









6. Use and to select Phase Invert, and press





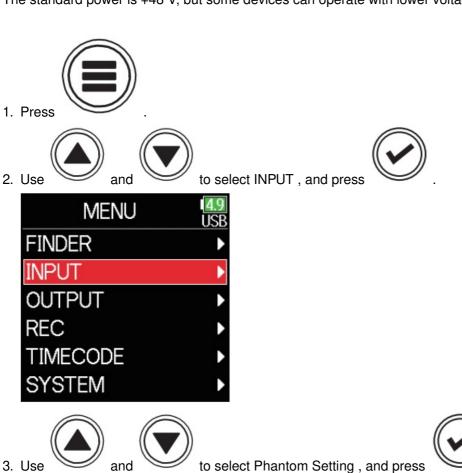


#### Changing the phantom power settings

The can provide phantom power. The voltage can be set to +24V or +48 V and it can be turned on/off for each input separately.

**HINT:** Phantom power is a function that supplies power to devices that require an external power supply, including some condenser mics.

The standard power is +48 V, but some devices can operate with lower voltages.

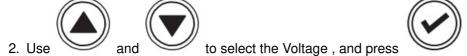


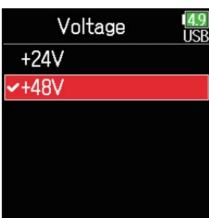


# Setting the voltage

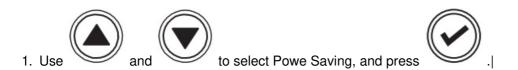


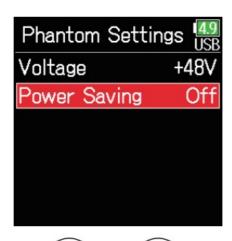






## Disabling phantom power during playback

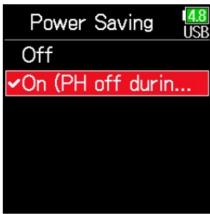












HINT: When using mics and other equipment that can operate with voltages

less than +48 V, selecting the lower voltage can reduce the power consumption

Setting	Explanation
Off	Phantom power is supplied even during playback.
On (PH off during playbac k)	Phantom power is not supplied during playback. This can reduce the power consumption.

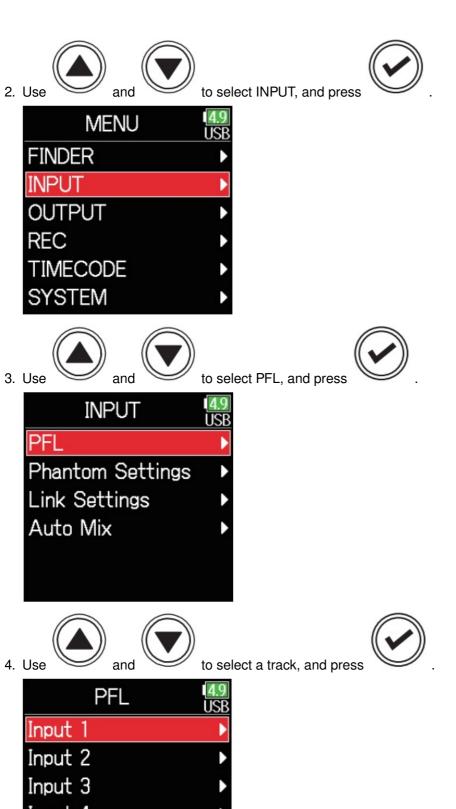
HINT: If mics do not need phantom power during playback, disabling it can

reduce **F6** power consumption.

#### Applying delay to input signals

If there are differences in the timing of input sounds, use this function to correct them when recording

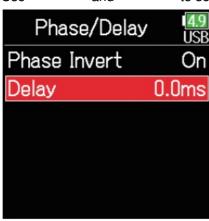


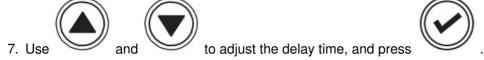




i. Use and to select Phase?Delay, and press









**HINT:** This can be set from 0 to 30.0 ms.

NOTE: When Sample Rate is set to 192 kHz, Delay is disabled.

## Linking inputs as a stereo pair

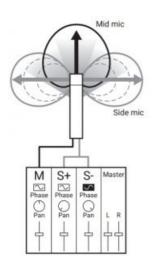
By enabling the stereo link for tracks 1/2, 3/4 or 5/6, the corresponding Inputs (1/2, 3/4 or 5/6) can be handled as a stereo pair. When linked, Input 1, 3 or 5 will be the left channel and Input 2, 4 or 6 will be the right channel.

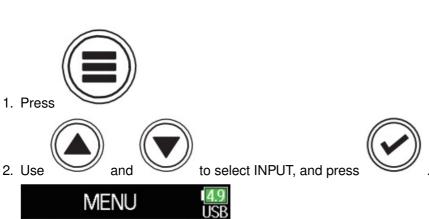
#### MS stereo format overview

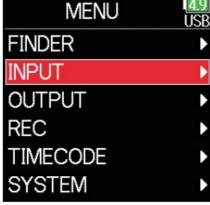
This method takes input from a directional mid mic, which captures sound in the

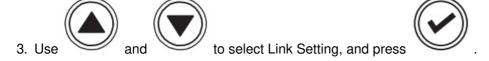
center, and a bidirectional side mic, which captures sounds from the left and right, and converts it to stereo.

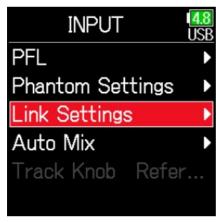
The stereo width can be changed as desired by adjusting the side mic level. Since this method can capture a wide stereo image, it is ideal for recording large open spaces with numerous sound sources, including orchestras, live concerts and soundscapes. This technique is also extremely effective when you want to adjust room ambience. Since it offers a high degree of freedom, it is used not only in studios but also for a wide range of recording, even for rehearsals and live performances.











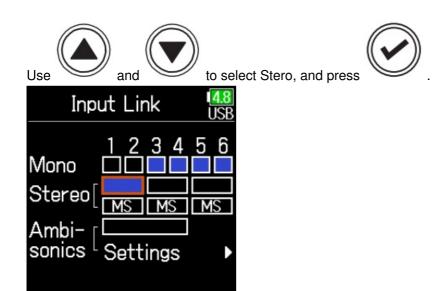


to select Input Link, and press



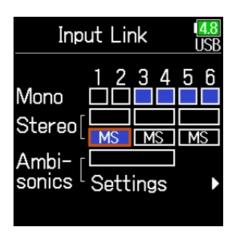


## Setting stereo links Stereo



MS





Setting	Explanation	
Stereo	When stereo-linked, inputs are handled normally.	
MS	When stereo-linked, signals from mid-side mics are converted to ordinary stereo	

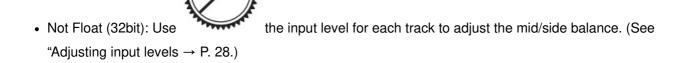
#### NOTE:

- When stereo-linked, odd tracks are handled as left and even tracks as right channels.
- When MS stereo-linked, odd tracks are handled as mid signals and even tracks as side signals.

#### **HINT**

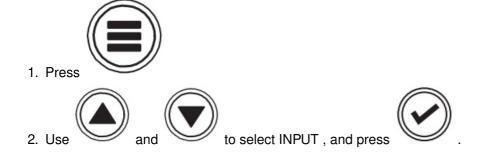
When MS stereo-linked, the method to balance mid and side is according to the recording mode as follows.

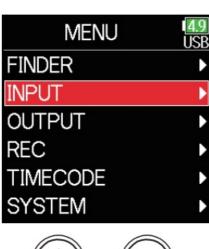
• Float (32bit): Use for each track to adjust the mid/side balance.



#### Adjusting multiple track input levels together

The input levels of multiple tracks can be linked and adjusted at the same time.



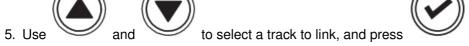


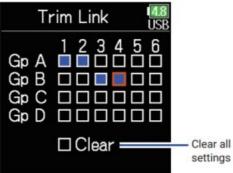


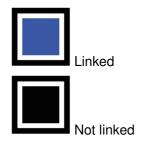








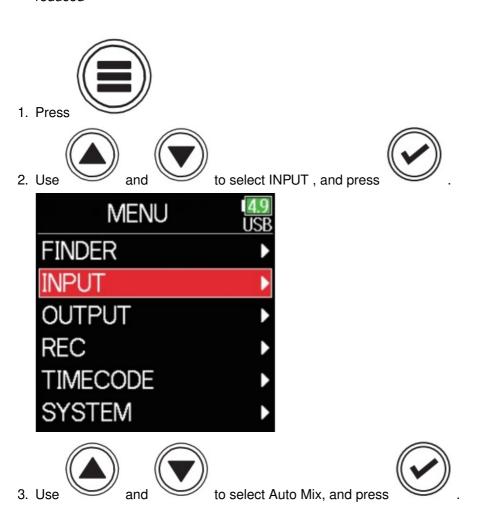




#### Changing the automatic mixing setting

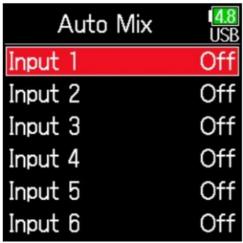
When using multiple mics to capture audio during a meeting, for example, automatically attenuating the inputs of mics that are not in active use provides the following benefits.

- The likelihood of feedback is reduced.
- Background noise, including fans and crowds, is suppressed to a certain level regardless of the number of people.
- Sound quality degradation due to phase differences caused by variations in the distances of multiple mics is reduced













#### **NOTE**

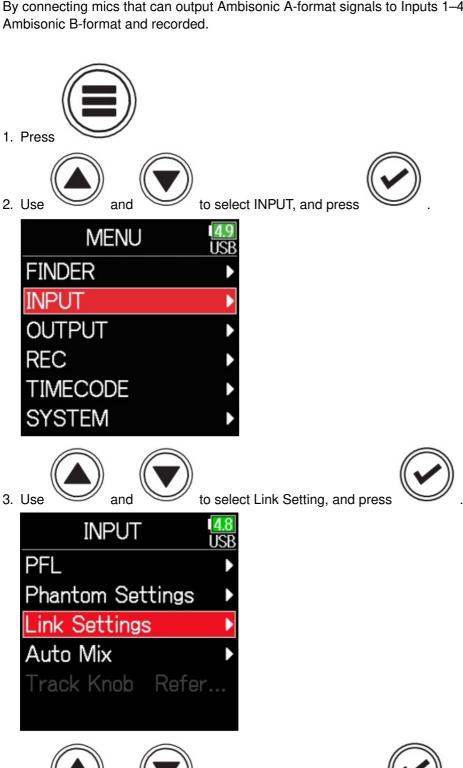
- The following functions and settings cannot be used with this function.
  - The sampling rate cannot be set to 192 kHz.
  - The Ambisonic format cannot be set to any value other than Off.
- · When monitoring sounds being recorded with a mic in real-time, increased latency can cause interference

between the sound being recorded that is transmitted through the air and the delayed monitored sound, possibly

making accurate monitoring difficult.

#### **Setting the Ambisonic format**

By connecting mics that can output Ambisonic A-format signals to Inputs 1-4, audio can be converted to



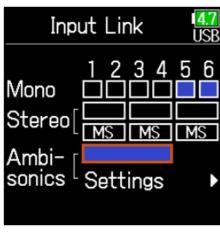
to select Input Link, and press



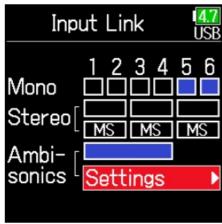


to Move the cursor to Ambisonics, and press









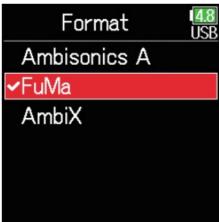








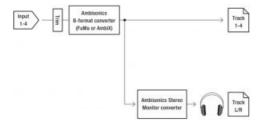




**FuMa:** This converts the signals from Inputs 1–4 to the Ambisonic FuMa B-format, and saves them as a 4-channel polyphonic file.

**AmbiX:** This converts the signals from Inputs 1–4 to the Ambisonic AmbiX B-format, and saves them as a 4-channel polyphonic file.

Format: FuMa, AmbiX



#### **Ambisonics A**

This saves the signals from Inputs 1–4 as a 4-channel polyphonic file without converting them to an Ambisonic B-format. The monitoring signal is converted to Ambisonic B-format and then to an ordinary stereo signal.

Format: Ambisonics A



#### **NOTE**

- The sampling rate can only be set to 192 kHz when Ambisonic Mode is Off. Ambisonic files are saved as 4-channel polyphonic files, not as mono or stereo files.
- The following parameters cannot be set for tracks using Ambisonic Mode input.
  - Phase Invert
  - Delay
  - Pan
  - Input Link
  - Trim Link
- Files recorded when Ambisonic format is not off will play back as Ambisonic audio sources rather than ordinary 4-channel polyphonic files. For this reason, these tracks cannot the panned or muted during playback
- This cannot be used with the Auto Mix function.

#### **HINT**

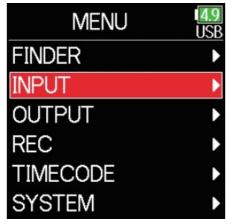
- Ambisonic can also be set during use as an audio interface (Multi Track).
- Even when Ambisonic format is not Off, PFL buttons can be selected to
  monitor their track input sounds. When Monitor is set to PFL, sounds can
  be monitored before they are converted to Ambisonic B-format. When
  PFL mode is set to SOLO, sounds can be monitored after they are converted to Ambisonic B-format.
- The following parameters that can be set on the PFL screen are linked for Ambisonic input tracks.
  - Source
  - Trim
  - HPF
  - Limiter
  - Phantom
  - Fader
  - PFL Monitor

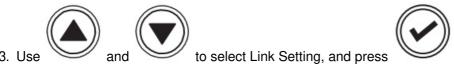
#### Setting the mic position used for Ambisonic recording

By setting the mic orientation used during Ambisonic recording as an parameter, proper positioning can be maintained when converting to Ambisonic B format if the mic orientation is changed from upright, upside down or horizontal.

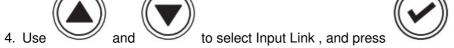


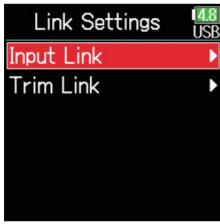


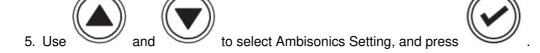


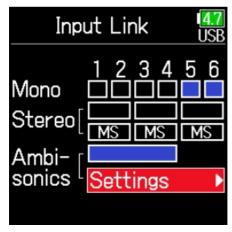














to select Mic Position, and press



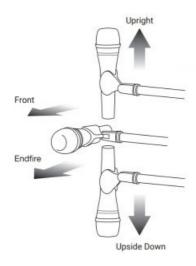


7. Use and

to select the Mic Orientation, and press







Setting	Explanation
Upright	Use this setting to record with the Mic upright.
Upside	Down Use this setting to record with the Mic upside down.
Endfire	Use this setting to record with the Mic oriented horizontally

#### **HINT**

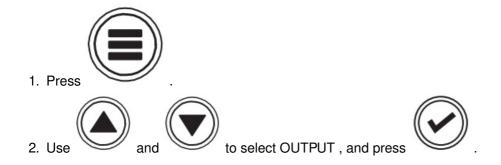
- Using the mic upright is recommended for Ambisonic recording in order to minimize reflections from the floor and the mic itself.
- When it is difficult to use the mic in an upright orientation, you can place it upside down or pointing forward and change the Mic Position setting accordingly.

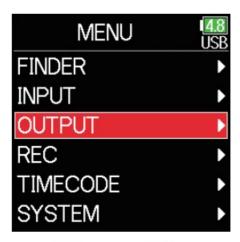
**NOTE:** If this setting and the mic position do not match, sound positioning will not be properly re-created during conversion to Ambisonic B format.

## **Output settings**

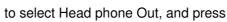
#### Setting signals sent to the headphone output

Signals sent to the headphone output can be set to either prefader or postfader for each track. Saving 10 setting combinations it is possible.





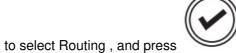


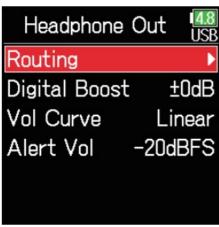


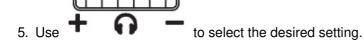




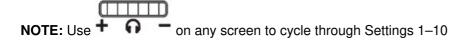






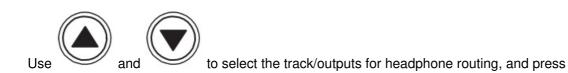




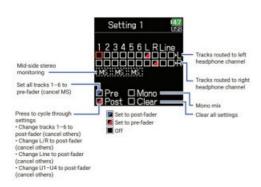


Continue to one of the following procedures.	
Setting the routing	P. 110
Using mono headphone output	P. 110
Monitoring mid-side stereo signals	P. 111

#### Setting the routing







**HINT:** Press ENTER to cycle through the options: prefader  $\rightarrow$  postfader  $\rightarrow$  off

## Using mono headphone output

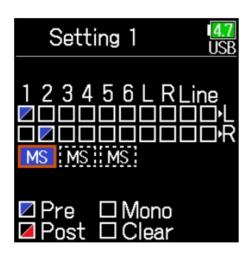




# Monitoring mid-side stereo signals

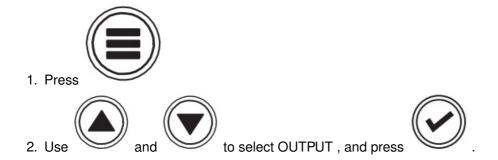
Signals from a mid-side stereo mic can be converted to an ordinary stereo signal for monitoring.

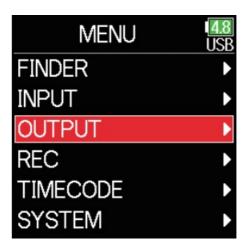




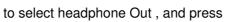
# Outputting alerts through headphones

The volume can be adjusted for alerts output from headphones when, for example, recording starts and stops.







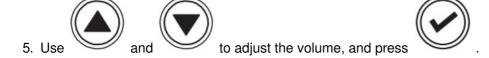










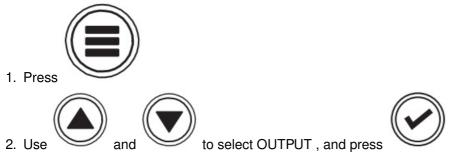


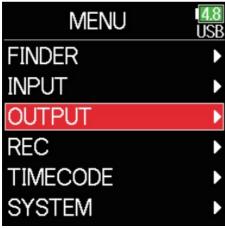


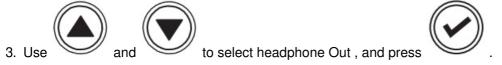
When alerts sound	Sound type
Remaining battery low	880Hz tone 4 times every 30 seconds
Recording starts	1000Hz tone 1 time
Recording stops	880Hz tone 2 times
Recording not possible	880Hz tone 3 times

## Setting the headphone output volume curve

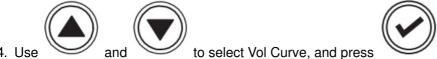
The volume curve used when adjusting the headphone volume knob can be set.

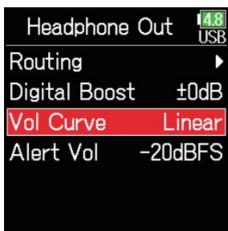




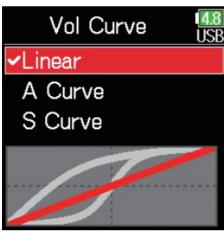






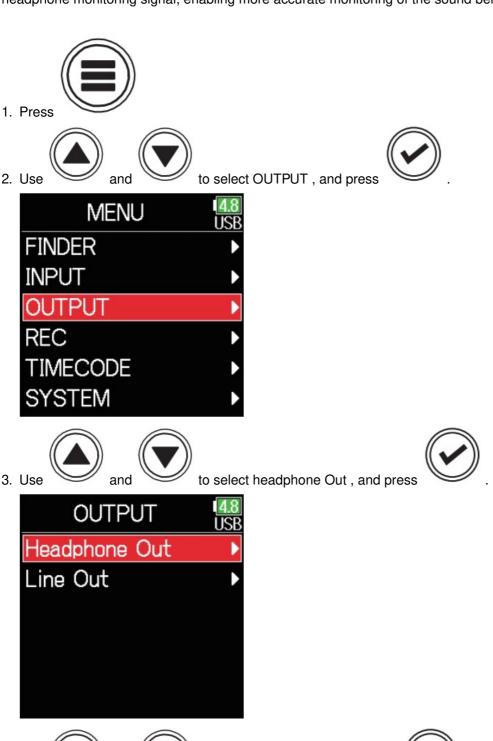




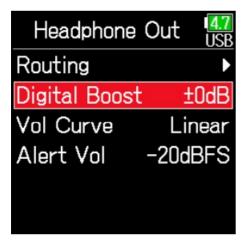


Setting	Explanation
Linear	The volume will change evenly from the minimum value to the maximum value.
A Curve	The closer the volume is to its minimum position, the more rapidly it will change.
S Curve	The closer the volume is to its middle position, the more rapidly it will change

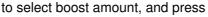
Boosting the headphone output alleviates the interference of sound waves traveling through the air with the headphone monitoring signal, enabling more accurate monitoring of the sound being recorded.



to select Digital Boost, and press









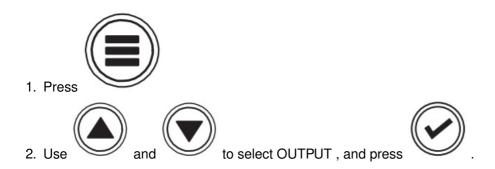


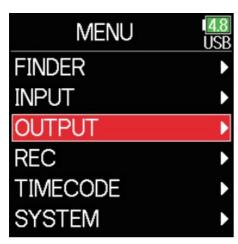
**HINT:** The amount of boost can be set from 0 to +24 dB.

**NOTE:** In situations where the sound being recorded can be heard at the headphone monitoring position, sound waves traveling through the air can interfere with the sound heard from the headphones, altering the monitored sound. The more the sound heard through the headphones is delayed and the lower its volume, the greater the impact of the sound waves. Digital Boost adds a set boost volume to the adjusted headphone volume level, reducing the impact of the sound waves that travel through the air.

## Setting the output level

The Line Out output level can be changed.





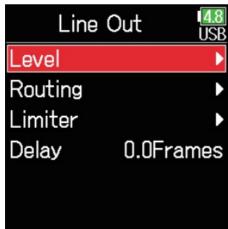


to select Line Out , and press

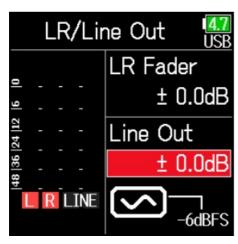




4. Use and to select Level, and press



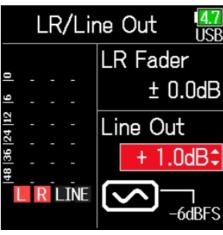












HINT: This can be set to Mute or from -48.0 to +12.0 dB

## Adjusting connected equipment levels (playing test tones)



to select the line output sine wave icon , and press

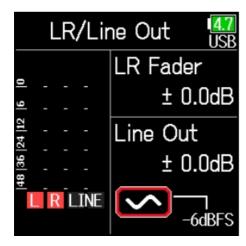


to play a test tone.





to stop test tone playback.



#### HINT:

- While checking the audio level meter of the connected device, make adjustments to the input gain of that device until the audio signal level is about –6 dB.
- The test tone is a 1kHz sine wave at -6 dBFS.

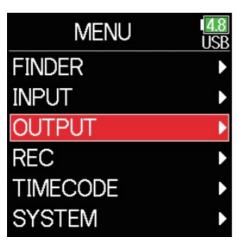
#### NOTE:

- See the manual of the connected device for information about its operation.
- If the automatic gain control function on the other device is on, turn it off.
- The test tone is output from both the LINE OUT and HEADPHONE jacks.
- Be careful with the volume if you are monitoring the sound with headphones, for example.

#### Applying delay to the output

By delaying output, timing differences for audio input into another device can be corrected.







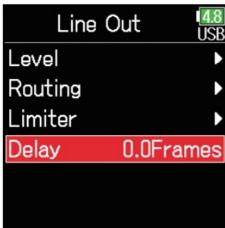
to select Line Out, and press





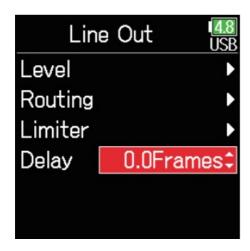


to select Delay, and press





to adjust the delay in frames, and press



#### **HINT**

This can be set from 0.0 to 10.0 frames.

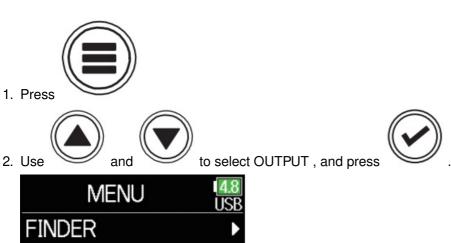
#### **NOTE**

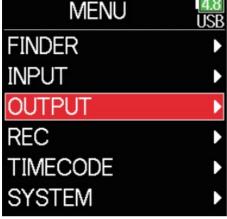
- Delays in milliseconds differ according to the frame rate of the selected timecode.
- When Sample Rate is set to 192 kHz, Output Delay is disabled.

# **Output Limiter**

Using a limiter on the output can protect devices connected to the output jacks.

HINT: For details about the limiter, see "Input limiter".





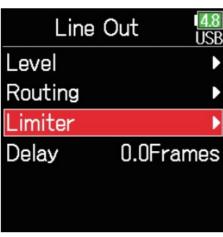






to select Limiter, and press





## Using the limiter





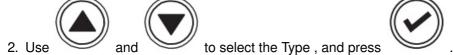




## Setting the type









## Setting the threshold

This sets the base level from which the limiter operates







to adjust the setting, and press





#### Setting the attack time

This sets the amount of time until compression starts after the input signal exceeds the threshold.



to select Attack Time, and press











## Setting the release time

This sets the amount of time until compression stops after the input signal goes below the threshold.



to select Release Time, and press





to adjust the time, and press





## Linking the limiter

The line output limiters can be linked or applied independently.









Setti ng	Explanation
Off	Separate limiter operation.
On	Link limiter operation. If the signal for either linked signal reaches the threshold, the limiter will operate on both channels.

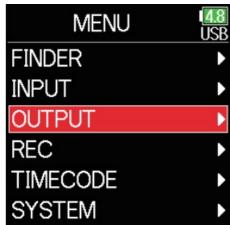
# Selecting signals sent to the line outputs

The type of signal sent to the line outputs can be set to either prefader or postfader for each track



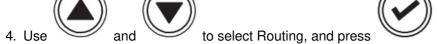
1. Press

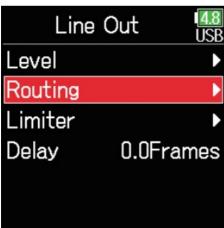


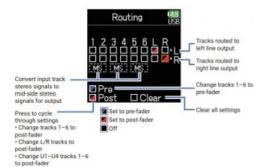












#### HINT:

Press Press

to cycle through the options: prefader  $\rightarrow$  postfader  $\rightarrow$  off

#### **NOTE**

- When AIF with Rec is set to On, USB track 1-4 can be assigned.
- Tracks 1–6 can be set to prefader or postfader.
- The L/R tracks can only be set to postfader.
- Tracks 1–6, L/R, and USB 1–4 cannot be set at the same time. Selecting one type will deselect the other.
- When mid-side stereo monitoring is enabled, the pre-fader tracks will be routed automatically to the line output channels, with odd tracks to the left and even tracks to the right. In this case, the routing cannot be changed manually.

#### **Timecode**

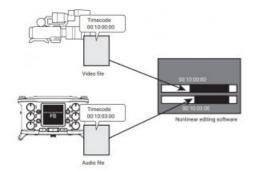
#### **Timecode overview**

The F6 can input and output SMPTE timecode.

Timecode is time information written to data when recording video and audio. It is used for video editing, control of other devices, and synchronization of audio and video, for example.

# Using timecode for editing

If video and audio data both have recorded timecode, aligning them to a timeline and synchronizing them together is easy when using nonlinear editing software for editing.



HINT: The uses a high-precision oscillator that enables the generation of accurate timecode with a discrepancy of less than 0.5 frames per 24 hours.

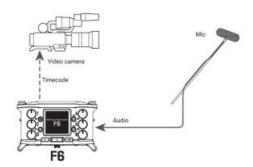
## **Connection example**

Connections like the following are possible according to application.

## Synchronizing with a video camera

The F6 records with a mic input and transmits timecode.

The records the timecode that it generates itself with the audio data. The timecode received by the video camera is recorded with the video data.

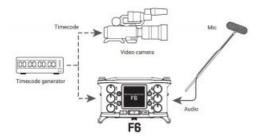


#### Inputting timecode

Timecode is transmitted from the timecode generator.

Both the F6 and the video camera receive timecode and record it with their audio and video data.

The input timecode can also be used to synchronize the audio clock of the



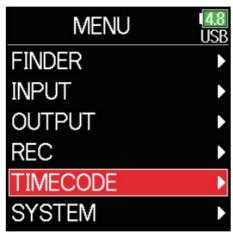
#### Setting timecode





to select TIMECODE , and press

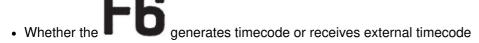




Mode	Use to set the timecode mode, timecode output when recording is stopped, synchronization with audi o clock and internal imecode operation when there is no external timecode input.
FPS	Use to set the frame rate of the internal timecode.
Jam	Use to set jamming of the timecode input through the TIMECODE IN/OUT jack by the internal timecode. This can be used to restart the internal timecode at a chosen set value.
Ubits	Use to set the mode and content of user bits that can be included in timecode.
Auto R ec Dela y	Use to set the amount of time until recording starts after timecode is received.
Start T C	Use to set the value used when jamming timecode starts and for calibration to increase the precision when jamming to RTC.

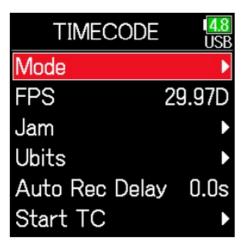
## Setting the mode

The following types of settings can be made:



• Whether timecode continues running or not when not recording







to select Mode, and press







to select the mode , and press  $% \left\{ 1,2,...,n\right\}$ 





Se tti ng	Explanation
Off	No timecode will be written to the recording file. Timecode will not be output from the TIMECODE IN/OUT j ack.
Int	Internal timecode will be generated regardless of the recording mode. The internal timecode can be set m anually using the following menu items.
Fre	MENU > TIMECODE > Jam
e Ru	MENU > TIMECODE > Restart
n	Timecode will always be output from the TIMECODE IN/OUT jack.
Int	Internal timecode will be generated only when recording. The internal timecode can be set manually using the following menu items
Re c R	MENU > TIMECODE > Jam
un	MENU > TIMECODE > Restart
	When switching from another mode, the internal timecode willstop at the last value.
Int RT	Internal timecode will be generated regardless of the recordingmode. In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock).
С	At startup
Ru	When Date/Time (RTC) changed
n	When switching to this timecode modeTimecode will always be output from the TIMECODE IN/OUTjack.
Ext	The internal timecode will chase the external timecode. When there is no external timecode, automatic gen eration of internal timecode can also be enabled.
Ext Aut o Re c	The internal timecode will chase the external timecode. When there is no external time code, automatic generation of internal timecode can also be enabled. Recording starts automatically when external timecode input is detected. Recording stops automatically when external timecode stops.

# Outputting timecode only when recording

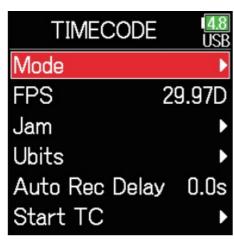
Whether or not timecode is output from the TIMECODE IN/OUT jack when recording is stopped can be set.



to select Mode, and press



1. Use



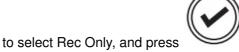


to select TC Out, and press











## **NOTE**

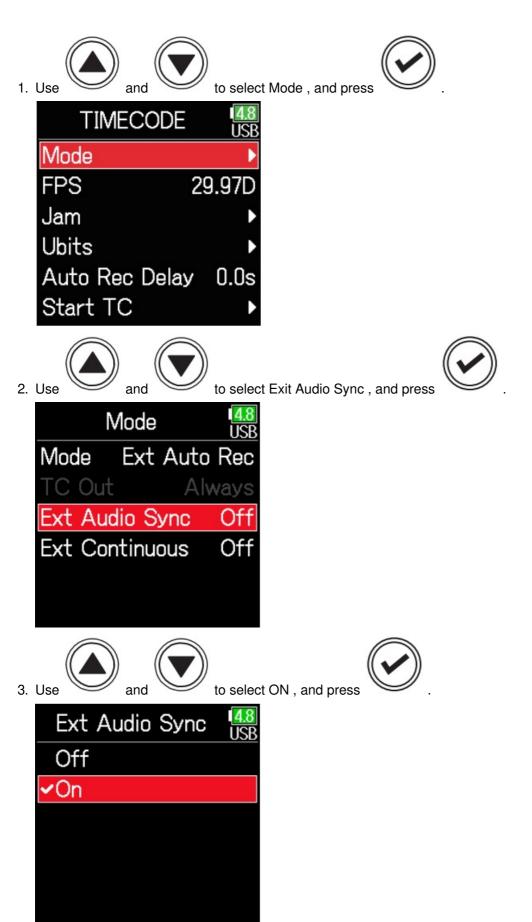
- Timecode will continue to be output when recording/playback is paused.
- This cannot be set when Mode is set to Off, Ext or Ext Auto Rec.

#### HINT

**Always:** Timecode is always output regardless of the recorder status.

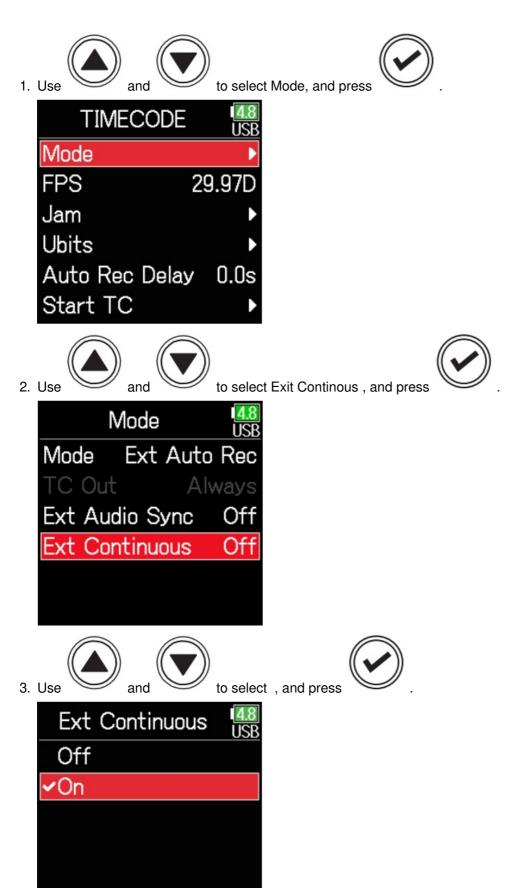
Rec Only: Timecode is output only when recording.

Synchronizing audio clock with external timecode



## Automatically enabling internal timecode when no external timecode is input

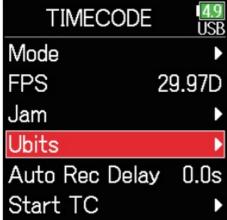
The automatic generation of internal timecode can be enabled to preserve continuity when there is no external timecode



#### Setting the user bits for internal timecode

User bits are data that can be set for inclusion in the timecode. Up to 8 numbers (0–9) and letters (A–F) can be included. Recording date information, for example, can be useful when editing later. Setting the user bits (Ubits) mode





2. Use and to select Mode, and press



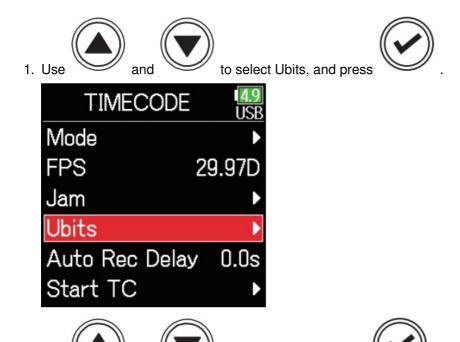
3. Use and to select the mode, and press

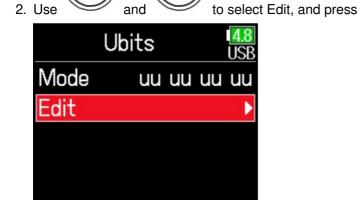


Setting	Explanation
uu uu u u uu	These values can be set as desired on the Edit screen.
mm dd yy uu	The month, day and year are entered automatically in that order using the RTC setting. The "uu" value can be set as desired on the Edit screen.
dd mm yy uu	The day, month and year are entered automatically in that order using the RTC setting. The "uu" value can be set as desired on the Edit screen.
yy mm dd uu	The year, month and day are entered automatically in that order using the RTC setting. The "uu" value can be set as desired on the Edit screen.

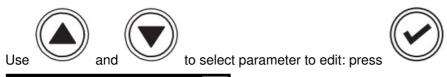
HINT: Only "uu" items can be changed.

# Setting the user bits (Ubits)





Edit the ValueMove cursor or change value



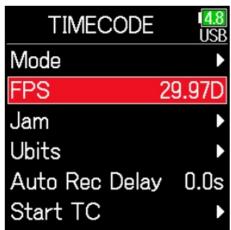


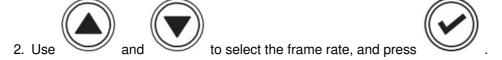
4. When done setting, use and to select Enter, and press



## Setting the frame rate for internal timecode







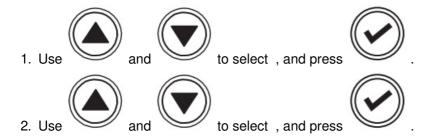


S e tt i n g	Explanation
2 3 9 7 6 N D	This is the most common frame rate used with HD cameras and other high-definition video recording. The count is slower than the actual time by 0.1%. 24ND This is the standard frame rate used for recording film. This is also used with HD cameras. 25ND This is the frame rate for PAL video. This is used for PAL video, which is used in Europe and other regions.
2 9 9 7 N D	This is a frame rate used for NTSC color video and HD cameras. The count is slower than the actual time b y 0.1%. This is used for NTSC video, which is used in Japan, the United States and other countries.
2 9 9 7 D	This is an adjusted frame rate that uses a drop frame to make NTSC match the actual time. This is used wit h video for broadcast that requires the actual time frame to be matched.
3 0 N D	This is used to synchronize sound with film that is being transferred to NTSC video. This is the standard fra me rate used for black-and-white television in Japan, the United States and other countries.
3 0 D	This rate is used for special applications. This synchronizes at 29.97 fps drop frame with film sound to be tr ansferred to NTSC. The count is faster than the actual time by 0.1%.

NOTE: Frame rates must be set in advance to match on devices used for all video and audio data.

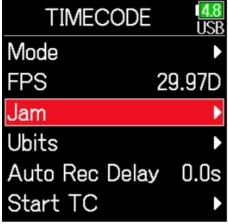
# Jamming internal timecode

Timecode input through the TIMECODE IN jack is used to set internal timecode



## Restarting internal timecode with a specified value

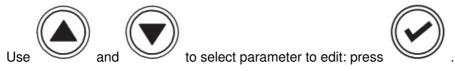








3. Select the restart Value Move cursor or change value





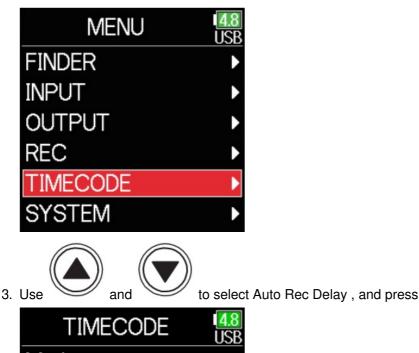


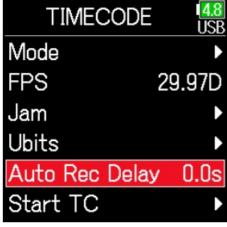


## Setting the automatic timecode recording delay

If set to record automatically when external timecode is received, unnecessary recording could occur if timecode is received for a brief amount time. In order to prevent this, the amount of time until recording starts after timecode is received can be set.







4. Use and to adjust the time, and press



**HINT:** This can be set from 0.0 to 8.0 s.

## Setting timecode initialization used at startup

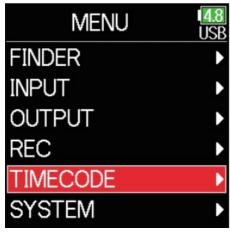
When the is turned off, the internal timecode stops, so the timecode is automatically initialized (jammed) during startup. The value that is used for jamming at that time can be set.





to select TIMEMODE, and press

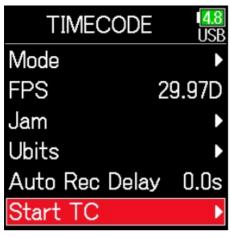






to select Start TC , and press



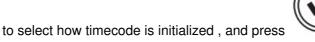


## Setting how timecode is initialized at startup











Sett ing	Explanation
Rest art Tim e	When the starts, the value set by Restart is used to jam the internal timecode.
RTC	When the starts, its timecode is restored from the timecode when the power was turned off and advanced by the elapsed time using the Date/Time (RTC) setting Since RTC is less precise than internal timecode, discrepancies will occur.

## Correcting timecode errors after the power has been turned off

When the Start TC Mode is set to RTC, timecode precision will decrease if the power is turned off. This function can be used to improve precision to almost 0.2 ppm even if the power is turned off.













3. Calibration completes





. then use



) and



to select Exit, and press

4. To cancel calibration, press



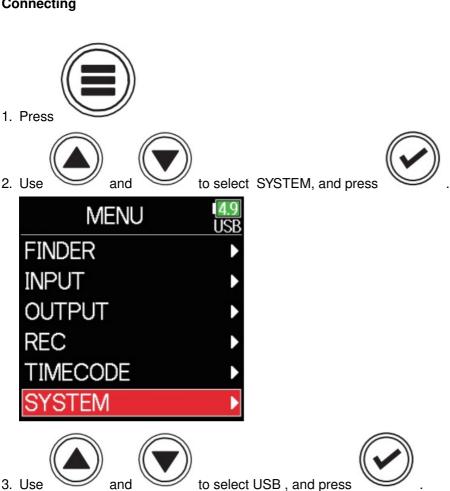


## **Using USB functions**

# Exchanging data with a computer

By connecting with a computer, data on the cards can be checked and copied.

# Connecting













5. Use a USB cable to connect the **F 6** and the computer



**NOTE:** The supported operating systems are as follows.

Windows: Windows 7 or later macOS: Mac OS X 10.8 or later

#### **Disconnecting**

1. Disconnect on the computer.

#### Windows:

Select F6 with "Safely Remove Hardware".

macOS:

Prag the F6 icon to the Trash and drop it.

**NOTE:** Always conduct computer disconnection procedures before removing the USB cable.

F6, and press

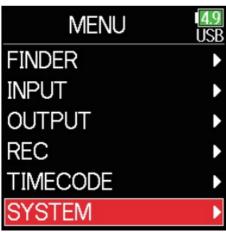
2. Disconnect the cable from the computer and the

## Using as an audio interface

F6 input signals can be input directly to a computer or iOS device, and playback signals on a computer or iOS device can be output from the F6.

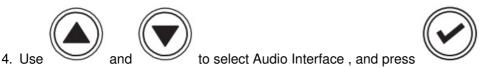
## Connecting



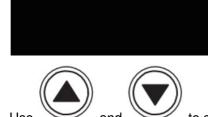












to select the mode and connected device, and press

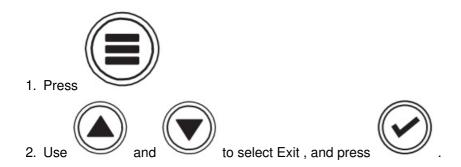


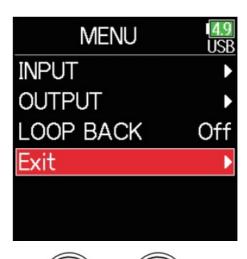


6. Use a USB cable to connect the **b** with the compute or iOS device.



## **Disconnecting**









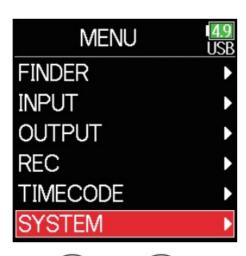
4. Disconnect the cable from the computer or iOS device and the

## Using SD card recording and audio interface functions at the same time

In addition to SD card recording, a computer can also be used to record a backup

## Connecting







SYSTEM

SD Card

USB

Bluetooth
Settings

Firmware Version

Language



English





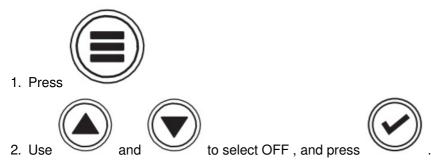


6. Use a USB cable to connect the and the computer.

## **NOTE**

- AIF with Rec cannot be used with the following settings and functions.
  - Sample rate settings other than 44.1/48 kHz
  - SD card reader
  - Audio interface
- A driver is necessary for use with Windows. Download the driver from the ZOOM website (www.zoom.co.jp/).
- When AIF with Rec is set to On, the sample rate cannot be changed.
- When AIF with Rec is set to On, files with sample rates that differ from the setting cannot be played.
- Set the input source to USB1-4 to monitor sound played back from the computer or select USB1-4 in the output routing.

#### **Disconnecting**





3. Disconnect the cable from the computer and the



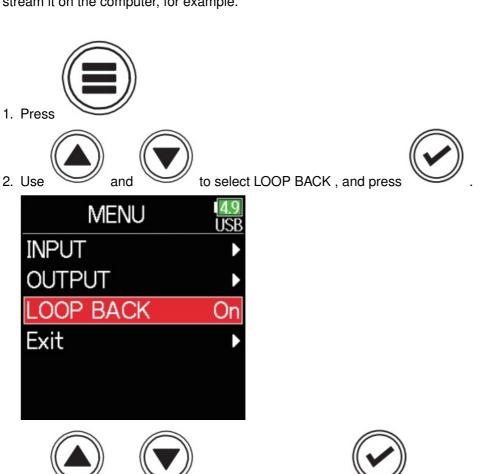
## **Audio interface settings**

The following settings can be made when using the **F6** as an audio interface

## **Setting loop back (Stereo Mix only)**

This function mixes the playback sound from the computer or iOS device with the input and sends the mix back to the computer or iOS device (loop back).

This function can be used to add narration to music played back from the computer and to record the mix or stream it on the computer, for example.





#### **Mixing inputs**

The mix balance of the inputs can be adjusted. Input signals will be sent to the computer or iOS device using the balance settings made here. When using a Stereo Mix setting, the mixed stereo signal will be sent.

1. Open the mixer on the Home Screen



2. Adjust the parameter settings. See "Adjusting the input signal monitoring balance" for how to change settings.

### Using an FRC-8 as a controller

When an FRC-8 is connected to the , it can be used to adjust settings, including trim, fader and pan.

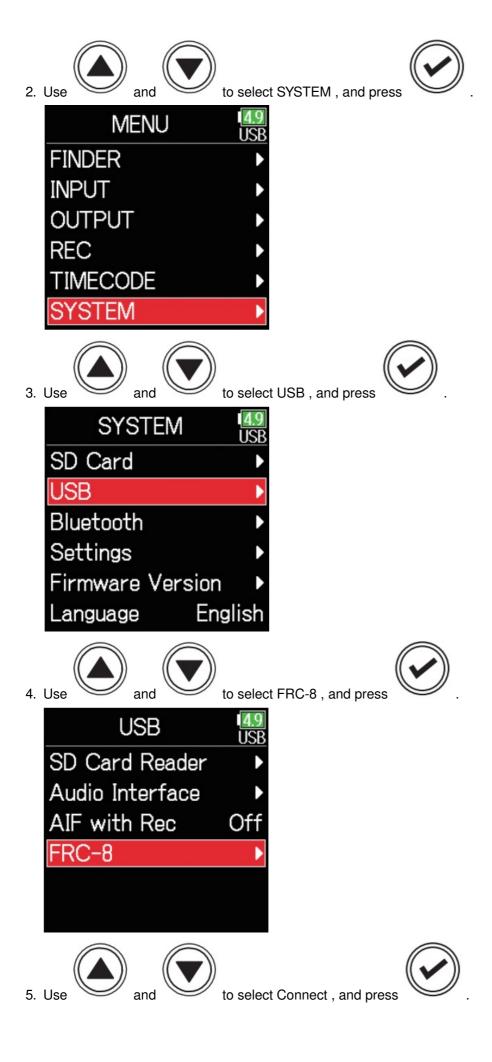
#### **NOTE**

An FRC-8 cannot be used when operating with AA batteries. When multiple power supplies are connected to an

**F6**, the power supply being used

will automatically change according to the remaining battery charge. When it switches to AA batteries, connection with an FRC-8 will be interrupted.







he **F6** and the FBC-8.

- 6. Use a USB cable to connect the
- 7. Turn the FRC-8 power on.

#### **NOTE**

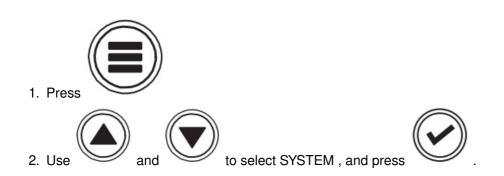
- When disconnecting the , select Disconnect before unplugging the USB cable
- Select Connect and press to supply bus power from the USB port.
   When bus power is being supplied, do not connect any device other than the . Doing so could damage the or a connected device.

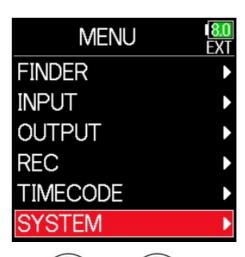
## HINT

When an FRV-8 are connected, the FRC-8 will always operate on USB bus power. AA batteries and DC power supply connected to the it are disabled.

#### Setting the type of keyboard connected to the FRC-8

A PC keyboard can be connected to the FRC-8 and used to input characters. Set the type to use the PC keyboard connected to the FRC-8.



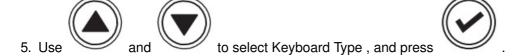






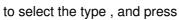
4. Use and to select FRC-8, and press













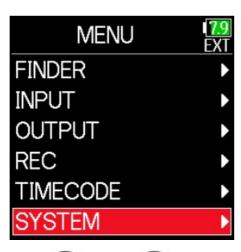


Setting	Explanation
US	This setting is for English-language keyboards.
JP	This setting is for Japanese keyboards.

## Setting user keys for the FRC-8

Functions can be assigned to the FRC-8 user keys



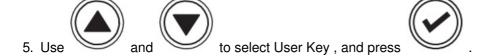
















to select the key to which to assign a function, and press





7. Use and to select the function to assign, and press

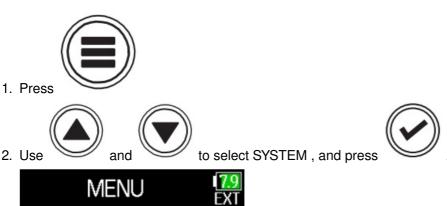


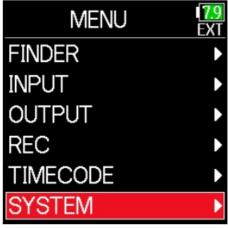


Setting	Explanation
None	No function is assigned.
Mark	Add marks to WAV format takes during recording and playback.
Key Hold	Use to disable the controls set with Key Hold Target.
Clear Clip Indicator	Clear the level meter clipping indicators.
Circled	Circle the currently selected take.

## **Setting the FRC-8 LED brightness**

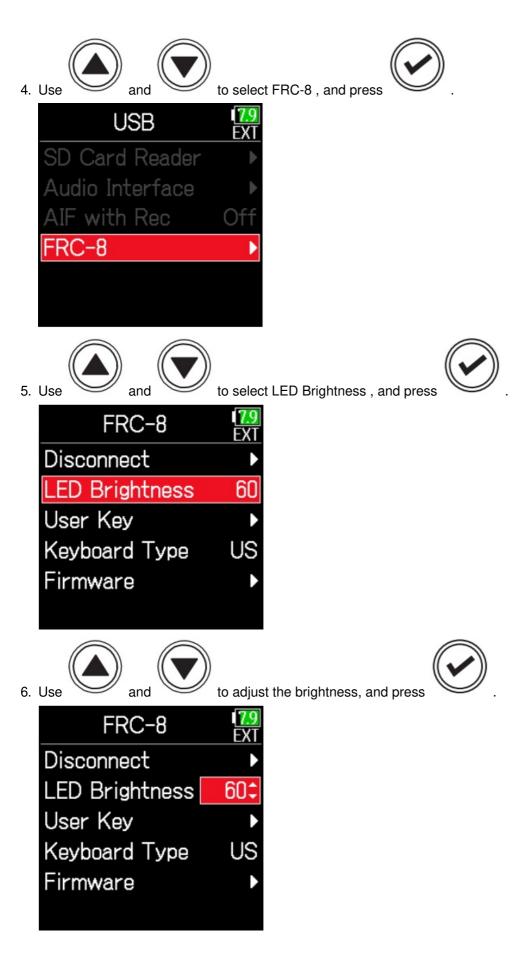
The brightness of the LEDs on the front of the FRC-8 can be adjusted.











HINT This can be set from 5 to 100.

#### **Updating the FRC-8 firmware**

The FRC-8 firmware version can be checked and updated to the latest version. The latest update file can be

1. See "Using an FRC-8 as a controller", and connect the

NOTE: Updating is not possible if the remaining power of the L battery is low. In this case, use a charged L battery.

- 2. Copy the update file to the root directory on an SD card.
- 3. Load the SD card into the SD slot.



4. Press



6. Use to select USB, and press











### Updating the firmware













4. After the update completes, turn the FRX-8 power off.



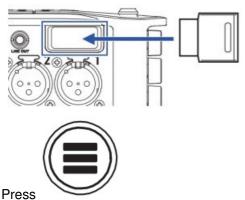
**NOTE:** Do not turn the power off, remove an SD card or disconnect the USB cable during an update. Doing so could cause the to become unstartable.

Operating with an iOS device

Connecting with an iOS device

#### **NOTE**

- The dedicated app must be installed on the iOS device beforehand. The dedicated app can be downloaded from the App Store.
- See the manual for the app for procedures to set and operate it.
- 1. Remove the wireless adapter connector cover and connect the wireless adapter.

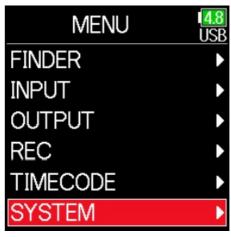


2. Press



to select SYSTEM, and press







to select Bluetooth, and press







.Select this according to the version used by the connected iOS device.



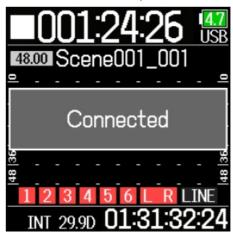






7. Launch the dedicated app on the iOS device.

This starts connection.



#### **HINT**

• If a request for pairing appears from Control, input the password shown on the recorder.



 If connection is not successful, move the iOS device closer to the recorder or move both to a place where nothing interferes with radio waves and

start Control again. Confirm also that the Bluetooth function of the iOS device can be used. If connection is still not possible, follow the instructions in the iOS device operation manual to unregister the as a Bluetooth device on it. Then, repeat the procedures from the beginning.

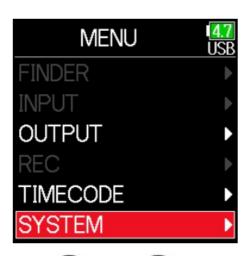
#### Disconnecting from an iOS device

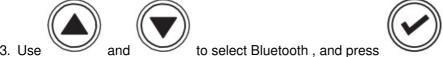


2 Use and



to select SYSTEM, and press











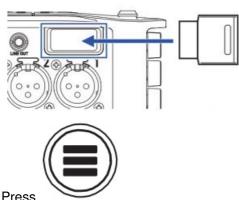




## Connecting with an UltraSync BLUE

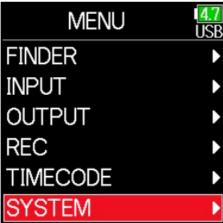
If the is connected to an UltraSync BLUE, it can receive timecode from the UltraSync BLUE and add it to recording files.

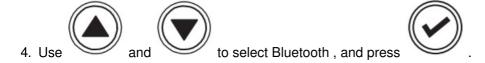
1. Remove the wireless adapter connector cover and connect the wireless adapter.



2. Press











to select Timecode, and press







to select Connect, and press

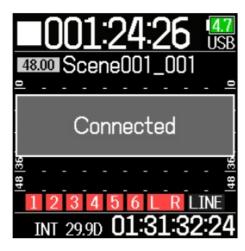


Searching for the connected device will begin and "Searching" will appear on the display.



7. Select the as a connected device on the UltraSync BLUE.

When pairing completes, "Connected" will appear on the figure 1.



#### **HINT**

• See the UltraSync BLUE manual for the procedures to select connected devices.

and the UltraSync BLUE as close together as possible to make communication more reliable.

• Even if communication with the UltraSync BLUE is interrupted, timecode

will be added to recording files.

#### Disconnecting from an UltraSync BLUE

and the UltraSync BLUE to stop recording timecode from it. Pairing information will be retained even when disconnected.

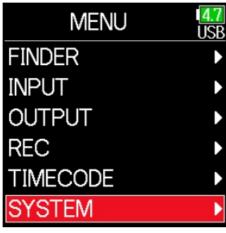


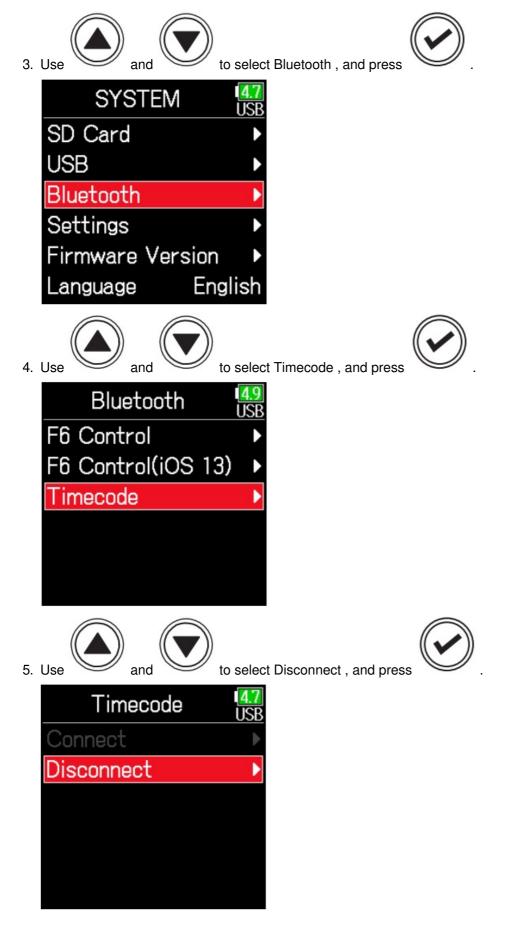
1. Press



to select SYSTEM, and press







## Connecting to a different UltraSync BLUE

To receive timecode from an UltraSync BLUE other than the one connected to the , the pairing with the current UltraSync BLUE must be removed, and pairing with the other UltraSync BLUE must be conducted.

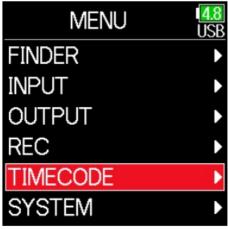


1. Press



to select TIMECODE, and press







to select Pair/Forget, and press







to select Forget, and press







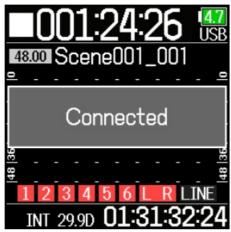
Searching for the connected device will begin and "Searching" will appear on the display.



#### HINT

- Searching can be canceled by pressing any button.
- After canceling searching, it can be restarted by selecting
   Menu > Timecode > Pair/Forget > Pair again.
- 6. Select as the connected device on the other UltraSync BLUE.

When pairing completes, "Connected" will appear on the 6 display



#### HINT

- See the UltraSync BLUE manual for the procedures to select connected devices.
- Use the and the UltraSync BLUE as close together as possible to make communication more reliable.
- Even if communication with the UltraSync BLUE is interrupted, timecode generated by the will be added to recording files.

### Other settings

## Setting the level meter peak hold time

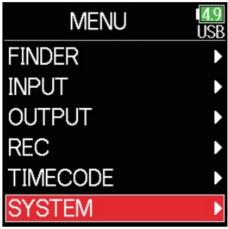


1. Press



to select SYSTEM, and press







to select Settings, and press

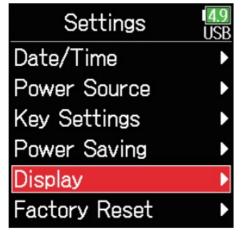


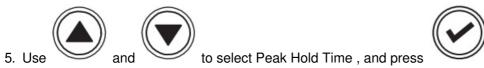




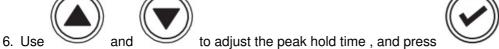
to select Display, and press









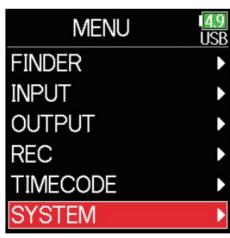




#### **Setting the LED brightness**

The brightness of the LEDs on the front of the Can be s

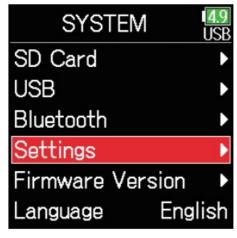






to select Setting, and press

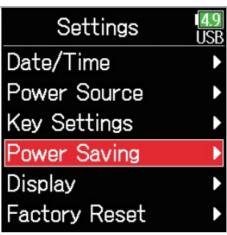






to select Power Saving , and press







to select LED Brightness , and press







to adjust the brightness, and press



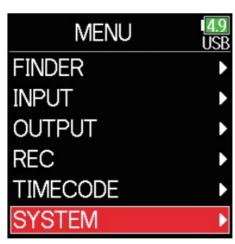


HINT: This can be set from 5 to 100.

#### Making display settings

Settings related to the display can be made.







to select Setting, and press



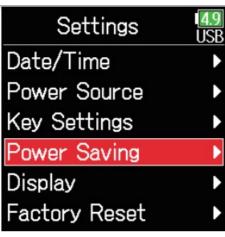


## Setting the display brightness



to select Power Saving , and press







to select LED Brightness, and press





to adjust the brightness , and press  $% \left( 1\right) =\left( 1\right) \left( 1\right) \left($ 

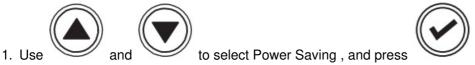


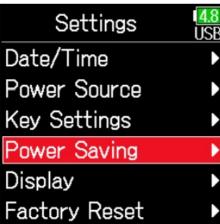


HINT: This can be set from 5 to 100.

#### Changing the display backlight setting

The display backlight can be set to dim when 30 seconds pass without use.







to select Power Saving, and press





3. Use and

to select the setting, and press





Setting	Explanation			
Off	The backlight brightness does not change even after time passes without use.			
On (Low-Backlight)	The backlight dims after time without use.			
On (Backlight-Off)	The backlight turns off after time without use			

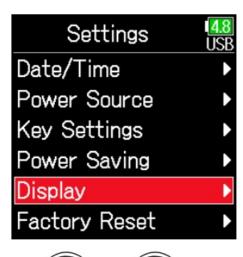
#### Making the display easier to read under bright light

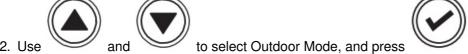
The display can be set to be easier to read in bright environments including in sunlight.



to select Display, and press

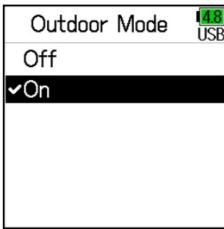












#### Setting how marks are added manually

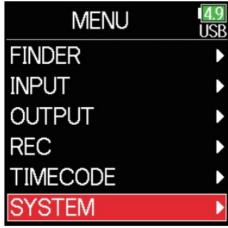
How marks are added when is pressed while recording or playing back a WAV format file can be set.





to select SYSTEM, and press







to select Setting, and press

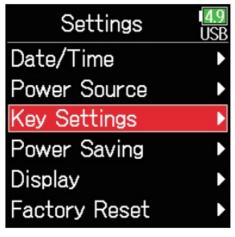






to select Key Setting, and press





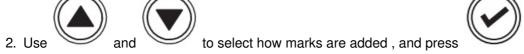




#### Setting how marks are added when recording







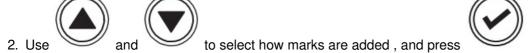


Setting	Explanation
Pause Only	Pressing will pause without adding a mark.
Pause & Mark	Pressing will pause and add a mark.
Mark Only	Pressing will add a mark without pausing.

## Setting how marks are added when playing





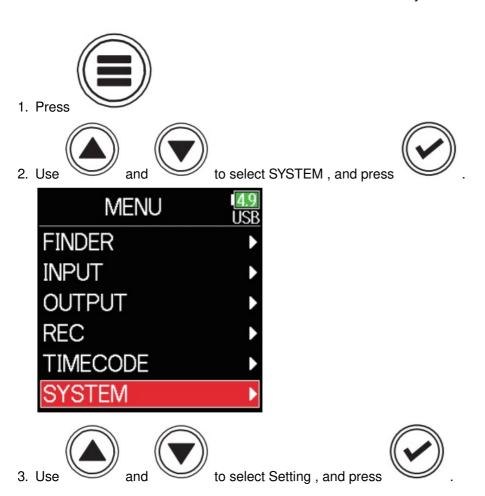




Setting	Explanation
Pause Only	Pressing will pause without adding a mark.
Pause & Mark	Pressing will pause and add a mark.
Mark Only	Pressing will add a mark without pausing.

## Setting the buttons held

Use the hold function to prevent misoperation during recording. Press and hold to enable and disable the hold function. Follow these instructions to set which keys are disabled by the hold function.

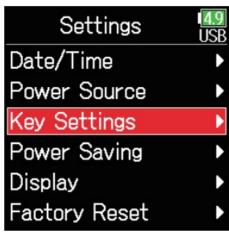






to select Key Setting, and press







to select Key Hold Target, and press

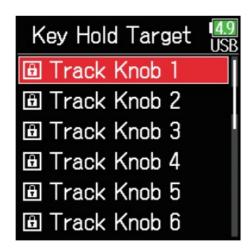






to select Key to be held, and press





**HINT:** Track Knobs 1–6, MENU, ENTER, UP, DOWN, PLAY, REC, STOP, HP Volume Push and HP Volume Turn can be selected.

HINT



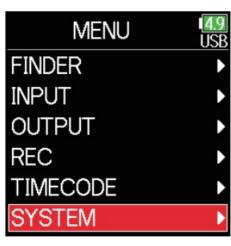
- Even when hold is on for HP Volume Push, pressing and holding will turn the hold function off.
- Operation using the FRC-8 and Control is possible even when the hold function is on.

#### Other functions

#### **Checking SD card information**

The size and open space of SD cards can be checked.







to select SD Card , and press

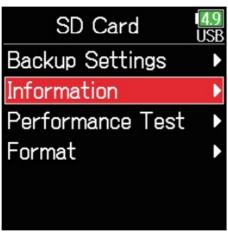




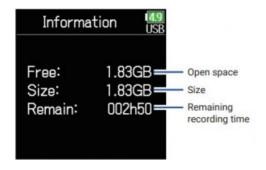


to select Information, and press



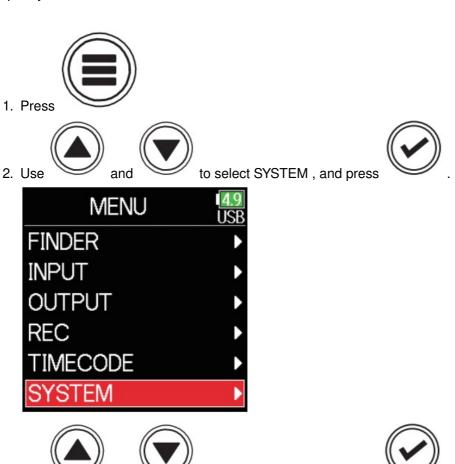


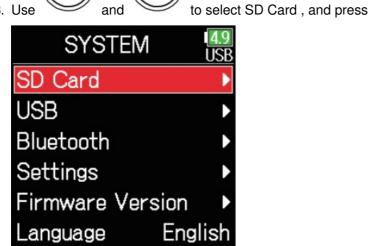
**SD** card information



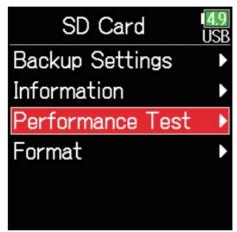
#### **Testing SD card performance**

SD cards can be tested to confirm whether they can be used with the . A basic test can be done quickly, while a full test examines the entire SD card.





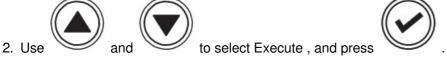


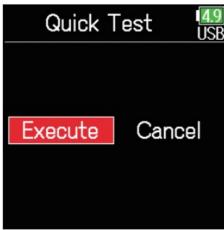


## Conducting a quick test







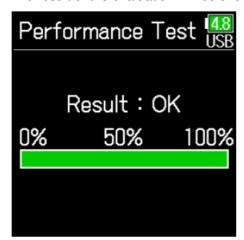


The card performance test will start.

The test should take about 30 seconds.

#### 3. The test completes.

The result of the evaluation will be shown.





#### Conducting a full test



The amount of time required for the full test will be shown.

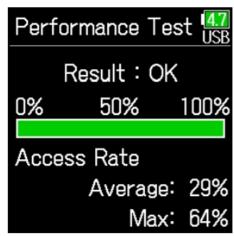






#### 3. The test completes.

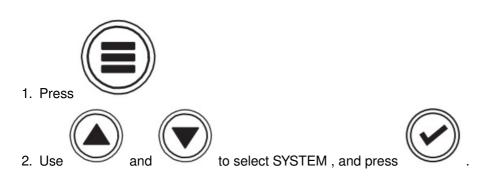
The result of the evaluation will be shown. If the access rate MAX reaches 100%, the card will fail (NG).

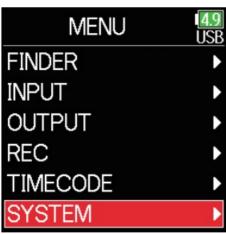




#### Formatting SD cards

Formatting SD cards for use with the







to select SD Card, and press

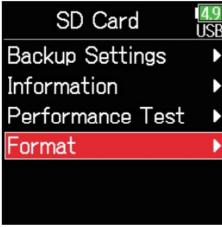




4. Use and to select



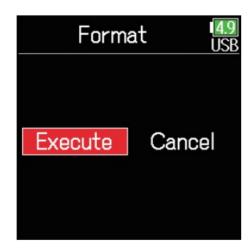
I. Use and to select Format, and press







to select Execute, and press



#### **NOTE**

• Before using SD cards that have just been purchased or that have been

formatted on a computer, they must be formatted by the

• Be aware that all data previously saved on the SD card will be deleted when it is formatted.

has a shortcut feature that allows quick access to various functions. See the "List of shortcuts" to check the shortcut functions.

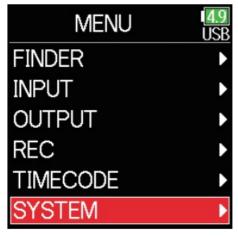


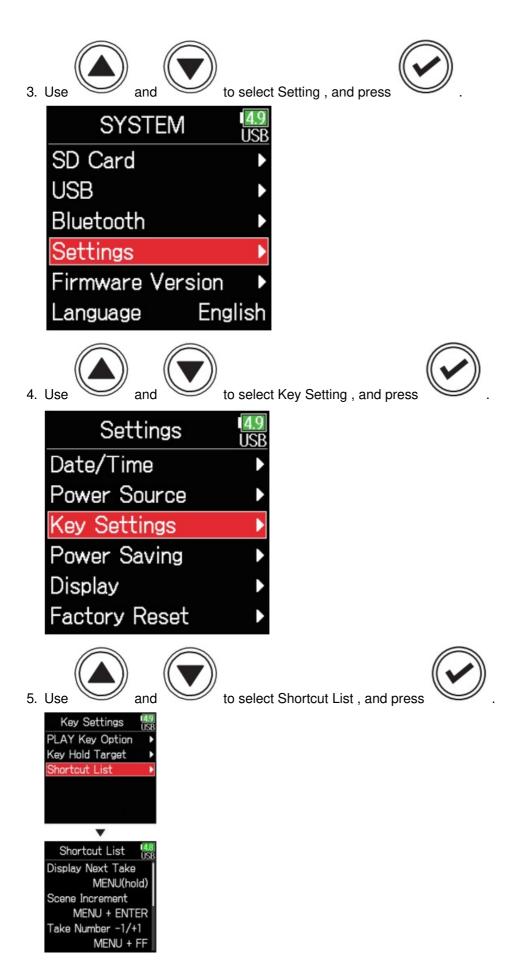
1. Press



to select SYSTEM, and press









F6 settings can be backed up to and loaded from SD cards.

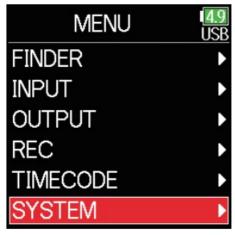


1. Press



to select SYSTEM, and press







to select SD Card, and press

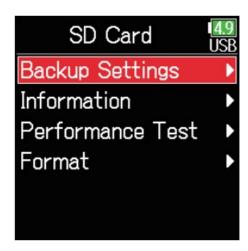






to select Backup Setting, and press





#### Backing up

This saves a backup file to the "F6\_SETTINGS" folder in the root directory of the SD card.

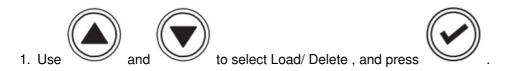


2. Edit the name of the file saved. See "Character input screen" for how to input characters.



# Loading

Backup files that are saved in the "F6\_SETTINGS" folder in the root directory of the SD card can be loaded.

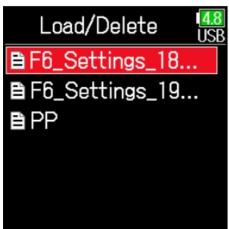






to select the file to load , and press











## Restoring default setting values

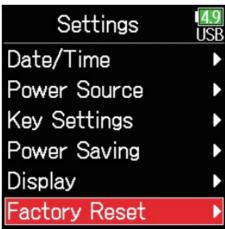
The factory default settings can be restored.



1. Press



4. Use and to select Factory Reset, and press



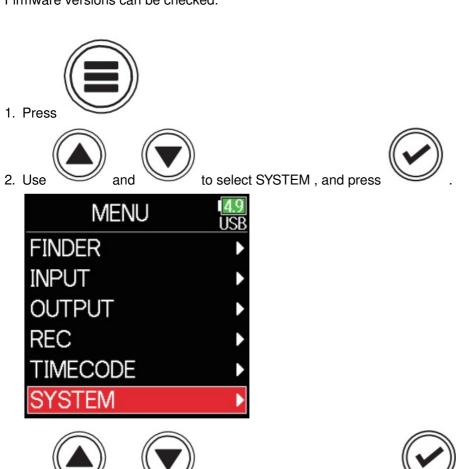


The settings will be reset and the power will automatically turn off.



## Checking the firmware version

Firmware versions can be checked.



to Firmware Version , and press



#### Updating the firmware

The firmware can be updated to the latest versions.

The latest update file can be downloaded from the ZOOM website (www.zoom.co.jp)

1. Install new batteries in the or connect the dedicated AC adapter to the USB connector.

**NOTE:** Upgrading is not possible if the remaining battery power is low. In this case, replace the batteries with new ones or use the dedicated adapter.

- 2. Copy the update file to the root directory on an SD card.
- 3. Load the SD card into the card slot, and turn the power





NOTE:Do not turn the power off or remove the SD card during the update. Doing so could cause the

F6 to become unstartable.

5. After the update completes, turn the power off.



**NOTE:** In the unlikely event that a firmware update should fail while in progress, conduct the procedures from the beginning to update the firmware again.

#### **Appendix**

#### **Troubleshooting**

If you think that the is operating strangely, check the following items first.

#### Recording/playback trouble

- · There is no sound or output is very quiet
  - Check the connections to the monitoring system and its volume setting.
  - · Confirm that the volume of the is not too low.
- · Sound from connected equipment or inputs cannot be heard or is very quiet
  - Check the input level settings.
  - If a CD player or other device is connected to an input jack, raise the output level of that device.
  - Check the input signal monitoring settings.
  - Check the phantom power and plug-in power settings.
  - · Check the headphone and line output routing settings.
- Recording is not possible
  - · Confirm that the status indicators are lit red.
  - Confirm that the SD card has open space.
  - Confirm that an SD card is loaded properly in a card slot.
  - If "Card Protected!" appears on the display, the SD card write protection is enabled. Slide the lock switch on the SD card to disable write protection.
- The recorded sound cannot be heard or is very quiet
  - · Confirm that the volume levels of the tracks are not too low.
  - Confirm that the status indicators are lit green during playback.

#### Other trouble

- Computer does not recognize it even though it is connected to the USB port.
  - · Confirm that the operating system is compatible.

 $_{\circ}$  The operation mode must be set on the F6 to allow the computer to recognize the F6

- Battery operation time is short Making the following settings could increase the battery operation time.
  - Set the power supply used correctly.
  - Turn unnecessary tracks off.
  - Disconnect unneeded devices that are plugged into the HEADPHONE, LINE OUT or TIMECODE IN/OUT jacks, for example.
  - Set the phantom power voltage to 24V.
  - Disable phantom power during playback.
  - Turn timecode off if not using it.
  - Reduce the LED brightness.
  - Reduce the LCD brightness.
  - Set the display to dim when not used for some time.
  - Reduce the sampling rate used to record files.
  - Due to their characteristics, using nickel metal hydride batteries (especially high-capacity ones) or lithium batteries should enable longer use than alkaline batteries when power consumption is high.

#### Metadata list

Metadata contained in WAV file BEXT chunks

Tag	Explanation	Remarks
zSPEE D=	Frame rate	MENU > TIMECODE > FPS
zTAKE =	Take number	
zUBITS =	Ubits	MENU > TIMECODE > Ubits
zSCEN E=	Scene Name	<ul> <li>MENU &gt; REC &gt; Metadata &gt; Scene Name &gt; Mode</li> <li>MENU &gt; REC &gt; etadata &gt; Scene Name &gt; User Name</li> <li>MENU &gt; FINDER &gt; Option &gt; Metadata Edit&gt; Scene &gt; Scene/Take</li> <li>MENU &gt; FINDER &gt; Option &gt; Rename</li> </ul>
zTAPE =	Name of recording desti nation folder	MENU > FINDER (recording destination folder name)MENU > FINDER > Option > Metadata Edit > Tape Name
zCIRC LED=	Circled take	MENU > FINDER > Option > Metadata Edit > Circle
zTRK1 =	Left track name	
zTRK2 =	Right track name	
zTRK3 =	Track 1 name	
zTRK4 =	Track 2 name	Track names are written as follows.
zTRK5 =	Track 3 name	• TRK1=TrL, TRK2=TrR, TRK3=Tr1, TRK4=Tr2 TRK8=Tr6
zTRK6 =	Track 4 name	
zTRK7	Track 5 name	
zTRK8 =	Track 6 name	
zNOTE =	Take note	MENU > Metadata > Note MENU > FINDER > Option > Metadata Edit > No te

# Metadata contained in WAV file iXML chunks

iXML mas ter tag	iXML s ub tag	Writt	Read	Remarks
<proje CT&gt;</proje 			$\bigcirc$	MENU > FINDER (folder name at top SD card level)  MENU > FINDER > Option > Metadata Edit > Project Name
<scene &gt;</scene 		0		MENU > REC > Metadata > Scene Name > Mode  MENU > REC > Metadata > Scene Name > User Name  MENU > FINDER > Option > Metadata Edit > Scene >
Scene/Ta ke			$\bigcirc$	MENU > FINDER > Option > Rename
<take></take>			$\bigcirc$	MENU > FINDER > Option > Metadata Edit > Take  MENU > FINDER > Option > Rename
<tape></tape>		$\bigcirc$	$\bigcirc$	MENU > FINDER (recording destination folder name)MENU > FINDER > Option > Metadata Edit > Folder (Tape)Name
<circle D&gt;</circle 		0	$\bigcirc$	MENU > FINDER > Option > Metadata Edit > Circle
<wild t<br="">RACK&gt;</wild>		×	X	
<false START&gt;</false 		×	X	
<no go<br="">OD&gt;</no>		×	×	
<file_ui D&gt;</file_ui 		$\bigcirc$	×	
<ubits></ubits>		0	X	MENU > TIMECODE > Timecode > Ubits
<note></note>				MENU > REC > Metadata > Note  MENU > FINDER > Option > Metadata Edit > Note
<bext></bext>		×	×	
<user></user>		×	×	

iXML master t	iXML sub tag	Written	Read	Remarks
<speed></speed>				
<speed></speed>	<note></note>	$\bigcirc$	×	
<speed></speed>	<master_speed></master_speed>		$\bigcirc$	MENU > TIMECODE > F PS
<speed></speed>	<current_speed></current_speed>		×	MENU > TIMECODE > F PS
<speed></speed>	<timecode_rate< td=""><td></td><td>×</td><td>MENU &gt; TIMECODE &gt; F PS</td></timecode_rate<>		×	MENU > TIMECODE > F PS
<speed></speed>	<timecode_flag></timecode_flag>		×	MENU > TIMECODE > F PS
<speed></speed>	<file_sample_rate></file_sample_rate>		×	MENU > REC > Sample R ate
<speed></speed>	<audio_bit_depth></audio_bit_depth>		×	MENU > REC > Mode
<speed></speed>	<digitizer_sample_rate></digitizer_sample_rate>		×	MENU > REC > Sample R ate
<speed></speed>	<timestamp_samples_since_midnigh T_HI&gt;</timestamp_samples_since_midnigh 		×	
<speed></speed>	TIMESTAMP_SAMPLES_SINCE_MIDNIGH T_LO>		×	
<speed></speed>	<timestamp_sample_rate></timestamp_sample_rate>		×	MENU > REC > Sample R ate

iXML master tag	iXML sub tag	Written	Read	Remarks
<sync_point_list></sync_point_list>				
<sync_point></sync_point>	<sync_point_type></sync_point_type>	×	×	
<sync_point></sync_point>	<sync_point_function></sync_point_function>	×	×	
<sync_point></sync_point>	<sync_point_comment></sync_point_comment>	×	×	
<sync_point></sync_point>	<sync_point_low></sync_point_low>	×	×	
<sync_point></sync_point>	<sync_point_high></sync_point_high>	×	×	
<sync_point></sync_point>	<sync_point_event_duration></sync_point_event_duration>	×	×	

iXML master tag	iXML sub tag	Written	Read	Remarks
<history></history>				
<history></history>	<original_filename></original_filename>		×	
<history></history>	<parent_filename></parent_filename>	×	×	
<history></history>	<parent_uid></parent_uid>	×	×	

iXML master tag	iXML sub tag	Written	Read	Remarks
<file_set></file_set>				
<file_set></file_set>	<total_files></total_files>	$\bigcirc$	×	
<file_set></file_set>	<family_uid></family_uid>	$\bigcirc$	×	
<file_set></file_set>	<family_name></family_name>	×	×	
<file_set></file_set>	<file_set_start_time_hi></file_set_start_time_hi>	×	×	
<file_set></file_set>	<file_set_start_time_lo></file_set_start_time_lo>	×	×	
<file_set></file_set>	<file_set_index></file_set_index>		×	

iXML master t	iXML sub tag	Written	Read	Remarks
<track_list></track_list>				
<track_list></track_list>	<track_count></track_count>		×	
<track/>	<channel_index></channel_index>		×	
<track/>	<interleave_ind EX&gt;</interleave_ind 		×	
<track/>	<name></name>	$\bigcirc$	$\bigcirc$	MENU > REC > Metadata > Track Name  MENU > FINDER > Option > Metadata Edit > Track Name
<track/>	<function></function>	×	×	

Metadata	ID3 field	Format
Timecode	Artist Name	TC=[HH:MM:SS:FF]
Scene name, take number	Track Title	SC=[scene name] TK=[take number]
Frame rate, file length (time)	Album Title	FR=[frame rate] D=[file length (time)]

#### List of shortcuts

#### **Home Screen**

Operation from	Operation fro m FRC-8	Explanation
Press and hold	Press and hold MENU	Show the name that will be given to the next take recorded. Example: Sc ene001_002
	MENU + ENCO DER press	Advance the scene number by 1 (when the Home Screen is open).
	MENU + FF	The number given to the next recorded take can be increased or decreas ed by one when the Home Screen is open.

	MENU + REW	Move the previously recorded take to the FALSE TAKE folder (when the Home Screen is open).
<ul><li> **</li></ul>	ENCODER pre ss + FF	Open L/R track fader and line output level setting screen.
	ENCODER pre ss + REW	Click the level meter clipping indicators.
Press and hold	Press and hold FF	Circle the currently selected take.

# Input link, trim link and routing screens

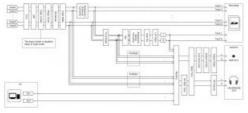
Operation from <b>F6</b>	Operation from FRC-8	Explanation
	_	Move the cursor up
	_	Move the cursor down.

#### All screens

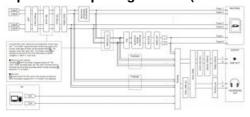
Operation from <b>F6</b>	Operation from FRC-8	Explanation
Press and hold + n -	_	Disable controls set with "Key Hold".

# **Block diagrams**

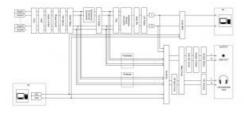
# Input and output signal flow (Linear and Dual modes)



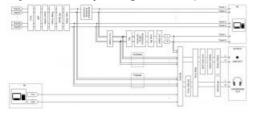
# Input and output signal flow (Float mode)



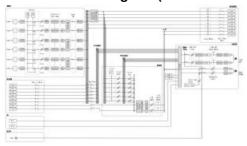
## Input and output signal flow (Audio Interface Stereo Mix)



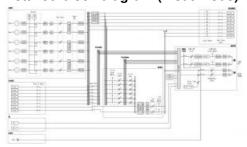
## Input and output signal flow (Audio Interface Multi Track)



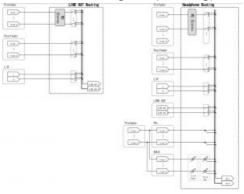
## **Detailed block diagram (Linear & Dual modes)**



# **Detailed block diagram (Float mode)**



# **Detailed block diagram (Routing)**



# **Specifications**

Recording m edia	SD cards, SDHC cards, SDXC cards (that conform to standards)		
Inputs	Inputs 1–6	Connectors	XLR jack (pin 2 hot)
	Input (mic)	Input gain	+12 dB - +75 dB
		Input impedance	3 kΩ
		Maximum input level	+4 dBu
	Input (line)	Input gain	-8 dB - +55 dB
		Input impedance	5 kΩ
		Maximum input level	+24 dBu
	Phantom power	+24/+48V 10mA maximum for each channel	
	Equivalent input noise	-127 dBu or less (A-weighted, +75 dB input gain, 150Ω input)	
Outputs	Line output	Connectors	3.5 mm stereo mini unbalanced output
		Output impedance	100 Ω or less
		Reference output level	–10 dBV, 1 kHz, 10kΩ load
		Maximum output level	+10 dBV, 1 kHz, 10kΩ load
		D/A dynamic range	95 dB typ (-60dBFS input, A-weighted)
		Connectors	3.5 mm stereo mini unbalanced output
	Headphone output	Output impedance	15 Ω or less
		Maximum output level	100 mW + 100 mW (32Ω load)

		D/A dynamic range	108 dB typ (–60dBFS input, A-weighted)	
Recording for mats	When WAV selected			
	Supported formats	44.1/47.952/48/48.048/88.2/96/192 kHz, 16/24-bit/32-bit float, mono stereo/2-8ch poly, BWF/iXML supported		
	Maximum simultaneous recording tracks	14 (6 inputs x 2 (Liner and Floating) + LR mix) 6 (6 inputs (Liner or Floating) at 192kHz sampling rate)		
	When MP3 selected			
	Supported formats	128/192/320kbps, 44.1/48 kHz, ID3v1 tags supported		
	Maximum simultaneous recording tracks	2		
	Using a 32 GB card			
Recording tim e	30:46:00 (48 kHz/24-bit stereo WAV)			
	7:41:00 (192 kHz/24-bit stereo WAV)			
	Connector	3.5 mm stereo mini (Tip	: IN, Ring: OUT)	
	Modes	Off, Int Free Run, Int Record Run, Int RTC Run, Ext, Ext Auto Rec (udio clock can be synchronized to timecode)		
	Frame rates	23.976 ND, 24 ND, 25 ND, 29.97 ND, 29.97 D, 30 ND, 30 D		
	Precision	±0.2 ppm		
Timecode	Allowed input level	0.2 – 5.0 Vpp		
	Allowed input impedanc e	4.6 kΩ		
	Output level	3.3 Vpp		
	Output impedance	$50 \Omega$ or less		
	AC adapter (ZOOM AD-17): DC 5V/1A (supports USB bus power)			
Power	Sony® L-Series battery			
	4 AA batteries (alkaline, lithium or rechargeable NiMH batteries)			
	48 kHz/16-bit 2ch recording to SD card			
	(LINE OUT off, TIMECODE off, LED/LCD Brightness 5, headphones into 32 $\Omega$ load, PHANTON off)			
	Alkaline batteries	7.5 hours or more		
	NiMH batteries (2450 mAh)	10.5 hours or more		
	Lithium batteries	16.5 hours or more		
0 11	48 kHz/24-bit 6ch recording to SD card (LINE OUT off, TIMECODE off, LED/LCD Brightness headphones into 32Ω load, PHANTOM off)			
Continuous re cording time	Alkaline batteries	5 hours or more		

	NiMH batteries (2450 mAh)	7 hours or more			
	Lithium batteries	10.5 hours or more			
	192 kHz/24-bit 6ch recording to SD card (LINE OUT on, TIME CODE set to Int Free Run, LED/ LCD Brightness 60, headphones into 32Ω load, PHANTOM at 48 V)				
	Alkaline batteries	0.5 hours or more			
	NiMH batteries (2450 mAh)	1.5 hours or more			
	Lithium batteries	3.5 hours or more			
Display	1.54" full-color LCD (240 × 240)				
	Mass storage operation				
	Class	USB 2.0 High Speed			
	Multitrack audio interface operation (driver required for Windows, no driver required for macOS )				
	Class	USB 2.0 High Speed			
	Specifications	Sampling rate	44.1/48/88.2/96 kHz		
		Bit Rate	16/24-bit		
		Channels	6 in/4 out		
	Stereo mix audio interface operation (no driver required)				
USB	Class	USB 2.0 Full Speed			
	Specifications	Sampling rate	44.1/48 kHz		
		Bit Rate	16-bit		
		Channels	2 in/2 out		
	Note: iOS device audio interface operation supported (stereo mode only)				
	AIF with Rec operation (driver required for Windows, no driver required for macOS)				
	Class	USB 2.0 High Speed			
		Sampling rate	44.1/48 kHz		
	Specifications	Bit Rate	16/24-bit		
Power consu		Channels	8 in/4 out		
	Main unit only		1 W		
mption	Using L battery with FRC	C-8 connected	10 W		
External dime nsions	100 mm (W) × 119.8 mm (D) × 62.9 mm (H)				
Weight	520 g				

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