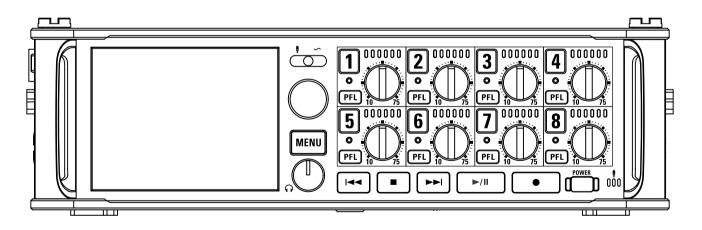




Operation Manual



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Safety Precautions

In this operation manual, symbols are used to highlight warnings and cautions that you must read to prevent accidents. The meanings of these symbols are as follows.



Something that could cause serious injury or death.

Something that could cause injury or damage to the equipment.

Other symbols used



An action that is prohibited.

A Warning

An action that is mandatory.

Operation using an AC adapter

• Never use any AC adapter other than a ZOOM AD-19.

Operation with external DC power supply

Use a 9V–16V external DC power supply.

Carefully study the warning indications of the external DC power supply before use.

Operation with batteries

- Use 8 commercially-available 1.5V AA batteries (alkaline dry cell batteries, nickel metal hydride batteries or lithium dry cell batteries).
- Carefully study the warning indications of the batteries before use.
- Always keep the battery cover closed during use.

Alterations

 $\bigotimes \ \mbox{Do not open the case or modify the product.}$

∧ Caution

Product handling

- Do not drop, bump or apply excessive force to the unit.
- Be careful not to allow foreign objects or liquids to enter the unit.

Operating environment

- \bigotimes Do not use in extremely high or low temperatures.
- O Do not use near heaters, stoves and other heat sources.
- O Do not use in very high humidity or where it could be splashed by water.
- \bigotimes Do not use in places with frequent vibrations.
- \bigotimes Do not use in places with much dust or sand.

AC adapter handling

- When disconnecting the power plug from an outlet, always pull on the plug itself.
- Disconnect the power plug from the outlet when the unit will not be used for extended periods and whenever there is lightning.

Battery handling

- Install batteries with the correct +/- orientations.
- Use the specified batteries. Do not use new and old batteries together. Do not use batteries of different brands or types together.
- Remove the batteries when the unit will not be used for extended periods. If a leak occurs, thoroughly wipe the battery case and battery terminals to remove the leaked fluid.
- Danger of explosion if battery is incorrectly replaced. Replace only with the same or equivalent type.
- A warning that batteries (battery pack or batteries installed) shall not be exposed to excessive heat such as sunshine, fire or the like.

Mic handling

- Always turn the power switch OFF before connecting a mic. Do not apply unnecessary force when connecting a mic.
- Attach the protective cap when no mic is connected for extended periods.

Connection cables and input/output jacks

- Always turn the power OFF for all equipment before connecting any cables.
- Always disconnect all connection cables and the AC adapter before moving the unit.

Volume

O Do not use at a loud volume for extended periods.

Usage Precautions

Interference with other electrical equipment In consideration of safety, the **C** has been designed to minimize its emission of electromagnetic waves and to suppress interference from external electromagnetic waves. However, equipment that is very susceptible to interference or that emits powerful electromagnetic waves could result in interference if placed nearby. If this occurs, place the **C** and and the other device farther apart.

With any type of electronic device that uses digital control, including the **FB**, electromagnetic interference could cause malfunction, corrupt or destroy data and result in other unexpected trouble. Always use caution.

Cleaning

Use a soft cloth to clean the exterior of the unit if it becomes dirty. If necessary, use a damp cloth that has been wrung out well to wipe it. Never use abrasive cleansers, wax or solvents such as alcohol, benzene or paint thinner.

Breakdown and malfunction

If the unit becomes broken or malfunctions, immediately disconnect the AC adapter or DC power supply, turn the power off and disconnect other cables. Contact the store where you bought the unit or ZOOM service center with the following information: product model, serial number and specific symptoms of breakdown or malfunction, along with your name, address and telephone number.

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Recording from copyrighted sources, including CDs, records, tapes, live performances, video works and broadcasts, without permission of the copyright holder for any purpose other than personal use is prohibited by law.

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Note about the Auto Power Off function

The power will automatically turn off if unused for 10 hours. If you want the power to instead remain on, see "Disabling the Automatic Power Saving function" on P.20 and turn the function off.

Introduction

Thank you very much for purchasing a ZOOM **FB** Multitrack Field Recorder. The **FB** has the following features:

• 8 analog input channels with super high-quality preamps

The 8 lockable XLR/TRS combo jacks provide high-quality analog inputs with EIN of -127 dBu or less, +75 dB maximum input gain and support for +4 dB input.

• PCM recording at up to 192kHz/24-bit resolution

• Recording of up to 10 tracks simultaneously

Inputs 1–8 and a stereo mix (left and right) can be recorded at the same time (8 tracks when the sampling rate is 192 kHz).

• Dual channel recording of separate files at lower levels simultaneously with ordinary recording (Inputs 1–4)

Using dual channel recording at a lower input level, you can create backup recordings to use when unexpected loud noise causes regular recordings to distort, for example.

• Newly redesigned limiters for overload protection

With 10 dB of headroom, this limiter prevents distortion even more than ordinary ones. The threshold can also be set to keep the signal below that level.

•Time code with pinpoint accuracy

The **F** utilizes a high-precision oscillator that generates timecode with accuracy of 0.2ppm, ensuring rock-solid stability when syncing audio and video.

• Outputs include a stereo headphone jack with a powerful 100mW amp as well as MAIN OUT 1/2 and SUB OUT 1/2 jacks

This allows you to send the audio signal to a video camera or other device while monitoring with headphones.

• Built-in digital mixer with flexible signal routing Prefader and postfader signals from inputs 1–8 can be freely routed to any outputs.

• Phantom power (+24V/+48V) can be supplied

This can be turned on/off for each input separately.

•Three possible power sources – batteries, an AC adapter and an external DC power supply

In addition to AA batteries and an AC adapter, a 9-16V external DC power supply can also be used.

• Double SDXC card slots

Simultaneous recording on 2 SD cards is possible, and support for SDXC cards up to 512 GB enables long-duration recording. In addition, the **F B** can be used as a card reader by connecting to a computer using USB.

• USB audio interface capabilities with up to 8 ins and 4 outs

The **F B** can be used not only as a 2-in/2-out audio interface, but also as an 8-in/4-out audio interface (driver required for Windows).

• Other useful features

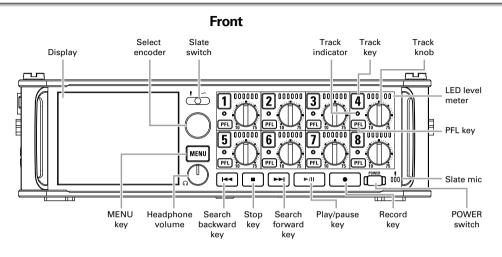
Other convenient functions include a built-in slate mic for voice memos and a variable frequency slate tone generator to confirm levels. There are also input and output delays and pre-recording of up to 6 seconds.

• ZOOM mic capsules can be connected

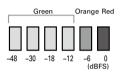
Use any ZOOM mic capsule instead of inputs 1/2.

Please read this manual carefully to fully understand the functions of the **FB** so that you can make the most of it for many years. After reading this manual, please keep it with the warranty in a safe place.

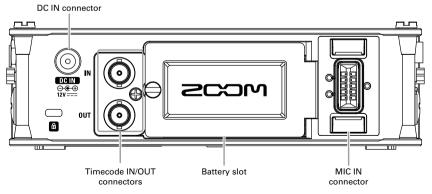
Names of parts



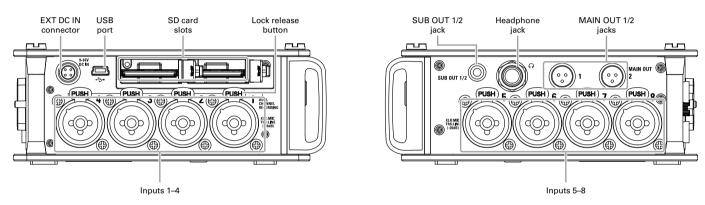




Back



Right side



Inputs 1-8

1: GND

2: HOT

3: COLD

3

XLR

 \Box

TRS

- TIP: HOT

RING: COLD

SLEEVE: GND

Left side

EXT DC IN

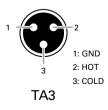
DC 9-16V

HIROSE 4-pin

2: NC

4: + 3: NC





Connecting mics/other devices to Inputs 1–8

The **FB** can record a total of 10 tracks simultaneously: 8 individual tracks with signals coming from Inputs 1–8 and a stereo mix of these inputs on left and right tracks.

You can connect mics and the outputs of line-level devices such as keyboards, mixers, or instruments with active electronics to Inputs 1–8 and record them to tracks 1–8. Alternatively, Inputs 1 and 2 can instead receive input from a ZOOM mic capsule connected to the FB MIC IN connector.

Connecting mics

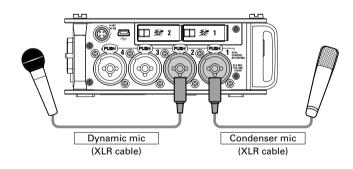
Connect dynamic and condenser mics with XLR plugs to Inputs 1–8.

Phantom power (+24V/+48V) can be supplied to condenser mics. (\rightarrow P.90)

Connecting line level equipment

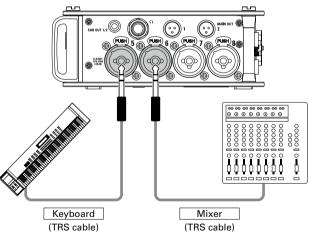
Connect the TRS plugs of keyboards and mixers directly to Inputs 1–8.

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



NOTE

When disconnecting a mic, gently pull on the XLR plug while simultaneously pushing the connector lock release button.



Connecting mic capsules

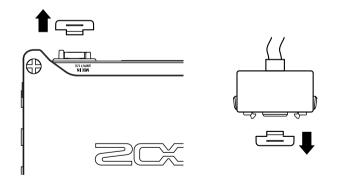
A ZOOM mic capsule can be connected to the MIC IN connector on the back of the \mathbf{FB} .

NOTE

- The mic capsule input is assigned to tracks 1/2.
- When a mic capsule is connected, Inputs 1/2 cannot be used.

Connecting and disconnecting mic capsules

1. Remove the protective caps from the **FB** and the mic capsule or extension cable.



2. While pressing the side buttons on the mic capsule or extension cable, connect it to the main unit, inserting it completely.



3. To disconnect the mic capsule or extension cable, pull

it away from the main unit while simultaneously

pressing the buttons on its sides.

NOTE

- Do not use too much force when disconnecting. Doing so could damage the mic capsule, extension cable or main unit.
- Reattach the protective cap when a mic capsule is not connected.

Stereo input

By enabling the stereo link for tracks 1/2, 3/4, 5/6 or 7/8, the corresponding Inputs (1/2, 3/4, 5/6 or 7/8) can be handled as a stereo pair. (\rightarrow P.27)

When linked, Input 1, 3, 5 or 7 becomes the left channel and Input 2, 4, 6 or 8 becomes the right channel.

Connecting mics/other devices to Inputs 1–8 (continued)

Connection examples

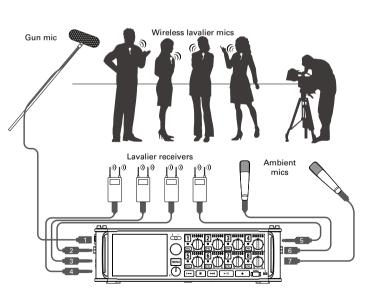
The **FB** allows you to record in a variety of situations, such as the following.

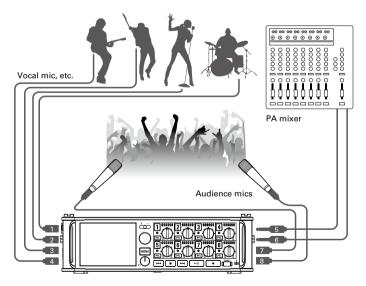
While filming

- Input 1: gun mic for main subject sound (XLR connection)
- Inputs 2–5: wireless lavalier mics for performers (TRS connections)
- Inputs 6-7: mics for ambient sound (XLR connections)

Concert recording

- Inputs 1-4: mics for stage performance (XLR connections)
- Inputs 5-6: line-level PA mixer outputs (TRS connections)
- Inputs 7-8: mics for audience sound (XLR connections)

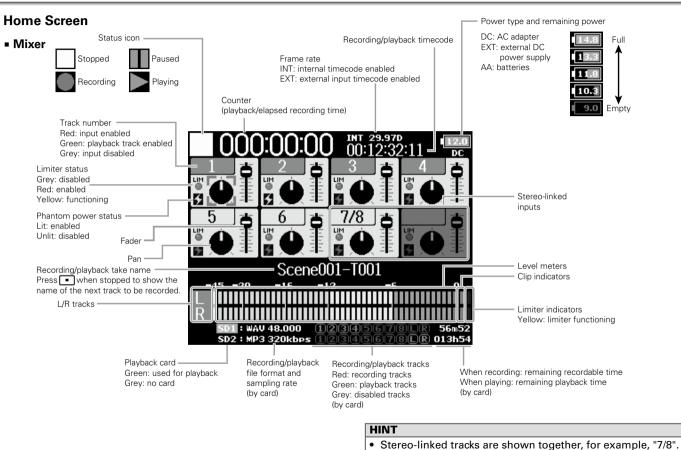




• When the Home Screen is not displayed, press and hold Imm to

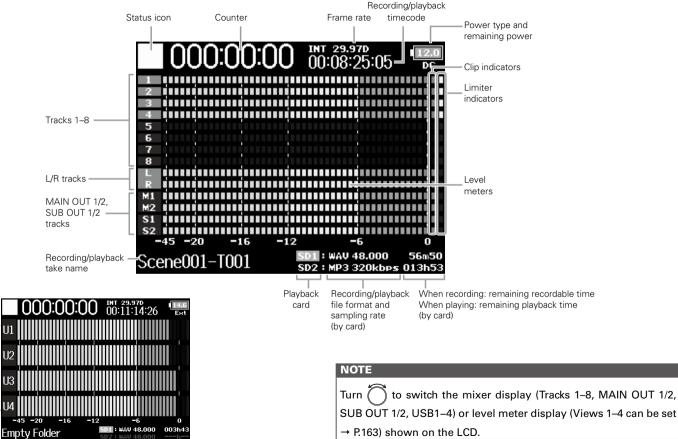
return to the Home Screen.

LCD display

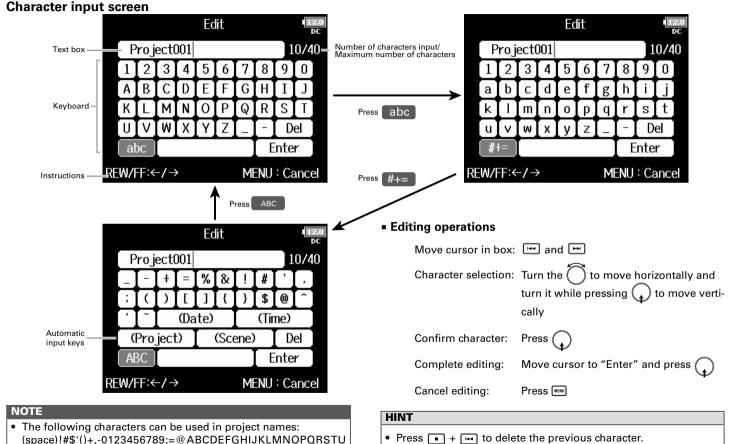


LCD display (continued)

Level meters



USB1-4 -



• Press + + to move the cursor to "Enter".

VWXYZ[]^_`abcdefghijklmnopqrstuvwxyz{}~

LCD display (continued)

Automatic input keys

(Date): Automatically inputs the date. Example: 150210 (Time): Automatically inputs the time. Example: 180950 (Project): Automatically inputs "Project***" in the field. (Scene): Automatically inputs the scene name.

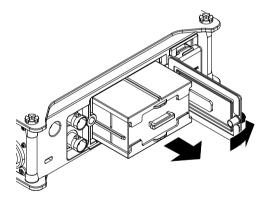
Supplying power

Using AA batteries

1 Turn the power off and then loosen the screw in the

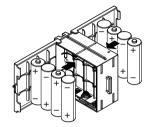
battery cover to open it.

2. Remove the battery case from the battery slot.



3. Open the battery case cover.

4. Install the batteries.



5. Replace the battery case cover.

6. Load the battery case.

NOTE

Load the case so that the side with the protruding rail is up.

7. Close the battery cover and tighten the screw.

NOTE

- Be careful because the battery case could become loose unexpectedly if the cover screw is not tightened firmly.
- Use only one type of batteries (alkaline, NiMH or lithium) at a time.
- After loading batteries, set "Power Source" to the correct type of battery. (→ P.22)
- If the remaining battery power indicator turns red, turn the power off immediately and install new batteries.

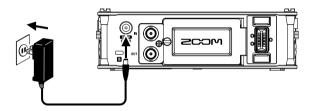
Supplying power (continued)

Using an AC adapter

1. Connect the dedicated AC adapter to the DC IN

connector.

2. Plug the dedicated AC adapter into an outlet.

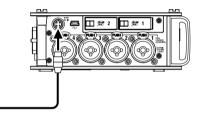


Using an external DC power supply

1. Connect the external DC power supply equipment to

the EXT DC IN connector.

Connect a 9-16V direct-current power supply.



2. If there is an adapter, plug the adapter into an outlet.

NOTE

 When connecting an external DC power supply, be sure to make the power supply settings. (→ P.22)

Preparation

Loading an SD card

1. Turn the power off and then open the SD card slot

cover.

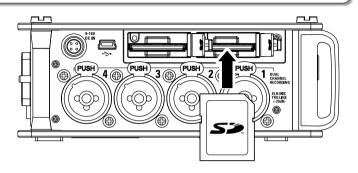
2. Insert the SD card into the SD CARD 1 or 2 slot.

To eject an SD card:

Push the card further into the slot until it clicks and then pull it out.

NOTE

- Always turn the power off before inserting or removing an SD card. Inserting or removing a card while the power is on could result in data loss.
- When inserting an SD card, be sure to insert the correct end with the top side up as shown.
- If an SD card is not loaded, recording and playback will not be possible.
- To format an SD card, see P. 176.

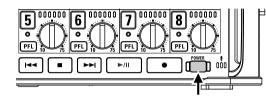


Turning the power on and off

Turning the power on

1. Press and hold briefly.

The 🛄 LED will light.



NOTE

- The first time you turn the power on after purchase, you must set the date/time (→ P. 19). You can also change this setting 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. To format an SD card, see P. 176.

Turning the power off

Press and hold D briefly.

NOTE

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

The **FB** will automatically turn off if it is unused for 10 hours.

To keep the power on continuously until powered off, see "Disabling the Automatic Power Saving function" on P.20 and set Auto Power OFF to Off.

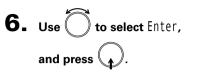
Setting the date and time (Date/Time (RTC))

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

Continue to one of the following procedures. Press MENU Setting the date and timeP.19 Setting the date formatP.20 **2.** Use to select SYSTEM, MENU Setting the date and time OUTPUT and press REC 4. Use to select Set PLAY Date/Time (RTC) TIMECODE Date Format yy/mm/dd SI ATE Date/Time, and press Set Date/Time MENU : Retur **3.** Use to select Date/Time MENU: Return SYSTEM ate/Time (RTC) (RTC), and press ower Source Auto Power Off **5.** Change the setting. Set Date/Time Home Timecode Display Size Smal Level Meter Year Month Dav LED Brightness Changing settings 2015 (WED) MENU : Retur 13 : 00 : 00 Move cursor or change value: Enter MENU : Cancel The first time you turn the **FB** on after purchasing it, you Select item to change: Press must set the date/time.

Setting the date and time (Date/Time (RTC))

Setting the date and time (Date/Time (RTC)) (continued)



This completes setting the date and time.

	Set Da	te/Time	5	12. DC
Year	Month	Day		
2015	04	01	(WED)	
13	: 0	0 :	00	
	Ent	:er		
		Μ	IENU : Car	ncel

Setting the date format

4. Use to select Date Format, and press . Date/Time (RTC) Date Format Set Date/Time MENU : Return **5.** Use to select the format, and press Date Format mm/dd/yy dd/mm/yy ✓ yy/mm/dd MENU : Return

Setting value	Explanation
mm/dd/yy	Month, day, year order
dd/mm/yy	Day, month, year order
yy/mm/dd	Year, month, day order

Disabling the Automatic Power Saving function (Auto Power Off)

The power will automatically turn off if the **FB** is unused for 10 hours. If you want the power to stay on continuously until powered off, disable the Automatic Power Saving function.

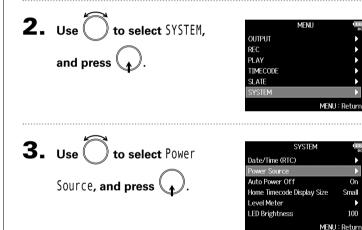
1. Press MENU.		4. Use to select Off,	Auto Power Off fee Pc On
2. Use \bigcirc to select SYSTEM, and press \bigcirc .	MENU CUIPUT CONTRUCTION CONTRUCTOR	and press ().	MENU : Return
3. Use to select Auto Power Off and press .	SYSTEM Date/Time (RTC) Power Source Auto Power Off On Home Timecode Display Size Small Level Meter LED Brightness 100 MENU : Return		

Setting the power supply used (Power Source)

Set the external DC power supply shutdown voltage, nominal voltage and type of batteries so that the remaining power supply charge can be shown accurately.

On this menu page, you can also check the voltage of each power supply and the remaining battery capacity.

Press MENU.



Continue to one of the following procedures.

Setting the external DC power supply (Ext DC) shutdown
voltage
Setting DC power supply (Ext DC) nominal voltageP.23
Setting the AA battery type (Int AA)P.23

Setting the external DC power supply (Ext DC) shutdown voltage

When an external DC power supply is being used, if the voltage drops below the value set here, the **FB** will automatically stop recording and turn off.

If AA batteries (Int AA) are installed, however, the power supply will switch to Int AA and operation will continue.



HINT

On

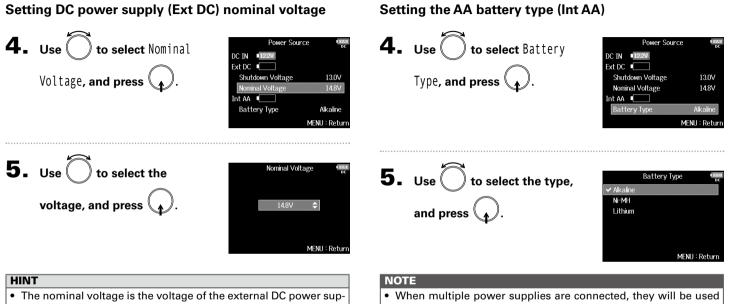
Small

100

- The shutdown voltage is the voltage when the external DC power supply runs out and can no longer supply power.
- See the manual for the external DC power supply for the shutdown voltage value.





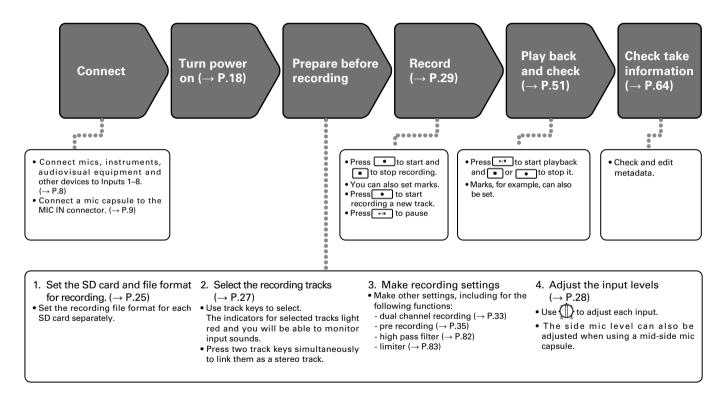


- ply under normal conditions. This value should be indicated on the outside of the external DC power supply.
- This can be set from 12.0 to 15.0 V in 0.2 V intervals.

- When multiple power supplies are connected, they will be used in the following order of precedence.
 - 1. Dedicated AC adapter (DC IN)
 - 2. External DC power supply (Ext DC)
 - 3. AA batteries in unit (Int AA)
- The voltages of each power supply are shown on the display.

Recording process

Recording with the **F B** follows the process shown below. The data created for each recording occurrence is called a "take".



Enabling recording on SD cards and setting file formats

Enabling recording on SD cards and setting file formats

The recording file format can be set independently for SD CARD slots 1 and 2.

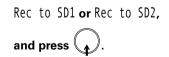
HINT

- Recording the same content to two cards is possible by using the same settings for both card slots. This function can be used to create a backup in case the sound skips on one card, for example.
- You can also record tracks 1–8 unmixed on one SD card while recording all tracks mixed together as MP3 or WAV data with left and right tracks.

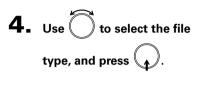
Press MENU

```
2. Use to select REC, and press .
```

3. Use to select



R	C IZ
Rec to SD1	Tr1-8 (Poly)
Rec to SD2	Tr1-8 (Poly)
Sample Rate	48kHz
WAV Bit Depth	24
MP3 Bit Rate	320kbps
Dual Channel Rec	· · · · · · · · · · · · · · · · · · ·
	MENU : Retur





Setting value	Tracks recorded	Explanation
None	-	Nothing is recorded on the SD card.
Track1-8 (Poly WAV)	Selected tracks	A single (multitrack) file is created that contains audio for multiple tracks.
Track1-8 (Mono/Stereo WAV)	1–8	A single mono file is created for each mono track and a single stereo file is created for each stereo track.
Track1-8 + L/R (Poly WAV)	All selected	A single (multitrack) file is created that contains audio for multiple tracks.
Track1-8 + L/R (Mono/Stereo WAV)	tracks	A single mono file is created for each mono track and a single stereo file is created for each stereo track.
L/R (Stereo WAV)	L/R tracks	A stereo file is created based on the
L/R (Stereo MP3)		mix created by the internal mixer.

Enabling recording on SD cards and setting file formats (continued)

NOTE

- When recording with a Mono/Stereo WAV setting, the audio files are saved in a take folder that is created. (→ P.38)
- When recording to 2 SD cards simultaneously, files will be saved in take folders with the same name on both cards. Folders will be created automatically if they do not already exist.
- If recording should stop on one SD card because, for example, it runs out of space, recording will continue on the other SD card. At such times, do not remove the card that has stopped recording from the slot. Doing so could damage the card or data.

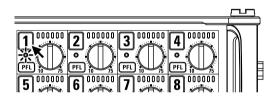
Selecting inputs and adjusting levels

You can select which of Inputs 1–8 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

• Make the track indicator light by pressing the track

key for the number of the input to record.



The background color of the track number on the LCD also changes at this time.

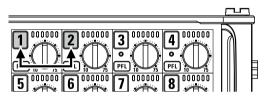
Track indicator	Track number background color	Explanation
Lit red	Red	The input is enabled.
Unlit	Gray	The input is disabled.

NOTE

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

Linking inputs as a stereo pair

• While pressing track key 1, press track key 2.



Tracks 1 and 2 will be linked as a stereo track (stereo link). Repeat the same procedure to disable the stereo link.

HINT

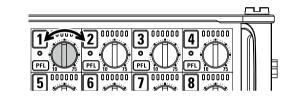
- The 3/4, 5/6 and 7/8 track pairs can also be stereo-linked in the same way.
- When a mic capsule that allows independent L and R input selection is connected, stereo-linking can also be enabled and disabled for those tracks.

Selecting inputs and adjusting levels (continued)

Adjusting input levels



level.



NOTE

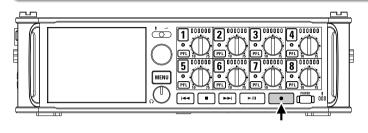
When a mic capsule is connected, for Inputs 1/2 is disabled. Use the level control on the mic capsule to adjust its input volume.

HINT

- Inputs connected with XLR plugs can be set from +10 to +75 dB, and inputs connected with TRS plugs can be set from -10 to +55 dB. Tracks with the USB set as the input source can be set from -35 to +30 dB.
- If the sound distorts even when you lower the input level, try changing mic positions and adjusting the output levels of connected devices.
- Using the limiter (\rightarrow P.83)
- Using the high pass filter (\rightarrow P.82)

Recording

Recording



1. Press **•**.

Press

This starts recording.

HINT

If the timecode function is enabled, recording will start from frame 00 (00 or 02 when using drop frame) and files will always end exactly on a second. This makes synchronization easy when editing later.

2.

to start a new take when recording.

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

NOTE

Pressing • during recording is only possible after recording for at least a second.

3. Press **▶**/**II** to pause.

NOTE

- When pausing, pausing will occur at a whole second increment.
- When recording is paused, a mark is added at that point. Press recording.
- A maximum of 99 marks can be added to a take.

HINT

- During playback, you can press and the to jump to points where marks have been added.
- You can also add marks without pausing. (→ P. 169)

4. Press **•** to stop.

NOTE

- If the maximum file size is exceeded during recording (→ P.36), recording will continue in a new take with a number that is one higher. No gap in sound will occur between the two takes when this happens.
- When recording on 2 SD cards simultaneously, if recording should stop on one because it runs out of space, recording will continue on the other SD card without interruption.

HINT

- Files are automatically saved at regular intervals during recording. Even if the power is interrupted or another unexpected problem occurs during recording, an affected file can be restored to normal by playing it with the F E.
- Press and hold when the HOME screen is open to check the name that will be given to the next take recorded.

Setting the sampling rate (Sample Rate)

You can set the sampling rate used to record files.

2. Use to select REC, and press

Press MENU.

MENU	12.0 DC
FINDER	•
META DATA (for Next Take)	
INPUT	Þ
OUTPUT	•
REC	Þ
PLAY	Þ
MENU	: Returi

```
3. Use to select Sample Rate, and press .
```

	REC IIII
Rec to SD1	Tr1-8 (Poly)
Rec to SD2	Tr1-8 (Poly)
Sample Rate	48kHz
WAV Bit Depth	24
MP3 Bit Rate	320kbps
Dual Channel Rec	•
	MENU : Return

Sample Rate

MENU : Return

44.1kHz 47.952kHz

47.952kHz (F) 48kHz

48.048kHz (F)

48.048kHz

4. Use () to select the

sampling rate, and

press

47.952kHz	per second if you want to edit at 24 frames per
	second later.
48.048kHz	Select this when recording video at 24 frames per
	second if you want to edit at NTSC 29.97 or 23.98
	HD later.
47.952kHz (F), 48.048kHz (F)	These function the same as the two above, but the
	<file_sample_rate> sampling rate metadata</file_sample_rate>
	will be recorded as 48kHz.
	This enables playback and editing with devices
	and software that do not support 47.952kHz and
	48.048kHz WAV files. Playback, however, will occur
	at the ±0.1% speed at which the file was recorded.
NOTE	
When the reco	ording file format is MP3, only 44.1kHz and 48kHz
• when the reco	nullig life format is wirs, only 44. IKHZ and 46K

Setting value

88.2kHz, 96kHz, 192kHz

44.1kHz, 48kHz,

48kHz can be selected.

Explanation

Select this when recording video at 23.976 frames

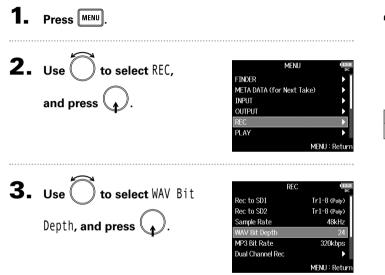
These are standard sampling rates.

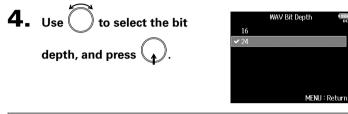
- When 192 kHz is selected, L/R tracks will not be recorded. The Input Delay and Output Delay are also disabled. Moreover, Auto Mix, Ambisonic Mode, and Input Limiter > On/Off > On (Advanced) cannot be set.
- Audio Interface with Rec cannot be used when values other than 44.1 kHz or 48 kHz are selected.

Recording

Setting WAV file bit depth (WAV Bit Depth)

You can set the bit depth of WAV files.





HINT

This can be set to 16-bit or 24-bit.

Setting MP3 file bit rate (MP3 Bit Rate)

You can set the bit rate of recorded MP3 files.





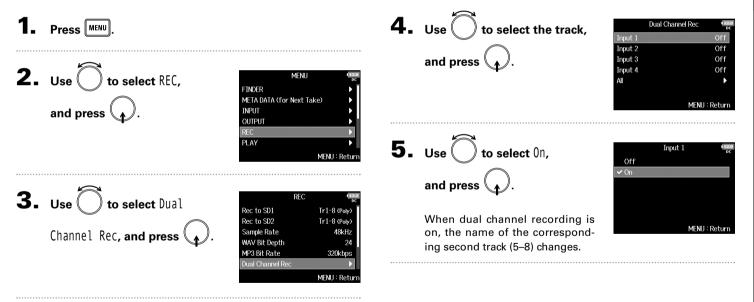
MENU : Return

HINT

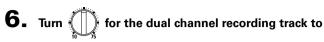
This can be set to 128 kbps, 192 kbps or 320 kbps.

Simultaneously recording tracks at different levels (Dual Channel Rec)

Along with regular recording, the **F B** can capture a second recording set to a different input level (dual channel recording). For example, by using dual channel recording to record at an input level 12 dB below that of the regular recording, you have an immediate replacement if the regular recording distorts because the track level is too high. Dual channel recording can be used with tracks 1–4.



Simultaneously recording tracks at different levels (Dual Channel Rec) (continued)



adjust the input level.

For example, when track 1 is selected, adjust for track 5.

Dual channel recording increases the amount of space used on SD cards.

NOTE

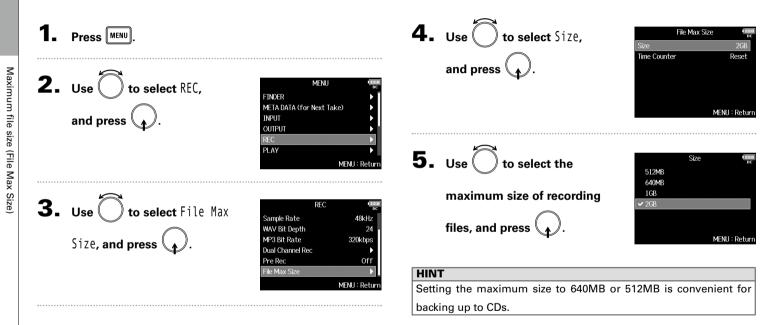
- When using dual channel recording, the track that is numbered 4 higher than the original track is used for the second recording. For example, track 5 is used for the dual channel recording of track 1 and track 6 is used for track 2. Dual channel recording tracks cannot be used independently.
- When dual channel recording is enabled, if stereo-linking is enabled or disabled for tracks 1/2 or 3/4, the same setting will be applied to tracks 5/6 or 7/8.
- The limiter, high pass filter and other functions can be set independently for the regular and dual recording tracks.
- When a mic capsule is connected, its dual recording track input level is fixed at –12 dB compared to the regular track.

Capturing audio before recording starts (Pre Rec)

The input signal can be captured for up to 6 seconds before • is pressed (pre-recording). This is useful if, for example, • is pressed too late. File format Sampling rate Maximum pre-recording time 44.1kHz 6 seconds Press MENU 47.952kHz 6 seconds 47.952kHz(F) 6 seconds 48kHz 6 seconds **2.** Use to select REC, and press . MENU WAV 48.048kHz 6 seconds FINDER 48.048kHz(F) 6 seconds META DATA (for Next Take) 88.2kHz INPUT 3 seconds OUTPUT 96kHz 3 seconds 192kHz 1 second PLAY 44.1kHz 6 seconds MENH: Return MP3 48kHz 6 seconds **3.** Use to select Pre Rec, and press . NOTE RFC Pre-recording will be disabled if MENU > TIMECODE > Timecode > Rec to SD2 Tr1-8 (Poly) Sample Rate 48kHz Mode (\rightarrow P.126) is set to Int Record Run, Ext or Ext Auto Rec. WAV Bit Depth MP3 Bit Rate 320kbps Dual Channel Rec Pre Rec MENU : Return 4. Use to select 0n, and press . Pre Rec Off ✓ On (6sec) MENU : Return

Maximum file size (File Max Size)

The maximum size of recording files can be set. If a recording file exceeds the maximum file size, recording will continue in a new take with a number that is one higher. No gap will occur in the sound between the two takes when this happens.



the counter shown on the Home Screen will be reset to

Showing total recording times for long recordings (Time Counter)

When recording for a long time, if the file size set with "File Max Size" is reached, recording will continue in a new take and the recording time will reset. You can change this, however, so that it is not reset and the total recording time is shown.

 1. Press MENU. 2. Use to select REC, and press . 	MENU (IIIIII) FINDER META DATA (for Next Take) INPUT OUTPUT REC PLAY MENU : Return	5. Use	File Max Size 2GB Size 2GB Time Counter Reset MENU - Return MENU - Return to select Continuous reset Reset
3. Use to select File Max Size, and press.	REC time Sample Rate 48kHz WAV Bit Depth 24 MP3 Bit Rate 320kbps Dual Channel Rec ▶ Pre Rec Off	continue	MENU : Return
	File Max Size	Setting value	Explanation
	MENU : Return		When recording, even if the file size set with "Size" is
		Continuous	reached, the counter shown on the Home Screen will not
			be reset.
			When recording, if the file size set with "Size" is reached,

Reset

000:00:00.

Folder and file structure

When recording with the **FB**, folders and files are created on SD cards as shown below.

Folders and files are used to manage scenes and takes.

Folder and file structure

تتك

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.

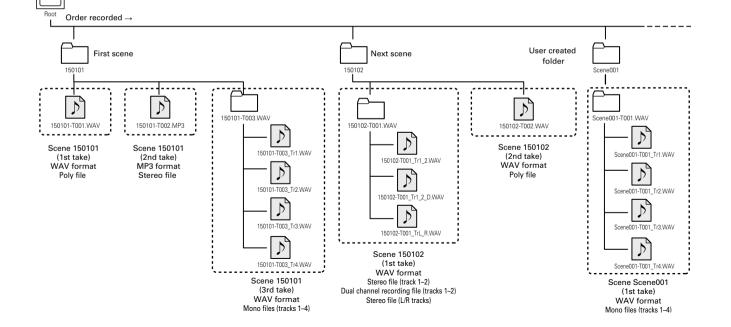
HINT

• A "take" is a unit of data created for a single recording.

 A "Scene" is a unit containing multiple files and takes that comprise a single scene.

NOTE

- Enabling recording on SD cards and setting file formats (→ P.25)
- Setting how scenes are named (mode) (→ P.43)



Take names

Structure	Explanation
Scene 001-T001 Take number (001-999) Scene number (1-9999) Scene name	Scene name: Select none, the folder name, the date or a name input by the user (\rightarrow P.42). Scene number: Press • + ++ to advance the number by one. Take number: This number increases by 1 for each recording made with the same scene name and number.

Audio file names

File names are given by the **FB** according to the file format—poly, mono or stereo. Track numbers and other data are added to file names.

File names

File names are given according to the following formats.

Туре	Structure	Explanation
Poly file	Scene001-T001.wav Take name	This is a file created by poly recording. Audio for multiple tracks is recorded to a single file.
Mono file	Scene001-T001_Tr1.wav Track number Take name	This is a file created by mono recording.
Stereo file	Scene001-T001_Tr1_2.wav Track number Take name	This is a file created by stereo recording.
Dual channel record- ing file	Scene001-T001_Tr1_D.wav Letter added to dual channel Take name Track number recording file	This is a file created by dual channel recording.

HINT

When recording with a Mono/Stereo setting, the audio files are saved in the take folder that is created.

Moving the previously recorded take to the FALSETAKE folder

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

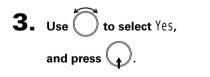
• Open the Home Screen.



2. Press and hold **—**.

HINT

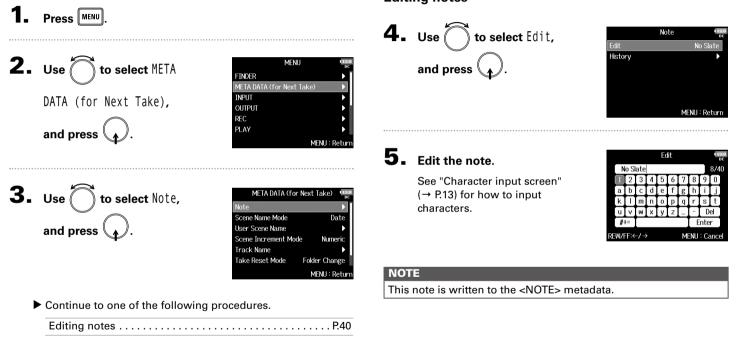
- Moving the take recorded most recently to the FALSETAKE folder
 - will reduce the number of the take recorded next by 1.
- Even during recording, you can move the previously recorded take to the FALSETAKE folder.





Changing the note for the next take recorded (Note)

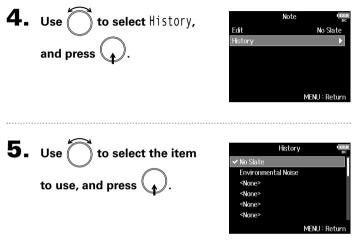
You can input characters for a note to use as metadata in the file.



Editing notes

Changing the note for the next take recorded (Note) (continued)

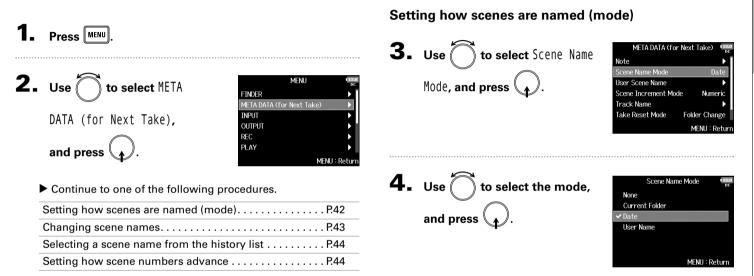
Selecting notes from the history list



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

Setting how recorded scenes are named and numbered

You can set how scenes are named (name mode), the base scene name and how scene numbers advance.



Setting how recorded scenes are named and numbered (continued)

Setting value	Explanation
	The scene name and number are not used.
	When recording files are created, they are named only
	with the take number, such as "T001", "T002", "T003" and
None	so on.
	■+ ➡ cannot be used to advance the scene number
	by 1.
	Example: T001.wav
	The name of the currently selected folder is used as the
	scene name.
	■+ ➡ can be used to advance the scene number
Current Folder	by 1. After advancing the scene number by 1, the cor-
Current i older	responding folder will be used as the recording des-
	tination. If that folder does not already exist, it will be
	created.
	Example: FOLDER001-T001.wav
	The date is used as the scene name.
	■+ ➡ cannot be used to advance the scene number
Date	by 1.
Date	If recording occurs after the date changes, a scene folder
	with the date will be created.
	Example: 20150101-T001.wav
	A scene name input by the user is used.
	+ + can be used to advance the scene number
User Name	by 1.
	No folder is created in this case.
	Example: MYSCENE001-T001.wav

Changing scene names

If Scene Name Mode is set to User Name, set the scene name used like this.

3. Use to select User Scene Name, and press . META DATA (for Next Take) Note Scene Name Mode Date User Scene Name Scene Increment Mode Numeric Track Name Take Reset Mode Folder Change MENII: Retur 4. Use to select Edit, and press . User Scene Name Scene00 History MENII: Retur **5.** Edit the scene name. Edit

See "Character input screen" (\rightarrow P.13) for how to input characters.



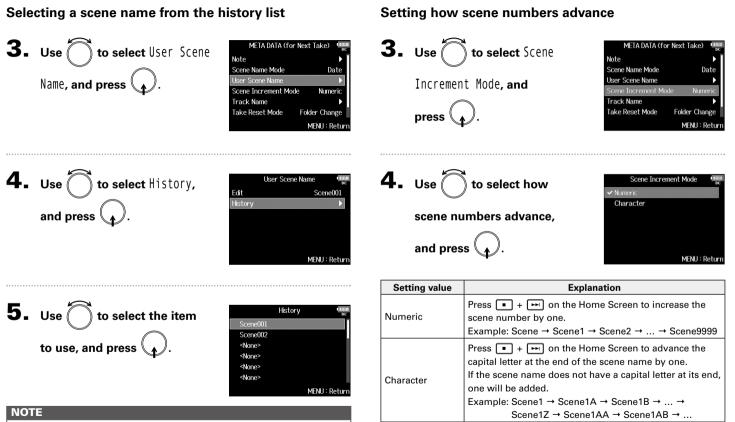
NOTE

The scene name is written to the <SCENE> metadata.

You cannot put a space or an @ mark at the beginning of the name.

Setting how recorded scenes

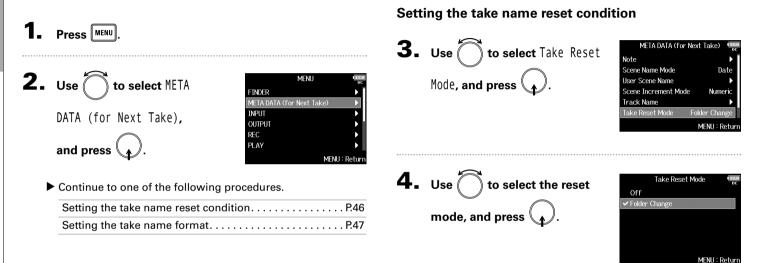
are named and numberec



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

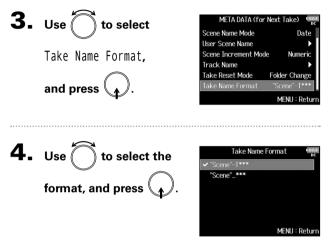
Setting the take name reset condition and format

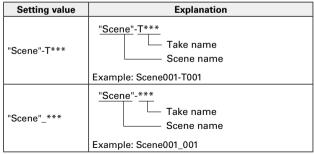
You can set the take name reset condition and format used when recording.



Setting value	Explanation
	The take number will not be reset.
	However, if the folder is changed and that folder con-
Off	tains a number higher than the current take number,
	the take number will be set to one higher than the
	highest existing take number.
	If the destination folder is changed, the take number
Folder Change	will be set to one higher than the highest take number
	in that folder.

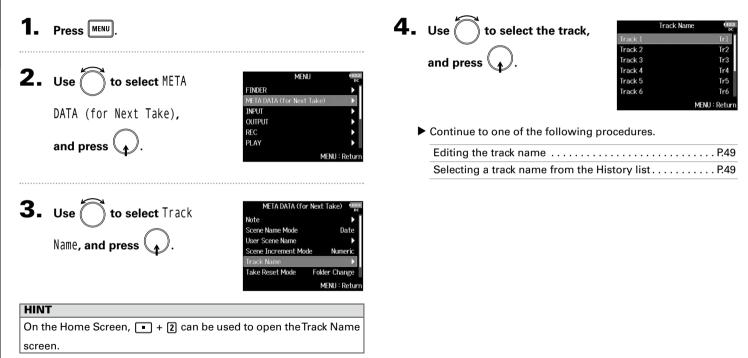
Setting the take name format





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.



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

Selecting a track name from the history list Editing the track name 5. Use to select Edit, and press . 5. Use to select History, and press . Track 1 Track 1 Edit History History MFNU: Return MFNU: Return 6. Use to select the item to use, and press . **6.** Edit the track name. History Edit Ir1 See "Character input screen" Mic Line $(\rightarrow P.13)$ for how to input Wireless characters. Mix Actor abc Enter MENU : Return $\mathbb{EW}/\mathbb{FF}: \leftarrow / \rightarrow$ MENU : Cance NOTE NOTE The history list will be erased if the Factory Reset function is used. The track name is written to the <TRACK> <NAME> metadata.

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.





decrease the take number

by one, and press (...).

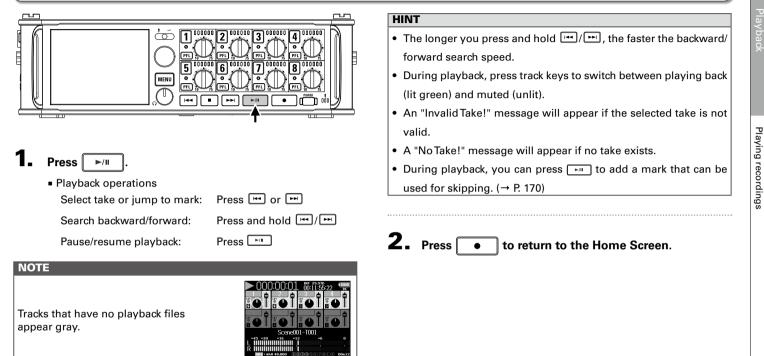


NOTE

This function cannot be used during recording and playback or when the scene naming method (Scene Name Mode) is set to Date. You can change how scenes are named with the following menu item.

MENU > META DATA (for NextTake) > Scene Name Mode

Playing recordings



Mixing takes

You can change the volume and panning of each track during playback.

Playback

• Open the mixer on the Home

Screen. (→ P.11)



2. Press **>**/**I** to start

playback.



HINT

- You can turn to move the cursor, and also adjust the settings of the MAIN OUT 1/2 and SUB OUT 1/2 tracks (→ P.116).
- When a fader or pan knob is selected, press and hold to reset it to its default value. If already set to its default value, selecting a fader mutes the track.

NOTE

- Settings are saved separately for each take and are used during playback.
- Mix settings are not saved with the take when the format is MP3.

3. Adjust the parameter settings.

See "Adjusting the input signal monitoring balance" (\rightarrow P.75) for how to change settings.

Monitoring the playback signals of specific tracks during playback

You can monitor the playback signals of specific tracks using SOLO mode.

1. Open the Home Screen.





playback.



3. Press **PFL** on the tracks that

you want to monitor.

The background colors for the selected tracks will become green, and their track indicators will light orange.



NOTE

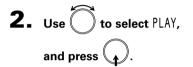
SOLO mode can only be used with tracks that can be played back (indicators lit green).

4. Press **PFL** of a track being monitored to stop

monitoring it.

Changing the playback mode (Play Mode)

You can change the playback mode.



MENU	12.0 DC
FINDER	
META DATA (for Next Take)	
INPUT	►
OUTPUT	
REC	
PLAY	Þ
MENU	: Returi

Setting value	Explanation	
Play One	Only the selected take will be played.	
(single playback)		
Play All	Takes will be played back continuously from	
(all playback)	the selected one until the last take.	
Repeat One	The selected take will be played repeatedly.	
(single repeat playback)		
Repeat All	All takes in the selected folder will be played	
(all repeat playback)	repeatedly.	

3. Use to select Play Mode, and press .

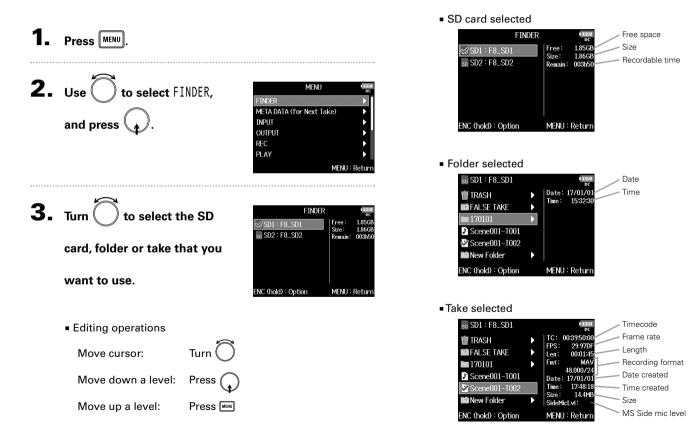




Play Mod	e 12.0
Play One	
🗸 Play All	
Repeat One	
Repeat All	
	MENU : Return

Take and folder operations (FINDER)

The FINDER allows you to select and view the contents of SD cards, takes and folders, and to create project/scene folders. It also allows you to, for example, set and delete recording/playback folders and view their information.



Take and folder operations (FINDER) (continued)

NOTE

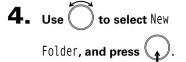
•	When the cursor is on a take, you can press Fin to play the
	selected take. You can also use 🖼, 폐 and 🔳.
•	A check mark appears on the playback take and recording/play-
	back folder.

Continue to one of the following procedures.

Creating folders	. P.56
Selecting the take recording/playback folder	. P.57
Checking take marks and using them for playback	. P.57
Changing folder and take names	. P.58
Copying takes to other cards and folders	. P.58
Deleting folders and takes	. P.59
Emptying the TRASH/FALSE TAKE folder	. P.60

Creating folders

Folders can be created inside the currently selected SD card/folder.





5. Edit the folder name.

See "Character input screen" (\rightarrow P.13) for how to input characters.



NOTE

- The folder created will be set as the recording folder.
- The name of the folder created is written to the <PROJECT> or <SCENE> metadata.
- You cannot put a space or an @ mark at the beginning of the name.

Selecting the take recording/playback folder

Use this procedure to select the folder that contains the take to be played or the folder to use for recording takes.



NOTE

- The first take inside the selected SD card or folder will be set as the playback take.
- After selecting the take recording/playback folder, the Home Screen will reopen.

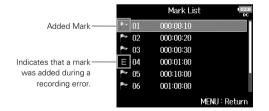
Checking take marks and using them for playback

You can view a list of the marks in a recorded take.



5. Use to select a mark, and press

The Home Screen will reopen, and playback will start from the mark.

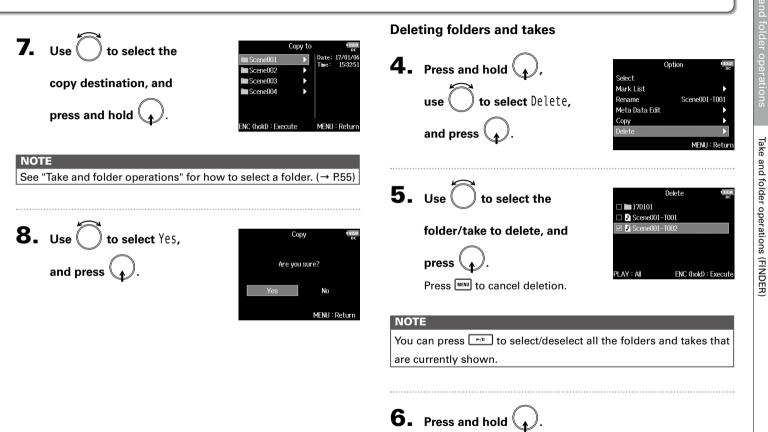


Take and folder operations (FINDER) (continued)

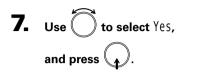
Changing folder and take names

4. Press and hold **4.** Press and hold (Option Option Select Select Mark List Mark List **to select** Copy, **to select** Rename, use Scene001-T00 use Rename Scene001-T001 Rename Meta Data Edit Meta Data Edit Сору and press and press Delete Delete MENU : Retur MENH : Return **5.** Use to select the take to **5.** Edit the folder/take name. Rename Copy 🗹 🎝 Scene001-T001 Scene002-T001 See "Character input screen" Scene001–T002 copy, and press Scene002–T001 $(\rightarrow P.13)$ for how to input Scene002-T002 characters. Scene003-T001 Scene003–T002 Enter MENII: Canci $FIM/FE: \leftarrow / \rightarrow$ PLAY: All ENC (hold) : Copy to NOTE • The edited name of the folder/take is written to the 6. Press and hold <PROJECT> or <SCENE> metadata. • You cannot put a space or an @ mark at the beginning of the name.

Copying takes to other cards and folders



Take and folder operations (FINDER) (continued)



Del	ete 📭
Are yo	u sure?
Yes	No
	MENU : Return

NOTE

- Deleted folders and takes are not immediately erased from the SD card. They are moved to the TRASH folder.
- Deleting the folders and takes in the TRASH folder will completely erase their data.

Emptying the TRASH/FALSE TAKE folder

4. Use to select TRASH or FALSE TAKE.

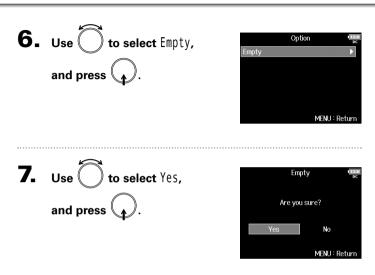


TRASH folder



FALSETAKE folder





NOTE

- Emptying the TRASH folder will completely erase the data in it.
- Emptying the FALSE TAKE folder will not immediately erase the data in it from the SD card. Instead, this data will be moved to the TRASH folder.

Take and folder operations

Overview of take information (metadata) stored in files

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

When these files are read by an application that supports metadata, you will be able to check and use the saved information.

HINT

- Metadata is data that contains information related to other data.
 The **S** saves scene names and take numbers, for example, as metadata in audio files.
- A chunk is a unit that contains multiple data in a single block.
- To use BEXT and iXML chunk metadata, an application that supports both data formats is necessary.

WAV file metadata

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

For information about the metadata saved in these chunks, see the "Metadata contained in BEXT chunks in WAV files" (\rightarrow P.186) and "Metadata contained in iXML chunks in WAV files" (\rightarrow P.187).

MP3 file metadata

The metadata saved in files recorded by the FB in MP3 format is written as ID3v1 tags.

For information about the ID3 fields and formats for saving metadata, see the "Metadata and ID3 fields contained in MP3 files" (\rightarrow P.189).

HINT

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

• MP3 metadata cannot be edited.

Overview of take information (metadata) stored in files

Press MENU.





TC: 00:39:50:0 FPS: 29.970

Len: 00:01 Fmt: W

Date: 17/01/0

Time: 17:48:1 Size: 14.4M

SideMicLvI

MENU : Return

48.0007

FALSE TAKE

Scene001–T001

Scene001-T002

ENC (hold) : Option

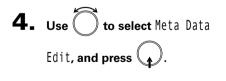
New Folder

170101

3. Use to select the take,

and press	•
-----------	---

This opens the Option Screen. See "Take and folder operations" for how to use the Finder. (\rightarrow P.55)

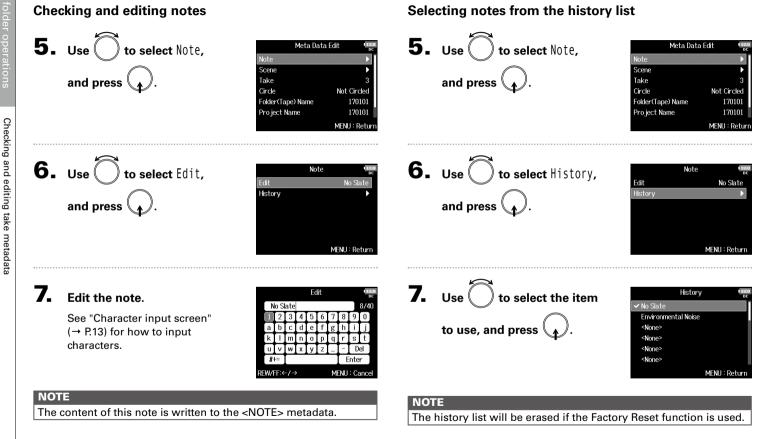


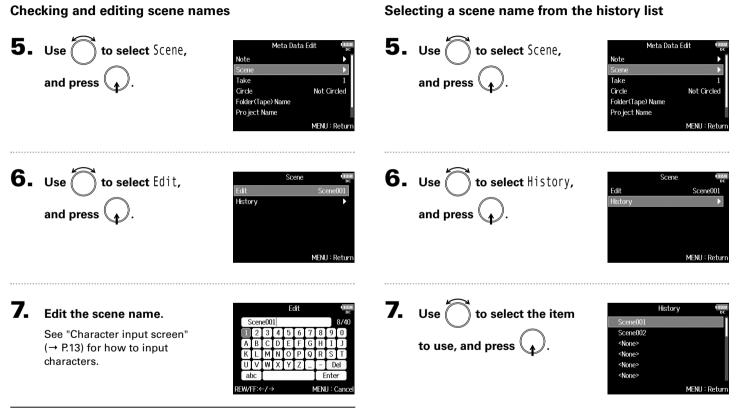
	Option	12.2 DC
Select		
Mark List		Þ
Rename	Scene0	01-T001
Meta Data Edit		•
Сору		•
Delete		•
	MEN	IU : Return

Continue to one of the following procedures.

Checking and editing notesP.64
Selecting notes from the history listP.64
Checking and editing scene names
Selecting a scene name from the history list $\ldots \ldots . P.65$
Checking and editing take namesP.66
Circling takes
Editing folder (tape) namesP.67
Editing project names P.68
Checking and editing track names
Selecting a track name from the History list

Checking and editing take metadata





NOTE

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

The scene name is written to the <SCENE> metadata.

Checking and editing take metadata (continued)

Note Scene

Circle

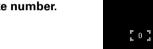
Folder(Tape) Name Pro ject Name

Checking and editing take metadata

6. Change the take number.

5. Use to select Take, and press .

Checking and editing take names





Meta Data Edit

Not Circled

MENU : Return

Editing operations

Move cursor or change value: Turn Select parameter to change: Press

HINT

This can be set from 1 to 999.

NOTE

The take number is written to the <TAKE> metadata.

7. When done changing,

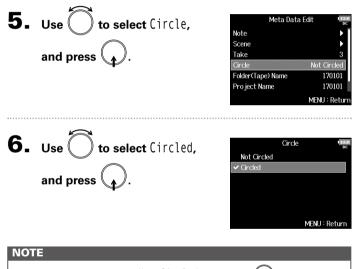
use \bigcirc to select Enter, and press \bigcirc .



Checking and editing take metadata

Circling takes

Use this function to add an @ mark to the beginning of the name of the best take to make it stand out. This is called a "circled take".



- To clear a circle, select Not Circled and press
- This circled status is written to the <CIRCLE> metadata.

Editing folder (tape) names

5. Use () to select Folder

(Tape) Name, and press .

Meta Data	a Edit 📫
Note	▶
Scene	Þ
Take	3
Circle	Not Circled
Folder(Tape) Name	170101
Project Name	170101
	MENII: Return

6. Edit the folder (tape) name.

See "Character input screen" $(\rightarrow P.13)$ for how to input characters.



NOTE

- The folder (tape) name is written to the <TAPE> metadata.
- The folder (tape) name used immediately after recording is the name of the folder in which the take was recorded.
- You cannot put a space or an @ mark at the beginning of the name.

Checking and editing take metadata (continued)

Checking and editing track names 5. Use to select Project 5. Use to select Track Name, and press . Meta Data Edit Meta Data Edit Note Scene Name, and press . Scene Take Take Circle Not Circled Circle Not Circleo Folder(Tape) Name Pro ject001 Project001 Folder(Tape) Name 170101 Pro ject Name Pro iect Name MENU : Retur MENU : Retur **6.** Edit the project name. Pro iect Name **6.** Use () to select the track, Track Name 170101 and press See "Character input screen" Track 2 Tr₂ $(\rightarrow P.13)$ for how to input Track 3 Tr3 characters. Track 4 Tr4 Track 5 ahr Enter Track 6 $FW/FE: \leftarrow / \rightarrow$ MENU : Cano MFNU: Return NOTE • The project name is written to the <PROJECT> metadata. 7. Use () to select Edit, • The project name used immediately after recording includes the Track 1 name of the highest level folder (inside the SD card root direcand press History tory) that contains the folder in which the take was recorded. • You cannot put a space or an @ mark at the beginning of the name. MENU : Returi

Selecting a track name from the history list Edit Tr1 5. Use to select Track Name, and press . Meta Data Edit Scene Take Circle Not Circled ahr Enter Folder(Tape) Name Pro ject001 $FW/FF: \leftarrow / \rightarrow$ MENU: Canc Project Name Project001 Checking and editing take metadata MENU : Return 6. Use to select the track, and press . Track Name Track 2 Tr₂ Track 3 Tr3 Track 4 Tr/ Track 5 Track 6 MFNU: Return 7. Use to select History, and press . Track 1 Edit Tr1 MENU : Return

8. Edit the track name.

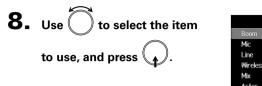
See "Character input screen" $(\rightarrow P.13)$ for how to input characters.

NOTE

The track name is written to the <TRACK> <NAME> metadata.

69

Checking and editing take metadata (continued)

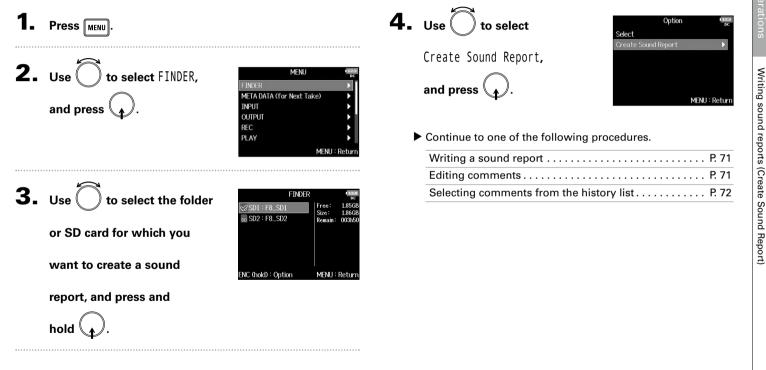


History		12.2 DC
Boom		
Mic		
Line		
Wireless		
Mix		
Actor		
		MENU : Return

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

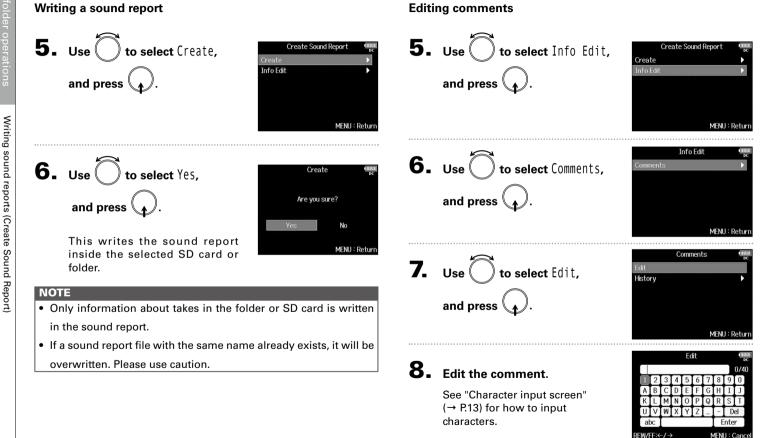
Writing sound reports (Create Sound Report)

A sound report includes information about recording times and takes. Reports can be written as CSV format files (F8_[folder name].CSV). You can edit the comments written in sound reports.



ake and tolder operatio

Writing sound reports (Create Sound Report) (continued)

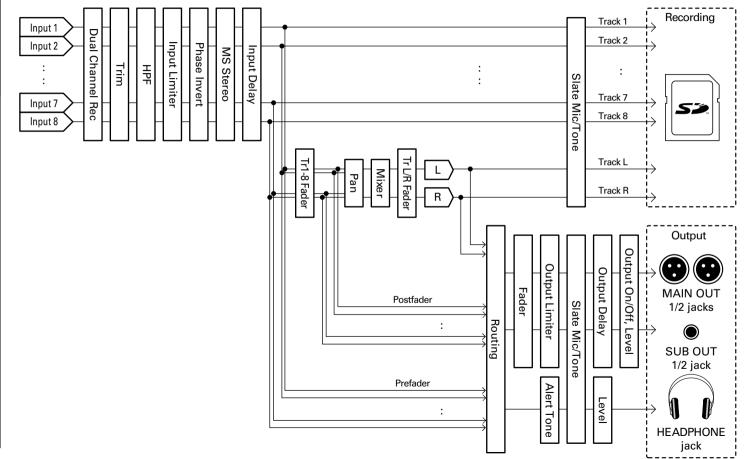


Writing sound reports (Create Sound Report)

Selecting comments from the history list

5. Use to select Info Edit, and press .	Create Sound Report ©Create ☐Info Edit MENU : Return	8. Use to select the item to use, and press .
6. Use to select Comments, and press .	Info Edit unan cc Comments ► MENU : Return	HINT The history list will be erased if the Factory Reset function is used.
7. Use to select History, and press .	Comments Edit History MENU : Return	

Input and output signal flow



Input and output signal flow

Adjusting the input signal monitoring balance

You can adjust the volume and panning of each input signal when monitoring during recording.

• Open the mixer on the Home

Screen. (\rightarrow P.11)



2. Adjust the parameter settings.

Editing operations

Move cursor or change value: Tur

Select parameter to change:

Parameter	Setting range	Explanation
Fader	Mute, –48.0 – +12.0 dB	Adjusts the level of the input
Fauer	Wule, -40.0 - +12.0 ub	signal.
Donning	L100 – Center – R100	Adjusts the left-right stereo
Panning	L 100 – Center – R 100	position of the sound.

- You can turn O to move the cursor, and also adjust the settings of the MAIN OUT 1/2 and SUB OUT 1/2 signals. (→ P.116)
- When a fader or pan knob is selected, press and hold to reset it to its default value. If already set to its default value, selecting a fader mutes the track.

NOTE

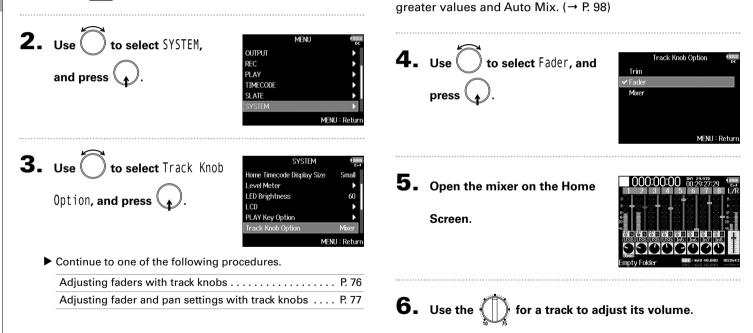
- The MAIN OUT 1/2 and SUB OUT 1/2 faders do not affect the levels of the slate mic and slate tone.
- These volume and pan settings only affect the monitoring signals and the data being recorded on the L/R track.
- Settings are saved separately for each take that is already recorded and can be changed during playback. (→ P.52)
- Mix settings are not saved with the take when the recorded file format is MP3.

Setting the track knob function (Track Knob Option)

The Home Screen layout and track knob functions can be changed.

Setting the track knob function

1. Press MENU.



Adjusting faders with track knobs

This restricts the track knob function to changing fader values.

This mode makes it easy to check volume changes due to

put settings

Setting the track knob function (Track Knob Option)

Setting the track knob function (Track Knob Option)

HINT

The TRIM and L/R figure values can be changed as follows.

Move cursor, change setting value: Turn

Select parameter to change: Press

Adjusting trim, fader and pan settings with track knobs

The track knobs can be used to quickly adjust the fader and pan settings of each track.

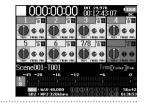


5. Open the mixer on the Home

Screen.







7. Use f the track you want to adjust to change

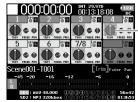
its setting value.

HINT

The position of the knob on the display always shows the current setting.

NOTE

After changing the parameter be adjusted, for example, if the positions of () and the knob on the display are different, the knob on the display will appear gray, and moving () will not affect that setting. In this case, if you adjust () to match the position of the knob on the display, the display knob and () will be relinked, and you will be able to use () to adjust its setting value again.



Parameters with setting values and positions that are different

Adjusting the L/R track volume

1. Open the Home Screen.



4. When finished adjusting, press MENU or \bullet + **6**.

NOTE

This is only enabled when Track Knob Option is not set to Fader.

When set to Fader, you can adjust by using () to select.

2. Press • + 6.

NOTE Shortcuts are disabled during playback.	
3. Use to adjust the volume.	0000:000:00 Mit 284780 40.0 00:13:33:21 500 500:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21 500 00:13:33:21
NOTE	

- Volume settings affect the results of recording.
- If only the L/R track is recorded, the L/R track fader setting for the take will be saved as 0dB.

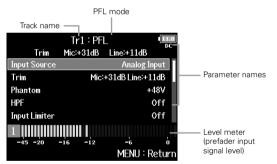
Monitoring the input signals of specific tracks (PFL/SOLO)

You can monitor the input signals of specified tracks. 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. You can also make various settings for these tracks.

1. Press **PFL** on the tracks that you want to monitor.

The selected track keys will light orange, and the $\ensuremath{\mathsf{PFL}}$ screen will open.

"PFL" or "SOLO" appears at the top of the display, and you will be able to monitor the input signal with headphones.



Parameter	Explanation
Input Source	This sets the input source.
Trim	This sets the input level.
Phantom	This sets phantom power.
HPF	This sets the high pass filter.
Input Limiter	This sets the limiter.
Fader	This sets the fader level.
Pan	This sets the panning.

Parameter	Explanation
Phase Invert	This sets the phase.
Side Mic Level	This sets the side mic level of a mid-side mic capsule.
Input Delay	This sets the input delay.
Plugin Power	This sets the plugin power.
Stereo Link	This sets the stereo link.
Stereo Link Mode	This sets the stereo link mode.
PFL Mode	This sets the monitoring volume on the PFL screen.

NOTE

This does not change the signals output from MAIN OUT and SUB OUT.

HINT

Use $\left(\right)$ to select parameters and change setting values.

2. Press **PFL** or **MENU** for the monitored tracks.

Open the Home Screen.

Setting the input source (Input Source)

Follow these procedures to set the input source of each track.

1. Press MENU.



MENU	12.2 DC
INDER	- ▶ [
META DATA (for Next Take)	- ▶
NPUT	Þ
DUTPUT	
REC	
PLAY	
MENU :	Return

3. Use to select Input Source, and press .

	INPUT	12.2 DC
Input Source		
Ambisonic Mode		
HPF		
Input Limiter		
Phase Invert		
Phantom		
	Menu : R	eturn

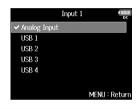
4. Use to select a track, and

	Input Source
Input 1	Analog Input
Input 2	Analog Input
Input 3	Analog Input
Input 4	Analog Input
Input 5	Analog Input
Input 6	Analog Input
	MENU : Return

HINT

Select ALL to set all the tracks at the same time.

5. Use to select the input source, and press .



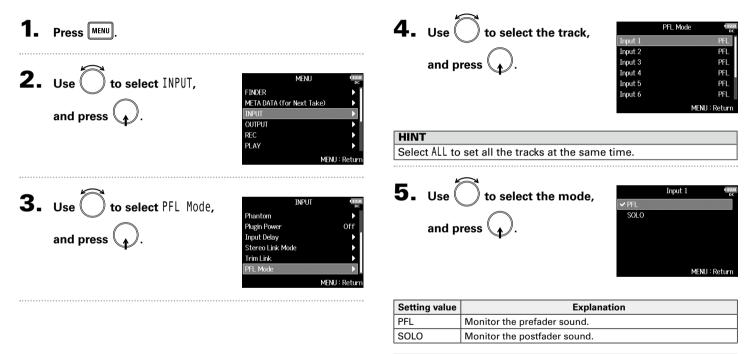
Setting value	Explanation
Analog Input	This treats signals input through INPUT 1–8 as input signals.
USB 1, USB 2, USB 3, USB 4	When Audio Interface with Rec (\rightarrow P.147) is set to On, computer output signals are treated as input signals.

NOTE

- When a mic capsule is connected, the Input Source cannot bechanged for Inputs 1 and 2.
- When dual channel recording is enabled (\rightarrow P.33), the Input Source cannot be changed for the dual channel recording tracks.

Setting the monitoring volume on the PFL screen (PFL Mode)

On the PFL screen, you can set the monitored sound to be either prefader listening (PFL) or postfader solo (SOLO).

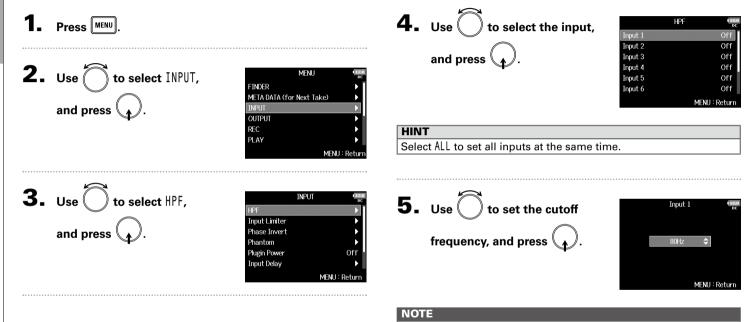


NOTE

When the PFL screen is open during playback, the monitoring sound will be post-fader (SOLO) regardless of the setting.

Cutting low-frequency noise (HPF)

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



The HPF also affects dual channel recording data.

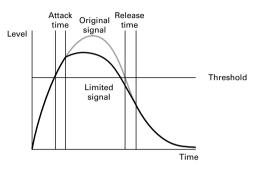
HINT

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

Input limiter

Input limiter

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



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

The attack time is how long after the signal exceeds the threshold before the limiter starts operating. The release time is how long after the signal goes below the threshold before the limiter stops operating. You can change these two parameters to adjust the sound quality.

HINT

- The FB has a newly designed limiter that provides 10 dB of headroom, allowing signals to be kept well below the set threshold, therefore more effectively preventing distortion.
- The **B** limiter uses a ratio of 20:1.

Press MENU **2.** Use () to select INPUT, MENU FINDER META DATA (for Next Take) and press OUTPUT REC PLAY MENU : Retur 3. Use to select Input INPUT Limiter, and press Input Limiter Phase Invert Phantom Plugin Power Off input Delav MENU : Return **4.** Use () to select the input, Input Limiter Input 1 Input 2 and press Input 3 Input 4 Input 5 Input 6

HINT

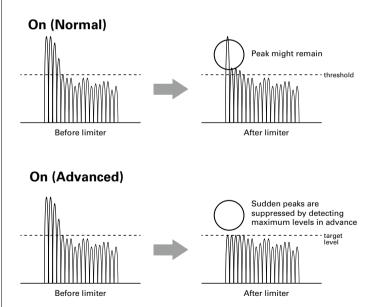
Select ALL to set all the inputs at the same time.

MENII: Return

Input limiter (continued)

Continue to one of the following procedures.

Using the limiter	P.84
Setting the type	P.85
Setting the threshold	P.86
Setting the attack time	P.86
Setting the release time	P.87
Setting the target level	P.88

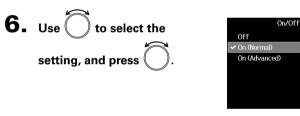


Using the limiter



	Input 1 🔍	1
On/Off	Off	1
Туре	Hard Knee	
	MENU : Retu	Jr

MENU : Return



Setting value	Explanation
Off	This disables the limiter.
On (Normal)	This applies an ordinary limiter. The ratio is 20:1.
On (Advanced)	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 **FB** increases 1 ms. When monitoring sounds being recorded with a mic in real-time, increased latency can cause interference between sound transmitted through the air from the source and the delayed monitored sound, possibly making accurate monitoring of the sounds difficult.

NOTE

- When set to On (Advanced), the following limiter parameter becomes available to set.
 - -Target Level
- When set to On (Advanced), the following limiter parameters cannot be set.
 - -Type
 - -Threshold
 - Attack Time
 - Release Time
- 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

5. Use to select Type, and press .

6. Use to select the type, and press .





Setting value	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.

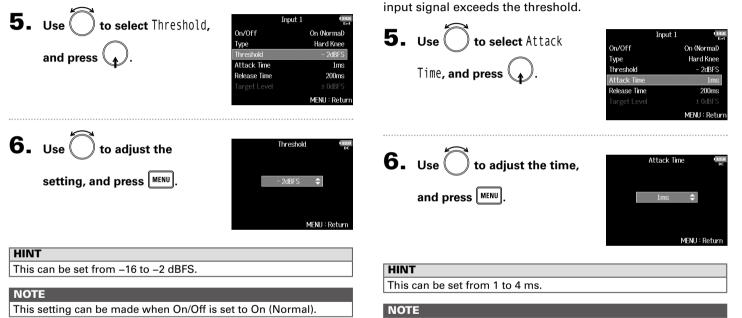
NOTE

This setting can be made when On/Off is set to On (Normal).

Input limiter (continued)

Setting the threshold

This sets the level at which the limiter begins operating.



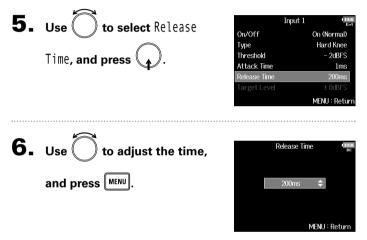
Setting the attack time

This setting can be made when On/Off is set to On (Normal).

This sets the amount of time until limiting starts after the

Setting the release time

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



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.
- When the limiter is operating, the right-most segment of the level meter and the mixer limiter indicator on the display light yellow.

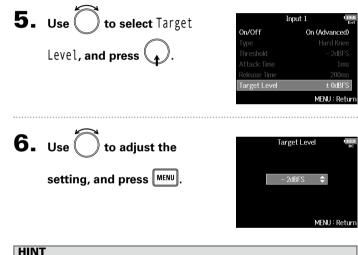
NOTE

This setting can be made when On/Off is set to On (Normal).

Input limiter (continued)

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.



- 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.

NOTE

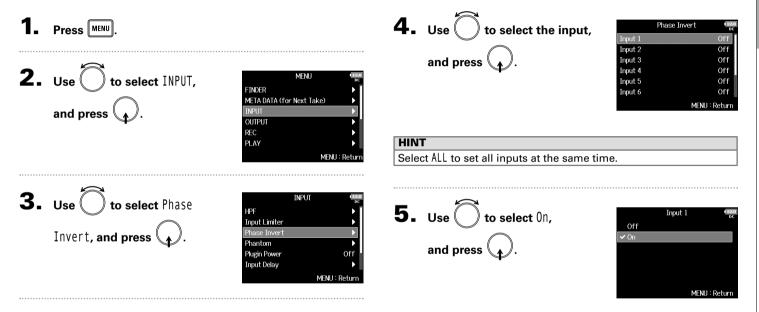
This setting becomes available when On/Off is set to On (Advanced).

Input limiter

Inverting the input phase (Phase Invert)

The phase of the input signal can be inverted.

This is useful when sounds cancel each other out due to mic positioning.



Changing the phantom power settings (Phantom)

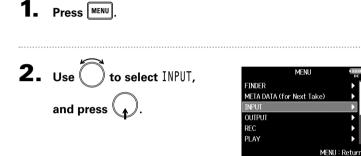
The **FB** can provide phantom power. The voltage can be set to +24V or +48V and it can be turned on/off for each input separately.

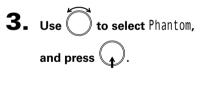
HINT

Phantom power is a function that supplies power to devices that require an external power supply, including some condenser mics. The standard power is +48V, but some devices can operate with lower voltages.

NOTE

Do not use this function with a device that is not compatible with phantom power. Doing so could damage the device.

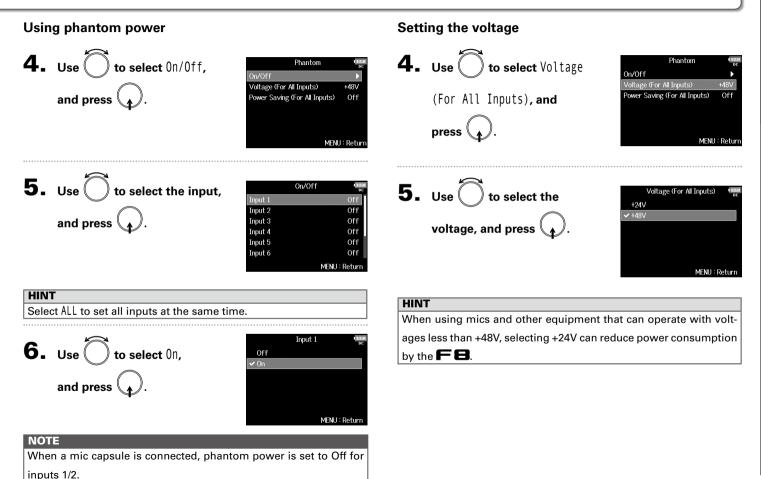






Continue to one of the following procedures.

Using phantom powerP.	
Setting the voltage	P.91
Disabling phantom power during playback	P.92



Changing the phantom power settings (Phantom) (continued)

Disabling phantom power during playback

Setting value	Explanation
Off	Phantom power is supplied even during playback.
	Phantom power is not supplied during playback. This can reduce the F B power consumption.

HINT

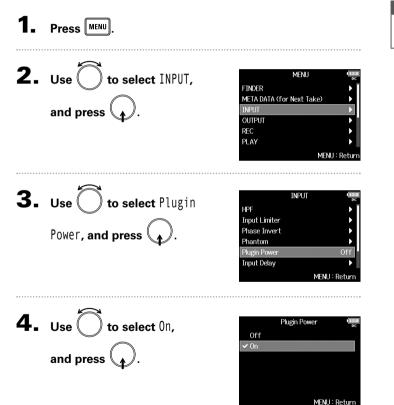
If mics do not need phantom power during playback, disabling it can reduce

NOTE

This setting affects all inputs.

Changing the plugin power setting (Plugin Power)

Make this setting when a mic that is compatible with plug-in power is connected to the mic capsule's MIC/LINE input jack.

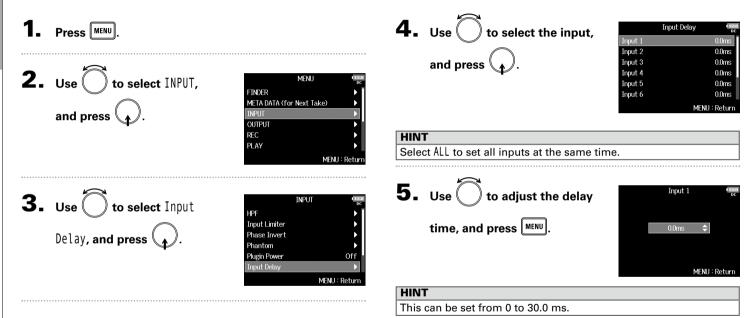


NOTE

This setting can be changed only when a mic capsule that supports plug-in power is connected.

Delaying input signals (Input Delay)

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



NOTE

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

Converting mid-side input to ordinary stereo (Stereo Link Mode)

Signals from a mid-side stereo mic input connected to stereo-linked inputs can be converted to an ordinary stereo signal. See "Linking inputs as a stereo pair" (\rightarrow P.27) for how to use stereo linking.

Mid-side stereo format overview

This technique creates a stereo recording from signals input by a directional mid mic that captures sound in the center and a bidirectional side mic that captures sounds from the left and right. Mid-side recording allows you to change the stereo width by the adjusting the level of the side mic.

Since this technique can capture a wide stereo image, it is ideal for recording open spaces with numerous sound sources, such as orchestras, live concerts and soundscapes.

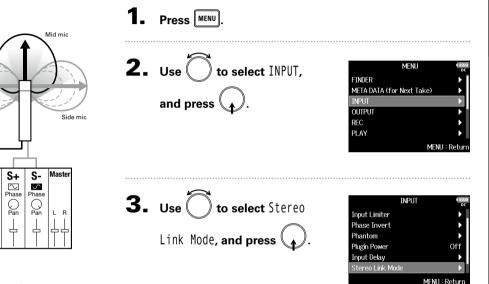
Mid-side recording is also extremely effective when you want to be able to control the amount of room ambience in a signal. For this reason, it is often used for live as well as studio recording. In addition, the stereo signal created by this technique is fully mono-compatible, making it especially useful when recording sound for film, video, or broadcast.

M

Phase

Dan

¢



Converting mid-side input to ordinary stereo (Stereo Link Mode) (continued)

4. Use to select the input pair, and press .

	Stereo Link Mode
Input 1/2	Stereo Link
Input 3/4	Stereo Link
Input 5/6	Stereo Link
Input 7/8	Stereo Link
All	•
	MENU : Return



5.	Use 问 to select MS Stereo	Stereo
	Link, and press 😱.	✓ MS Ste

Input 1/2	12.0 DC
Stereo Link	
✔ MS Stereo Link	
	MENU : Return

Setting value	Explanation
Stereo Link	When stereo-linked, inputs are handled normally.
MS Stereo Link	When stereo-linked, signals from a mid-side mic are con-
	verted to ordinary stereo.

NOTE

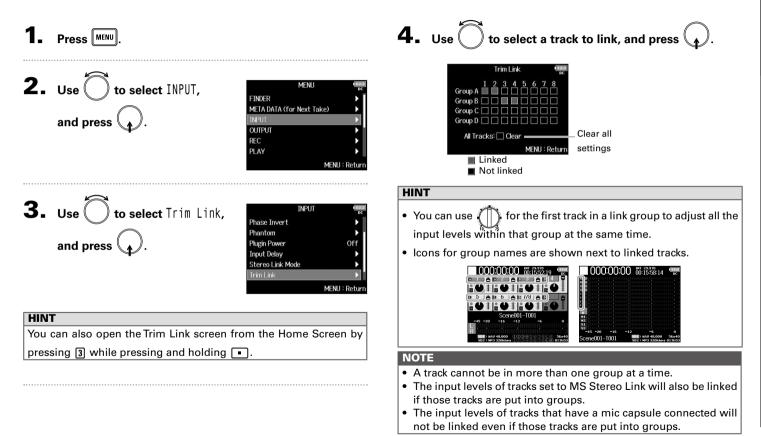
- When MS Stereo Link is selected, odd-numbered inputs are handled as mid signals and even-numbered inputs as side signals.
- The MS Stereo Link setting is disabled if a ZOOM mic capsule is connected that cannot have L/R signals routed individually to inputs 1/2.

HINT

- Use () for each input to adjust the mid/side balance.
- The PFL screen allows you to adjust the side mic level for inputs 1/2 when a mid-side mic capsule is connected.

Adjusting the input levels of multiple tracks simultaneously (Trim Link)

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



Adjusting the side level of a mid-side mic capsule (Side Mic Level)

You can adjust the side mic level (stereo width) before recording when a mid-side mic capsule is connected.

Use to select Side Mic Level, and press to. Use to adjust the side

mic level, and press MENU.

Press [PFL] for track 1 or 2.



HINT This can be set to Off, RAW or in a range from –24 to +6 dB.

NOTE

- The more the side mic level is increased, the greater the stereo width.
- When set to RAW, recording will occur without stereo encoding. The stereo width of audio in RAW format can be adjusted after recording by using ZOOM MS Decoder or other plug-in software.
- This can be adjusted only when a mid-side type mic capsule isconnected.

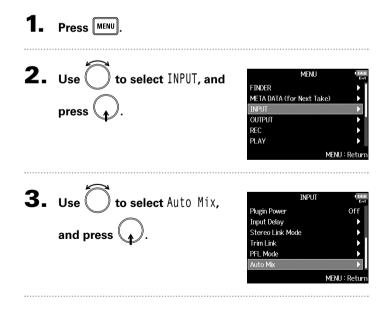
HINT

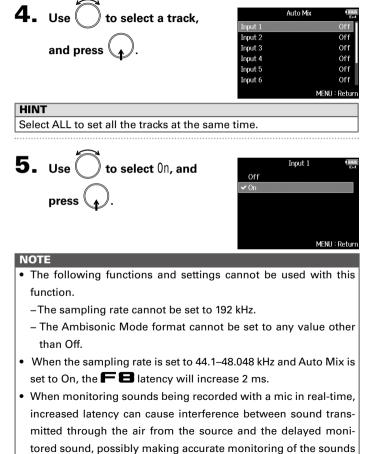
When dual channel recording is on, the side mic level can also be set for tracks 5/6, which correspond to tracks 1/2.

Changing the automatic mixing setting (Auto Mix)

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 using mics.
- Sound quality degradation due to phase differences caused by variations in the distances of multiple mics is reduced.

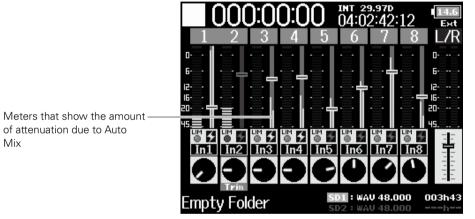




difficult.

Changing the automatic mixing setting (Auto Mix)

When Track Knob Option is set to Fader and Auto Mix is enabled, the Home Screen will appear as follows.



<Track Knob Option: Fader>

Mix

Setting the format of Ambisonic Mode

By connecting mics that can output ambisonics A-format signals to Inputs 1–4, audio can be converted to ambisonics B-format and recorded.



Changing the automatic mixing setting (Auto Mix) (continued)

FuMa

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

AmbiX

Input

1-4

Format: FuMa, AmbiX

Trim

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

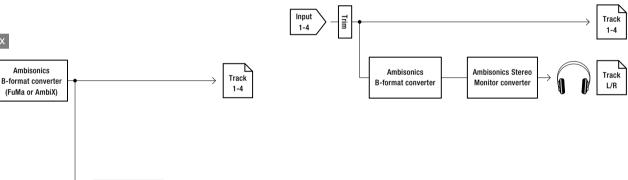
Ambisonics Stereo

Monitor converter

Ambisonics A (Stereo Monitor)

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

Format: Ambisonics A (Stereo Monitor)



Track

L/R

FuMa(Dual)

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

AmbiX(Dual)

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

FuMa + AmbiX

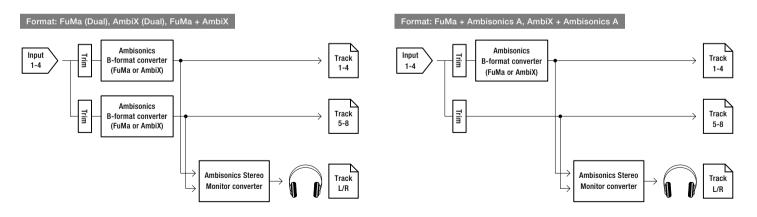
This converts the signals from Inputs 1-4 to the ambisonics FuMa B-format, and records them to tracks 1-4. It also converts the signals from Inputs 1-4 to the ambisonics AmbiX B-format, and records them to tracks 5-8. These can be recorded at different input levels.

FuMa + Ambisonics A

This converts the signals from Inputs 1-4 to the ambisonics FuMa B-format, and records them to tracks 1-4. It also records the signals from Inputs 1-4 to tracks 5-8 without converting them to an ambisonics B-format. These can be recorded at different input levels.

AmbiX + Ambisonics A

This converts the signals from Inputs 1-4 to the ambisonics AmbiX B-format, and records them to tracks 1-4. It also records the signals from Inputs 1-4 to tracks 5-8 without converting them to an ambisonics B-format. These can be recorded at different input levels.



iput settings

Changing the automatic mixing setting (Auto Mix) (continued)

NOTE

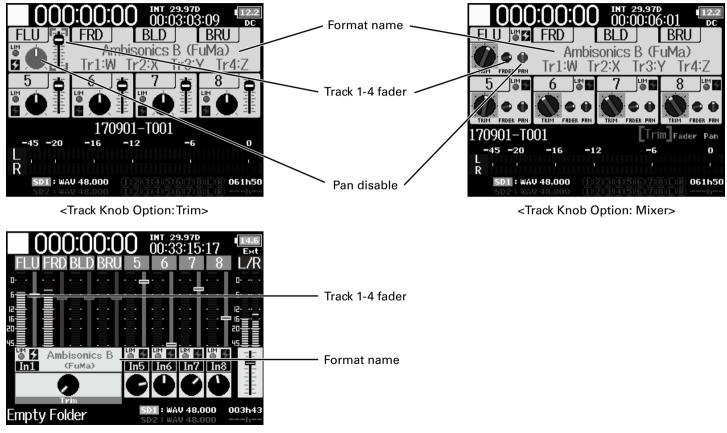
- The sampling rate can only be set to 192 kHz when the format of Ambisonic Mode is Off.
- Ambisonic files are saved as 4-channel polyphonic files, not as mono or stereo files.
- A ZOOM mic capsule can only be used when the format of Ambisonic Mode is Off.
- The following parameters cannot be set for tracks using Ambisonic Mode input.
 - Pan
 - Phase Invert
 - Side Mic Level
 - Input Delay
 - Stereo Link
 - Stereo Link Mode
 - Dual Channel Rec
 - Trim Link
- Files recorded when the format of Ambisonic Mode 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.
- When the sampling rate is set to 44.1–48.048 kHz and Ambisonic Mode is not set to Off, the **FB** latency will increase 2 ms. When monitoring sounds being recorded with a mic in real-time, increased latency can cause interference between sound transmitted through the air from the source and the delayed monitored sound, possibly making accurate monitoring of the sounds difficult.
- This cannot be used with the Auto Mix function.

HINT

- Ambisonic Mode can be set during use as an audio interface (MultiTrack).
- Even when the format of Ambisonic Mode is Off, you can press track **PFL** to monitor their input sounds. When PFL mode is set to PFL, you can monitor sounds before they are converted to ambisonics B-format. When PFL mode is set to SOLO, you can monitor sounds after they are converted to ambisonics B-format.
- The input levels of the Ambisonic Mode input tracks are linked, so for Input 1 (or Input 5) can be used to adjust their input levels.
- The input enabled/disabled statuses of the Ambisonic Mode input tracks are linked, so the settings of all assigned tracks can be switched simultaneously by pressing any of their track keys.
- The following parameters that can be set on the PFL screen are linked for input tracks using Ambisonic Mode.
 - HPF
 - Input Limiter
 - Phantom
 - Fader
 - PFL Mode
 - Input Source
 - Input Level

Changing the automatic mixing setting (Auto Mix)

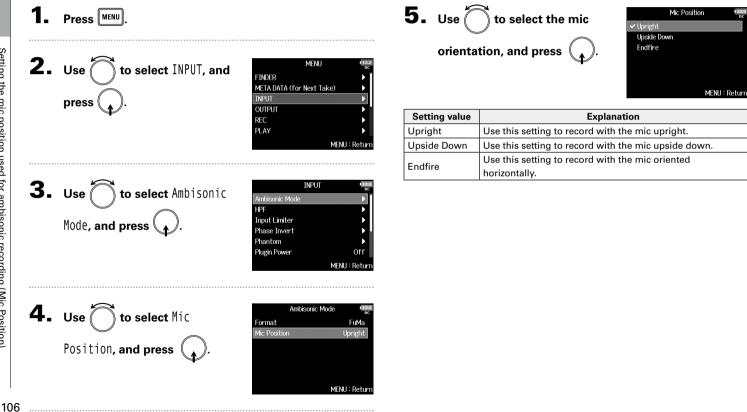
When Ambisonic Mode is enabled, the Home Screen will appear as follows.

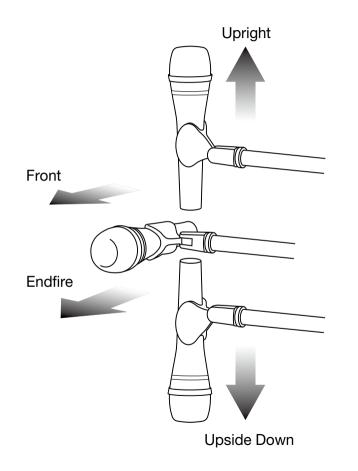


<Track Knob Option: Fader>

Setting the mic position used for ambisonic recording (Mic Position)

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





HINT

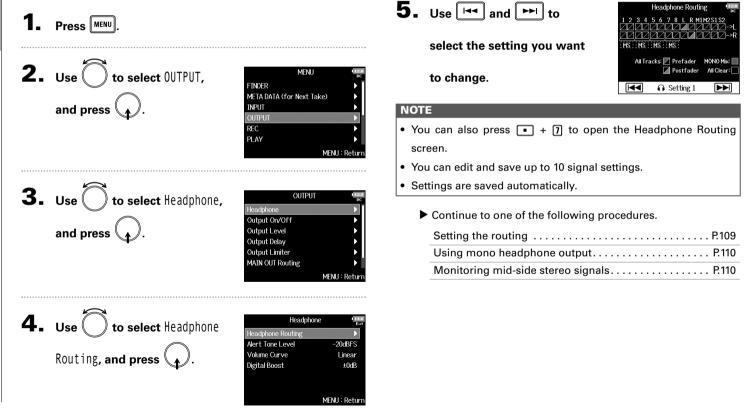
- Using the mic upright is recommended for ambisonic recording in order to minimize reflections from the floor and the mic itself.
- When it is difficult to use the mic in an upright orientation, you can place it upside down or pointing forward and change the Mic Position setting accordingly.

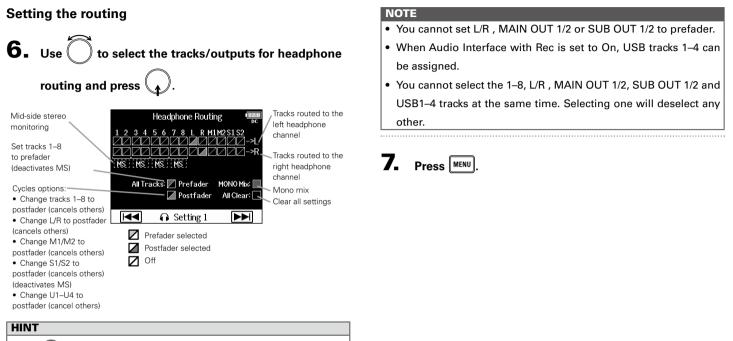
NOTE

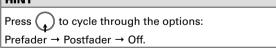
• If this setting and the mic position do not match, sound positioning will not be properly re-created during conversion to ambisonic B format.

Setting signals sent to headphones (Headphone Routing)

You can set the type of signal sent to the headphone output to either prefader or postfader for each track. You can also save 10 setting combinations (Setting 1 – Setting 10).

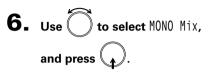






Setting signals sent to headphones (Headphone Routing) (continued)

Using mono headphone output





7. Press MENU

Monitoring mid-side stereo signals

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





7. Press MENU.

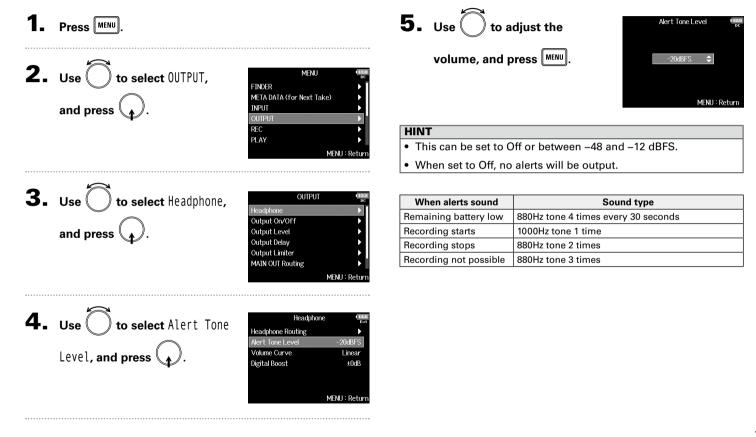
NOTE

- This is disabled for stereo-linked tracks that have Stereo Link Mode set to MS Stereo Link.
- This is only enabled for tracks that have a mid-side microphone or mid-side mic capsule connected and the Side Mic Level set to RAW.
- When mid-side stereo monitoring is enabled, the prefader tracks will be routed automatically to the headphone channels, with the odd-numbered to the left and the even-numbered to the right. In this case, the routing cannot be changed manually.

Outputting alerts through headphones (Alert Tone Level)

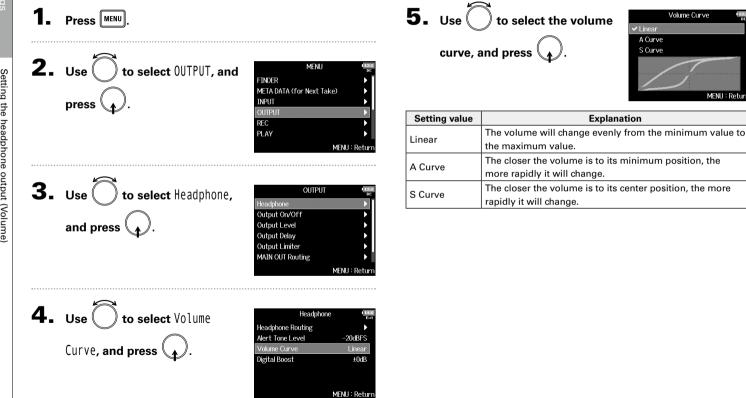
Outputting alerts through headphones (Alert Tone Level)

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



Setting the headphone output (Volume)

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



Volume Curve

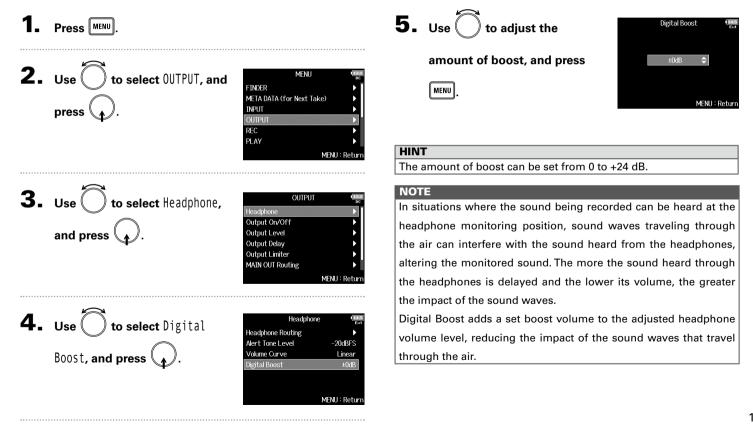
MENII: Retur

✓ Linear A Curve

S Curve

Boosting headphone output to alleviate interference from recorded sound (Digital Boost)

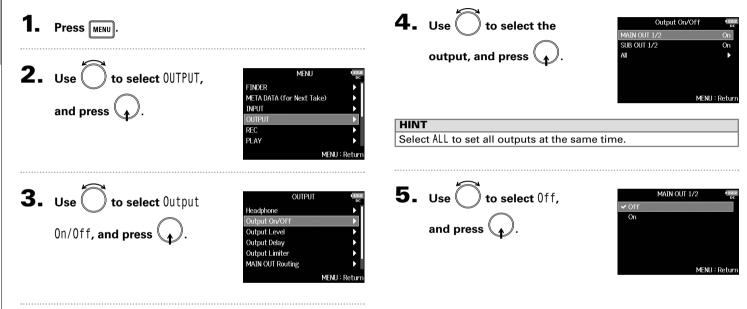
Boosting the headphone output alleviates the interference of sound waves traveling through the air with the headphone monitoring signal, enabling more accurate monitoring of the sound being recorded.



Boosting headphone output to alleviate interference from recorded sound (Digital Boost)

Disabling outputs (Output On/Off)

By disabling outputs that you are not using, you can reduce power consumption and increase the length of operation time when using batteries.



Setting the standard output level (Output Level)

The standard output level can be changed.

1. Press MENU.	4. Use	to set the output	Output Type Image pc MAIN OUT 1/2 Normal (-10dBV) SUB OUT 1/2 Normal (-10dBV)
2. Use to select OUTPUT, and press .	type, and	I press (p).	All → MENU : Return
3. Use to select Output Level, and press .	OUTPUT	to set the standard evel, and press .	MAIN OUT 1/2
	Setting value	Expla	nation

Normal (-10dBV)

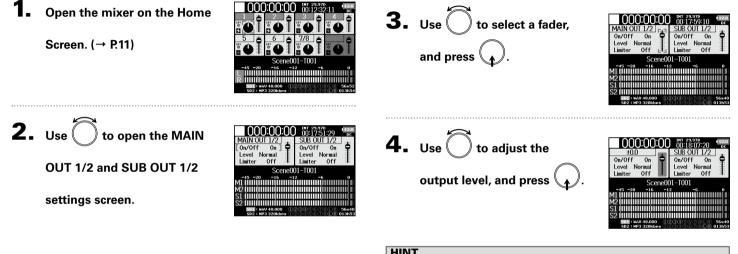
Mic (-40dBV)

This sets the standard level to -10 dBV.

This sets the standard level to -40 dBV.

Setting output levels

The MAIN OUT 1/2 and SUB OUT 1/2 levels can be changed.



HINT

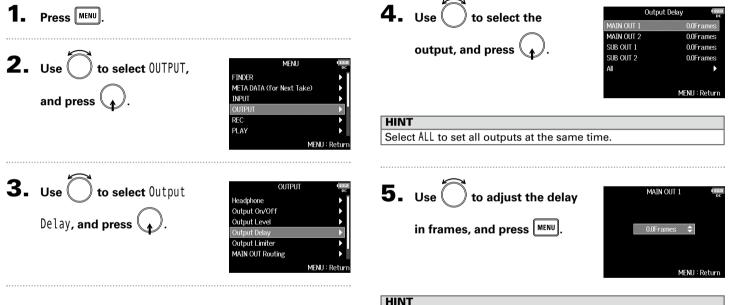
- This can be set to Mute or from -48.0 to +12.0 dB.
- · You can also check and adjust various output settings on the

MAIN OUT and SUB OUT setting screen.

Delaying output signals (Output Delay

Delaying output signals (Output Delay)

By delaying audio outputs, you can correct timing differences for signals input to other devices from the **FB**.



This can be set from 0.0 to 10.0 frames.

NOTE

- The delay in milliseconds depends on the frame rate of the selected timecode.
- When Sample Rate is set to 192kHz, Output Delay is disabled.

Output Limiter

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

HINT HINT For details about the effect of the limiter, see "Input limiter". (\rightarrow P.83) 1. Press MENU. **2.** Use to select OUTPUT, MENU FINDER and press META DATA (for Next Take) INPUT REC PLAY Using the limiter **3.** Use to select Output Limiter, and press . OUTPUT Headphone and press (Output On/Off Output Level Output Delay Output Limiter MAIN OUT Routing MENU : Retur 6. Use to select On, and press . **4.** Use () to select the **Output Limiter** MAIN OUT 1/2 output, and press SUB OUT 1/2 Al MENU : Return

Select ALL to set all outputs at the same time.

Continue to one of the following procedures.

Using the limiter	P.118
Setting the type	
Setting the threshold	P.119
Setting the attack time	
Setting the release time	P.120
Setting links	P.121





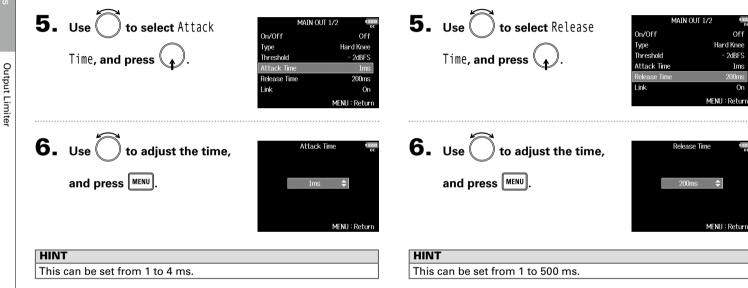


and pr	to select Type, on/Off Off Type Hard Knee Threshold - 2dBFS Attack Time Ims Release Time 200ms Link On MENU : Return	Setting the threshold This sets the level at which the limiter begins operating. 5. Use to select Threshold, and press to select Threshold, $\frac{MAIN OUT 1/2 (model)}{1000 for the select} = \frac{MAIN OUT 1/2}{1000 for the select} = \frac{MAIN OUT 1/2}{100$
6. Use (and pr	to select the type, ↓ Hard Knee Soft Knee MENU : Return	6. Use to adjust the setting, and press MENU.
Setting value	Explanation	MENU : Return
Hard Knee	Only peaks that exceed the threshold are attenuated. There is no effect below the threshold.	HINT
Soft Knee	The limiter gradually affects the output signal about 6 dB below the threshold for a gentler effect.	This can be set from –16 to –2 dBFS.

Output Limiter (continued)

Setting the attack time

This sets the amount of time until limiting starts after the output signal exceeds the threshold.



Setting the release time

put signal goes below the threshold.

This sets the amount of time until limiting stops after the out-

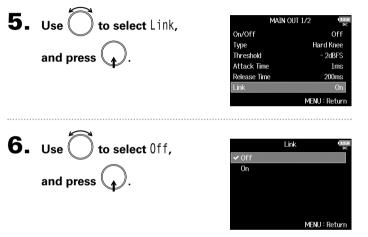
Off

1ms

On

Setting links

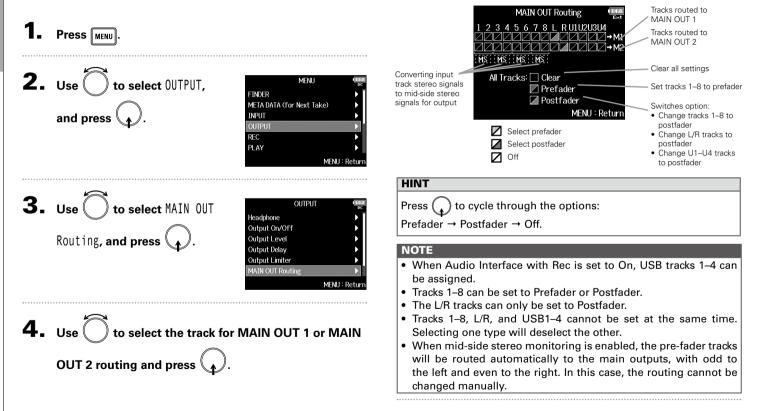
The limiter can be linked or applied separately to MAIN OUT 1 and MAIN OUT 2, as well as to SUB OUT 1 and SUB OUT 2.



Setting value	Explanation
Off	Separates limiter operation.
On	Links limiter operation. If the signal for either linked signal reaches the threshold, the limiter will operate on both
	channels.

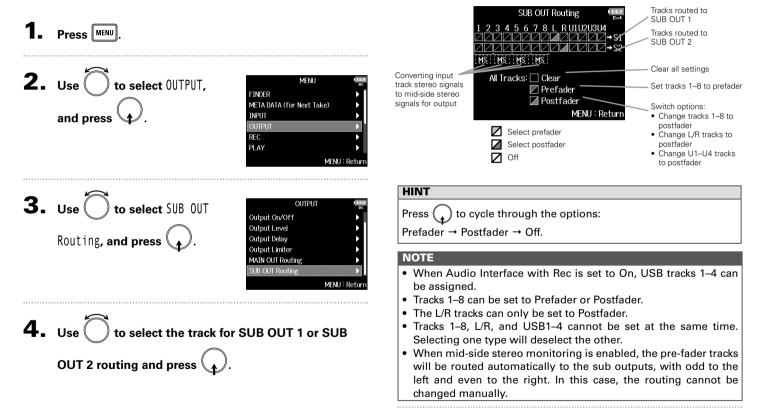
Selecting signals sent to the main outputs (MAIN OUT Routing)

You can send either prefader or postfader signals for each track to the main outputs.



Selecting signals sent to the sub outputs (SUB OUT Routing)

You send either prefader or postfader signals for each track to the sub outputs.







Timecode overview

The **FB** can input and output SMPTE timecode.

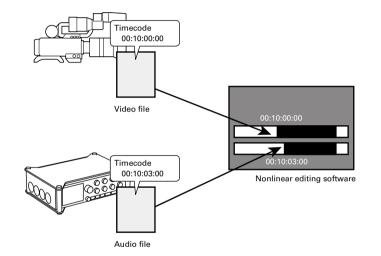
Timecode is time information written to data when recording video and audio. It is used for video editing, control of other devices, and synchronization of audio and video.

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.

HINT

The **FB** uses a precision oscillator that generates timecode with a high degree of accuracy (+/- 0.2 ppm, or approximately 0.5 frames per 24 hours).



ecode

Connection examples

Connections like the following are possible, depending upon the specific equipment being used with the **FB**.

Synchronizing with a video camera

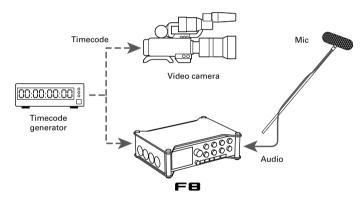
The **FB** records with mic input and transmits timecode. The **FB** saves the timecode that it generates with the audio data. The timecode received by the video camera is recorded with the video data.

Nic Video camera Timecode Audio

Inputting timecode

Timecode is transmitted from an external timecode generator. Both the **F B** 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 **FB** audio clock.



Setting timecode functions

Imecode

1. Press MENU.





3. Use to select Timecode, and press

TIMECODE	14.6 Ext
imecode	•
uto Rec Delay Time	0.0s
tart Timecode	Þ
	MENU : Return



Continue to one of the following procedures.

Setting the mode	.P.127
Stopping timecode output when recording is stopped	.P.128
Synchronizing audio clock with external timecode	.P.129
Automatically enabling internal timecode when no	
external timecode is input	.P.129
Setting the user bits for internal timecode	.P.130
Setting the frame rate for internal timecode	.P.131
Jamming internal timecode	.P.132
Restarting internal timecode with a specified value	.P.132

Setting the mode

- The timecode mode settings allow you to specify:Whether the **FB** generates timecode or receives external timecode
- Whether or not timecode continues running when not recording

4.	Use to select Mode, and press .	Timecode Timecode Mode: Int Free Run Int TC: 00x20x166.12r Ubits:01 15 00 FPS:30ND Ext TC: 00x01x07r.08r Ubits:00 00 00 FPS:30ND Jam Restart MENU: Return
5.	Use to select Mode, and press .	Mode Int Free Run Mode Int Free Run Int Auto Mute Off Ext Audio Clock Sync Off Ext Continuous Off MENU : Return MENU : Return
6.	Use \bigcirc to select the mode, and press \bigcirc .	Mode Off Int Free Run Int Record Run Int RTC Run Ext Ext Auto Rec MENU : Return

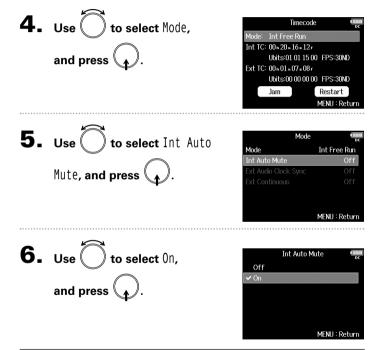
Off No timecode will be written to the recording file. Timecode will not be output from the TIMECODE OUT jack. Internal timecode will be generated regardless of the recording mode. Int Free Run The internal timecode can be set manually using the following menu items: MENU > TIMECODE > Timecode > Jam MENU > TIMECODE > Timecode > Restart Timecode will always be output from the TIMECODE OUT jack. Int Record Run Internal timecode will be generated only when recording. The internal timecode can be set manually using the following menu items: MENU > TIMECODE > Timecode > Jam MENU > TIMECODE > Timecode > Restart Timecode will always be output from the TIMECODE OUT jack. Int Record Run Internal timecode will be generated only when recording. The internal timecode can be set manually using the following menu items: MENU > TIMECODE > Timecode > Jam MENU > TIMECODE > Timecode > Jam MENU > TIMECODE > Timecode > Restart When switching from another mode, or when recording stops, the internal timecode will stop at the last value. Internal timecode will be generated regardless of the recording mode. In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): At startup When Date/Time (RTC) has changed (→ P:19) When switching to this timecode mode 	Setting value	Explanation	
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Int Free Run • MENU > TIMECODE > Timecode > Jam • MENU > TIMECODE > Timecode > Restart Timecode will always be output from the TIMECODE OUT jack. Internal timecode will be generated only when recording. The internal timecode can be set manually using the following menu items: • MENU > TIMECODE > Timecode > Jam • MENU > TIMECODE > Timecode > Restart When switching from another mode, or when recording stops, the internal timecode will stop at the last value. Internal timecode will be generated regardless of the recording mode. In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): • At startup • When Date/Time (RTC) has changed (→ P.19) • When switching to this timecode mode		, 3	
• MENU > TIMECODE > Timecode > Restart Timecode will always be output from the TIMECODE OUT jack. Internal timecode will be generated only when recording. The internal timecode can be set manually using the following menu items: • MENU > TIMECODE > Timecode > Jam • MENU > TIMECODE > Timecode > Bestart When switching from another mode, or when recording stops, the internal timecode will stop at the last value. Internal timecode will be generated regardless of the recording mode. In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): • At startup • When Date/Time (RTC) has changed (→ P.19)	Int Free Run	5	
Timecode will always be output from the TIMECODE OUT jack. Internal timecode will be generated only when recording. The internal timecode can be set manually using the following menu items: Int Record Run • MENU > TIMECODE > Timecode > Jam • MENU > TIMECODE > Timecode > Restart When switching from another mode, or when recording stops, the internal timecode will stop at the last value. Internal timecode will be generated regardless of the recording mode. In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): • At startup • When Switching to this timecode mode			
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Internal timecode will be generated only when recording. The internal timecode can be set manually using the following menu items: Int Record Run • MENU > TIMECODE > Timecode > Jam • MENU > TIMECODE > Timecode > Restart When switching from another mode, or when recording stops, the internal timecode will stop at the last value. Internal timecode will be generated regardless of the recording mode. In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): • At startup • When switching to this timecode mode			
Int Record Run The internal timecode can be set manually using the following menu items: Int Record Run • MENU > TIMECODE > Timecode > Jam • MENU > TIMECODE > Timecode > Jam • MENU > TIMECODE > Timecode > Restart When switching from another mode, or when recording stops, the internal timecode will stop at the last value. Internal timecode will be generated regardless of the recording mode. In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): • At startup • When Date/Time (RTC) has changed (→ P.19) • When switching to this timecode mode			
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stops, the internal timecode will stop at the last value. Internal timecode will be generated regardless of the recording mode. In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): Int RTC Run When Date/Time (RTC) has changed (→ P.19) When switching to this timecode mode		 MENU > TIMECODE > Timecode > Restart 	
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Int RTC Run recording mode. Int the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): • At startup • When Date/Time (RTC) has changed (→ P.19) • When switching to this timecode mode		stops, the internal timecode will stop at the last value.	
In the following situations, the internal timecode will be synchronized (jammed) with the RTC (internal clock): • At startup • When Date/Time (RTC) has changed (→ P.19) • When switching to this timecode mode		Internal timecode will be generated regardless of the	
Int RTC Run synchronized (jammed) with the RTC (internal clock): • At startup • When Date/Time (RTC) has changed (→ P.19) • When switching to this timecode mode		recording mode.	
Int RTC Run • At startup • When Date/Time (RTC) has changed (→ P.19) • When switching to this timecode mode		In the following situations, the internal timecode will be	
 When Date/Time (RTC) has changed (→ P.19) When switching to this timecode mode 		synchronized (jammed) with the RTC (internal clock):	
When switching to this timecode mode	Int RTC Run	At startup	
5		 When Date/Time (RTC) has changed (→ P.19) 	
		 When switching to this timecode mode 	
Timecode will always be output from the TIMECODE OUT		Timecode will always be output from the TIMECODE OUT	
jack.		jack.	
The internal timecode will chase the external timecode.		The internal timecode will chase the external timecode.	
Ext You can also enable the automatic generation of internal	Ext	You can also enable the automatic generation of internal	
timecode when there is no external timecode. (\rightarrow P. 129)		timecode when there is no external timecode. (\rightarrow P. 129)	

Setting timecode functions (continued)

Setting value	Explanation	
	The internal timecode will chase the external timecode.	
	You can also enable the automatic generation of internal	
Ext Auto Rec	timecode when there is no external timecode. (\rightarrow P.115)	
LXI AUTO Nec	Recording starts automatically when external timecode	
	input is detected. Recording stops automatically when	
	external timecode stops.	

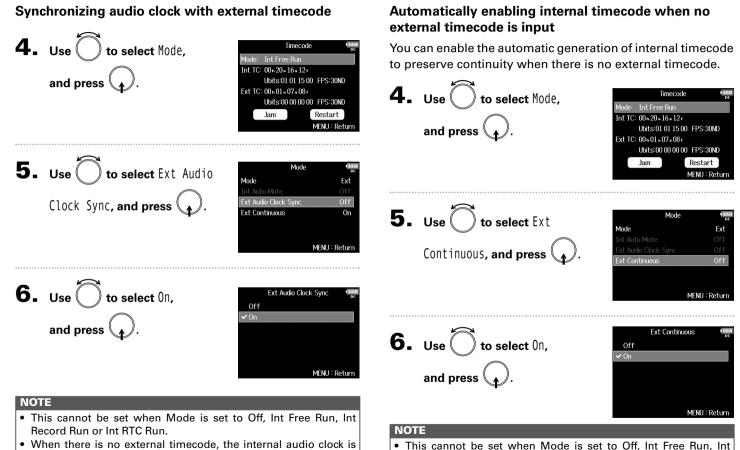
Stopping timecode output when recording is stopped

You can set whether or not timecode is output from the TIMECODE OUT jack when recording is stopped.



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.



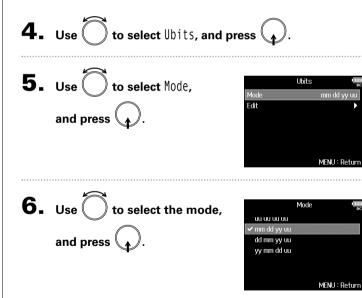
Record Run or Int RTC Run

Setting timecode functions (continued)

Setting the user bits for internal timecode

User bits are data that you can set to be included 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



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

HINT

Only "uu" items can be changed.

Setting user bits (Ubits)

4. Use to select Ubits, and press

5. Use to select Edit,

and press (

	Ubits	12. DC
Mode	mm	dd yy uu
Edit		►
		J:Return

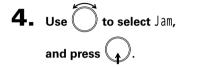
 Edit the value. Editing operations Move cursor or change value: Turn Select parameter 		to select the frame 23976ND 24ND 25ND 2997ND 2997D 30ND MENU : Return
to change: Press 😱	Setting value	Explanation
HINT User bits can only consist of numbers from 0 to 9 and letters from A to F.	23.976ND	This is the most common frame rate used with HD cameras and other high-definition video recording.The count is 0.1% slower than the actual time.
	24ND	This is the standard frame rate used for recording film. This is also used with HD cameras.
7. When done changing the	25ND	This is the frame rate for PAL video. This is used for PAL video, which is used in Europe and other regions.
setting, use to select 01 07 15 00 Enter, and press .	29.97ND	This is a frame rate used for NTSC color video and HD cameras. The count is 0.1% slower than the actual time. This is used for NTSC video, which is used in Japan, the United States and other countries.
MENU : Cancel	29.97D	This is an adjusted frame rate that uses drop frames to make NTSC match the actual time. This is used with video for broadcast that requires the actual time frame to be matched.
Setting the frame rate for internal timecode 4. Use to select FPS,	30ND	This is used to synchronize sound with film that is being transferred to NTSC video. This is the standard frame rate used for black-and-white television in Japan, the United States and other countries.
Mode: Int Free Run and press . Ext C: 00.03 - 38 - 23 - 18 - 100 FPS: 30 ND	30D	This rate is used for special applications. This synchronizes with film sound to be transferred to NTSC using 29.97fps drop frame. The count is 0.1% faster than the actual time.

Matching frame rates must be set in advance on all connected video and audio devices.

Setting timecode functions (continued)

Jamming internal timecode

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



	Timecode	12.0 DC
Mode:	Int Free Run	
Int TC:	00 h 01 m 21 s 24 f	
	Ubits:01 01 15 00	FPS:30ND
Ext TC:	00 n 03 m 34 s 29 r	
	Ubits:E0 00 00 00	FPS:30ND
	Jam	Restart
		MENU : Return

Restarting internal timecode with a specified value

4. Use to select Restart, and press . Timecode Mode: Int Free Run Int TC: 00x01x33x24r Ubits:01.01.15.00 FPS:30ND Ext TC: 00 n 03 n 46 s 29 r Ubits:E0.00.00.00 EPS:30ND Jam MENU : Return **5.** Set the restart value. Restart Int IC:

 Editing operations Move cursor or change value: Turn Select parameter Press (to change:

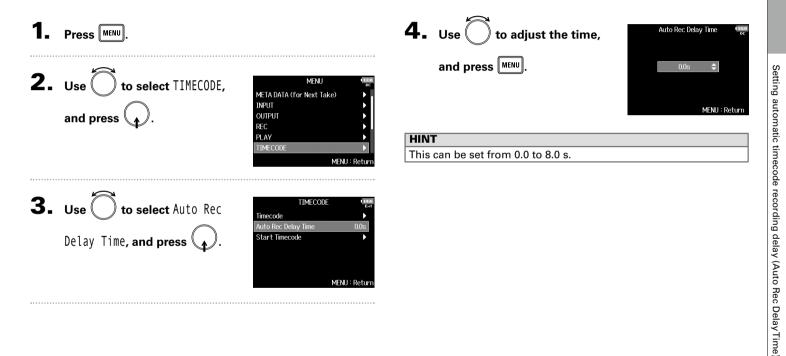


6. Use to select Restart, and press .



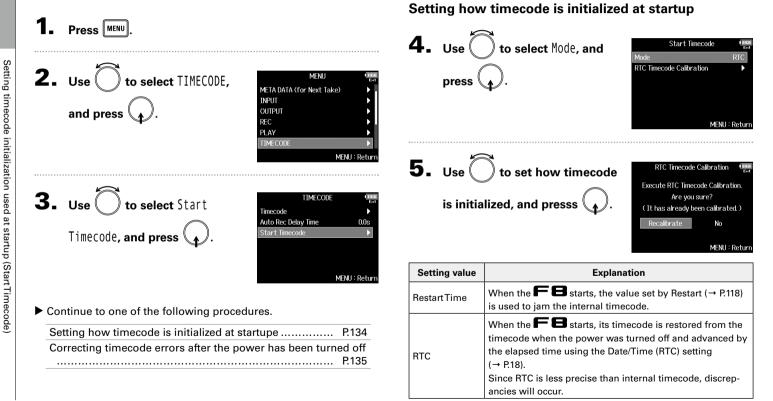
Setting automatic timecode recording delay (Auto Rec Delay Time)

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



Setting timecode initialization used at startup (Start Timecode)

Since internal timecode stops when the **FB** is turned off, the timecode is automatically initialized (jammed) during startup. You can set the value that is used for jamming at that time.



RTC Timecode Calibration

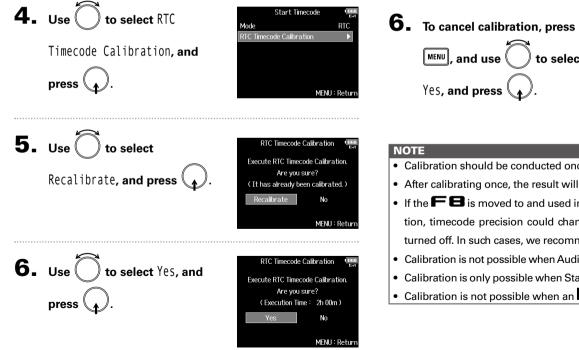
Complete !

MENU : Return

Setting timecode initialization used at startup (Start Timecode)

Correcting timecode errors after the power has been turned off

When the mode of Start Timecode is set to RTC, turning the power off lowers the timecode precision, this function can improve the error to the value close to 0.2 ppm when the power has been turned off.



2:00:00 / 2:00:00 MENU, and use to select

Calibration completes.



- Calibration should be conducted once after updating the firmware.
- After calibrating once, the result will be retained.
- If the **FB** is moved to and used in an extremely hot or cold location, timecode precision could change slightly when the power is turned off. In such cases, we recommend calibrating it again.
- Calibration is not possible when Audio Interface with Rec is set to On.
- Calibration is only possible when StartTimecode mode is set to RTC.
- Calibration is not possible when an **FRC-8** is connected.

Slate mic and slate tone overview

When recording with the **FB**, you can add audio comments that describe, for example, the scene being filmed or the anticipated cuts. You can also record slate tone signals that can be used to synchronize with video.

The **FB** has a built-in slate mic for recording comments and the ability to output a variable frequency tone signal.

HINT

A "slate" is a clapperboard used when recording video.

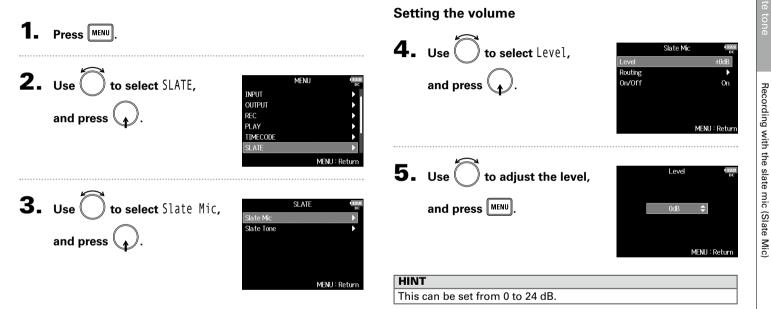
NOTE

- The slate mic and slate tone cannot be used at the same time.
- The slate mic and slate tone cannot be used during audio file playback.

137

Recording with the slate mic (Slate Mic)

You can use the built-in slate mic to record comments and to keep notes about recorded takes.



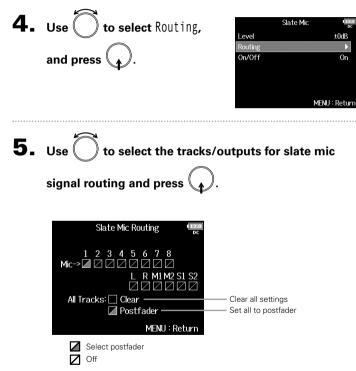
Continue to one of the following procedures.

Setting the volume	P.137
Setting the routing	P.138
Recording	P.139
Disabling the slate mic	P.139

Recording with the slate mic (Slate Mic) (continued)

Setting the routing

Set the destination for the slate mic signal.



NOTE

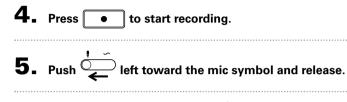
6. Press MENU.

Routing to tracks 1–8 is not possible when operating the **FB** as an audio interface (Stereo Mix).



Recording with the slate mic (Slate Mic

Recording



6. To disable the slate mic, push $\overset{!}{\smile}$ left toward the

mic symbol again and release.

NOTE

- When the slate mic is in use, other signals input to the tracks that it is routed to are muted.
- The slate mic signal is always routed to the headphone L/R channels regardless of other routing settings.
- The MAIN OUT 1/2 and SUB OUT 1/2 faders do not affect the levels of the slate mic and slate tone.

HINT If you push and hold $\overbrace{\leftarrow}^{i}$ left toward the mic symbol for two or more seconds, the slate mic will be enabled until you release the switch. Disabling the slate mic

and press

4. Use to select On/Off,

You can set the slate mic so that it will not be enabled if $\overleftarrow{\leftarrow}$ is accidentally pushed left toward the mic symbol.

5. Use to select Off (Lock), and press

On/Off Con On MENU : Return

Slate Mic

+0dB

MENH

Level

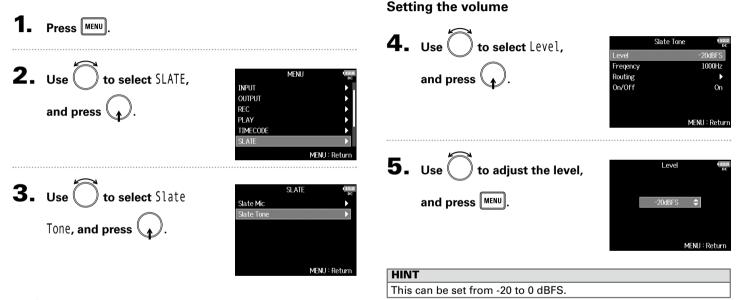
Routing

On/Off



Recording a slate tone (Slate Tone)

By adding a slate tone when the recording starts, aligning audio to video during editing will be easier. You can also use a slate tone to coordinate levels with connected equipment.



Continue to one of the following procedures.

Setting the volume	P.140
Setting the frequency	
Setting the routing	P.141
Recording	P.142
Disabling the slate tone	P.143

Recording a slate tone (Slate Tone)

Setting the frequency Setting the routing Set the destination for the slate tone signal. **4.** Use to select Frequency, Slate Tone Level -20dBFS **4.** Use) to select Routing, Slate Tone and press Level -20dBFS Routing Freqency 1000Hz On/Off On and press On/Off On MENU : Returi MENU : Return **5.** Use () to adjust the Freqency **5.** Use to select the tracks/outputs for slate tone frequency, and press MENU signal routing and press MENU : Return Slate Tone Routing HINT 1 2 3 4 5 6 7 8 Tone-> This can be set from 100 to 10,000 Hz. M1 M2 S1 S2 All Tracks: 🗌 Clear Clear all settings Postfader 🗸 Set all to postfader MENU : Return Select postfader ✓ Off NOTE Routing to tracks 1–8 is not possible when operating the **FB** as an audio interface (Stereo Mix).

HINT

Recording a slate tone (Slate Tone) (continued)

Recording Press () to switch between Postfader and Off. **4.** Press • to start recording **5.** Push $\stackrel{!}{\longrightarrow}$ right toward the tone symbol and 6. Press MENU. release.

NOTE

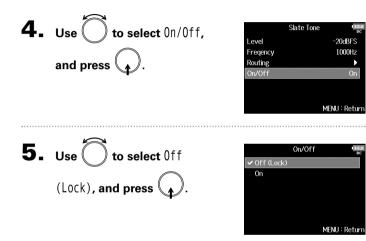
- When the slate tone is in use, other signals input to the tracks that it is routed to are muted.
- The slate tone signal is always routed to the headphone L/R channels regardless of other routing settings.
- The MAIN OUT 1/2 and SUB OUT 1/2 faders do not affect the levels of the slate mic and slate tone.

HINT

If you push and hold $\stackrel{\stackrel{\scriptstyle \bullet}{\longrightarrow}}{\longrightarrow}$ right toward the tone symbol for one or more seconds, the slate tone will be enabled until you push the switch toward the tone symbol again.

Disabling the slate tone

You can set the slate tone so that it will not be enabled if $\stackrel{\sim}{\longrightarrow}$ is accidentally pushed right toward the tone symbol.



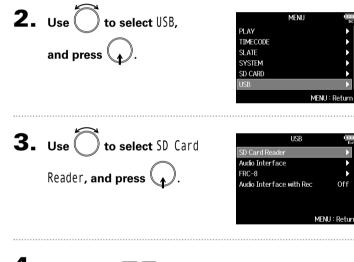
Ċ

Exchanging data with a computer (SD Card Reader)

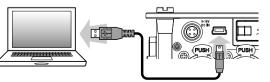
By connecting the **FB** to a computer, you can check and copy data on SD cards.

Connecting to a computer

1. Press MENU.



4. Connect the **FB** and computer with a USB cable.



NOTE

- The following operating systems are supported: Windows: Windows 7 or later Mac OS: Mac OS X (10.8 or later)
- The FB cannot operate on USB bus power. Use the internal batteries, the dedicated AC adapter or an external DC power supply to power it.

HINT

When the **FB** is connected to a computer, the SD cards loaded in slots 1 and 2 are recognized as separate SD cards.

Disconnecting

Disconnect on the computer.

Windows: Select F S from the "Safely Remove Hardware and Eject Media" icon on the bottom of the computer screen.

Mac OS: Drag and drop the FE icon to the Trash.

NOTE

Always follow correct computer disconnection procedures before removing the USB cable.

2. Disconnect the cable from the computer and the

FB, and press **MENU**.

Using as an audio interface (Audio Interface)

FB input signals can be input directly to a computer or an iOS device, and playback signals coming from a computer or an iOS device can be output from the **FB**.

Connecting to a computer or an iOS device

1. Press MENU.		
2. Use to select USB, and press .	MENU PLAY TIMECODE SLATE SYSTEM SD CARD USB MEN	J: Return
3. Use to select Audio Interface, and press	USB SD Card Reader Audio Interface FRC-8 Audio Interface with Rec MEP	(TERE Est ► ● Off NU : Return
4. Use to select the mode and connected device, and press .	Audio Interface Stereo Mix (PC/Mac) Stereo Mix (Pad) Multi Track (PC/Mac)	(1228) DC ► ►

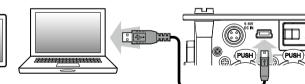
Setting value	Explanation
Stereo Mix (PC/Mac)	This is a 2-in/2-out connection mode for Mac/Windows and sends tracks 1–8 as a stereo mix.
Stereo Mix (iPad)	This is a 2-in/2-out connection mode for iOS devices and sends tracks 1–8 as a stereo mix.
Multi Track (PC/Mac)	This is an 8-in/4-out connection mode for Mac/Win- dows and sends tracks 1–8 as separate signals (cannot be used with an iOS device). A driver is necessary for use with Windows. Download the driver from the ZOOM website (www.zoom.co.jp/).



MENU : Return

5. Use a USB cable to connect the **FB** and the

computer or iOS device.

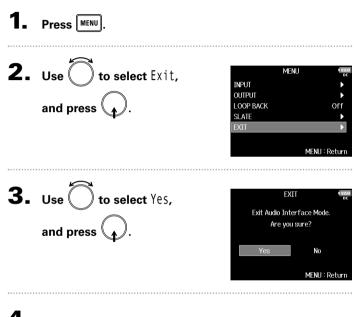


Using as an audio interface (Audio Interface) (continued)

NOTE

- A Lightning to USB Camera Adapter is necessary to connect an iOS device.
- The **FB** cannot operate on USB bus power. Use the internal batteries, the dedicated AC adapter or an external DC power supply to power it.
- When the F B is used as an audio interface and the sampling rate is set to 44.1/48 kHz, latency increases 2 ms. When latency increases while monitoring sounds being recorded with a mic in real-time, interference occurs between sound transmitted through the air from the source and the delayed monitored sound, possibly making accurate monitoring of the sounds difficult.

Disconnecting



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

Using SD card recording and audio interface functions at the same time (Audio Interface with Rec)

In addition to the two SD cards, a computer can also be used for recording backup.

Connecting

1. Press MENU.	
2. Use to select USB, and press .	MENU TIMECODE
3. Use to select Audio Interface with Rec, and press	USB term SD Card Reader Audio Interface FRC-8 Audio Interface with Rec Off MENU : Return
4. Use \bigcirc to select $0n$, and press \bigcirc .	Audio Interface with Rec Off ✓ On MENU : Return



computer.

Using SD card recording and audio interface functions at the same time (Audio Interface with Rec) (continued)

NOTE

- The **FB** cannot operate on USB bus power. Use the internal batteries, the dedicated AC adapter or an external DC power supply to power it.
- Audio Interface with Rec cannot be used with the following settings and functions.
 - Sampling rate settings other than 44.1/48 kHz
 - SD Card Reader (→ P.144)
 - Audio Interface (\rightarrow P.145)
 - FRC-8 (→ P.152)
- A driver is necessary for use with Windows. Download the driver from the ZOOM website (www.zoom.co.jp/).
- When Audio Interface with Rec is set to On, the sampling rate cannot be changed.
- When Audio Interface with Rec is set to On, files with sampling rates that differ from the **FB** setting cannot be played.
- Set the input source to USB1-4 (\rightarrow P. 80) or set USB1-4 to the output routing (\rightarrow P109, 122, 123) to monitor sound played back from the computer. (\rightarrow P. 80)
- When Audio Interface with Rec is set to On, the **FB** latency will increase 2 ms. When monitoring sounds being recorded with a mic in real-time, increased latency can cause interference between sound transmitted through the air from the source and the delayed monitored sound, possibly making accurate monitoring of the sounds difficult.

Disconnecting

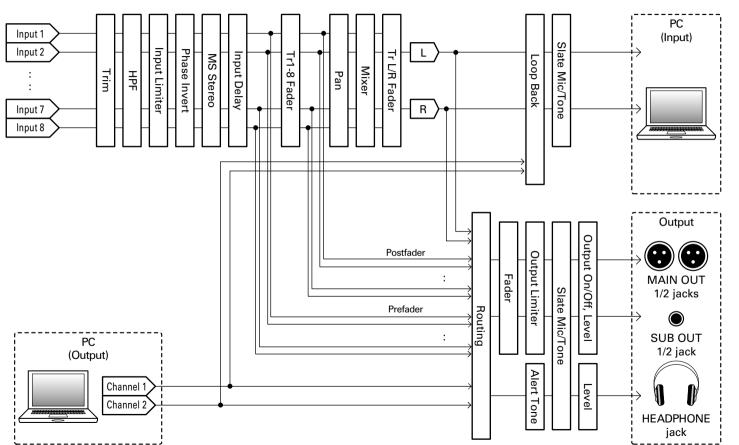
1. Press MENU. **2.** Use to select Off, and Audio Interface with Rec On MENU : Retur

3. Disconnect the cable from the computer and the

F8.

Audio interface block diagrams

Stereo Mix

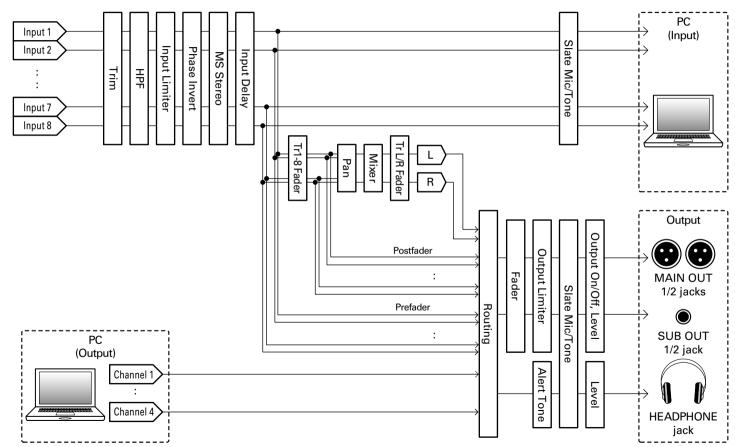


Audio interface block diagrams

Audio interface block diagrams

Audio interface block diagrams (continued)

Multi Track



Audio interface settings

The following settings can be made when using the **FB** as an audio interface. See the relevant pages for details about operation.

Setting loop back (Stereo Mix only)

This function allows the playback sound from the computer or iOS device and the **F B** inputs to be mixed and sent back to the computer or iOS device (loop back).

You can use this function to add narration to music played back from the computer and record the mix or stream it from the computer, for example.

Mixing inputs

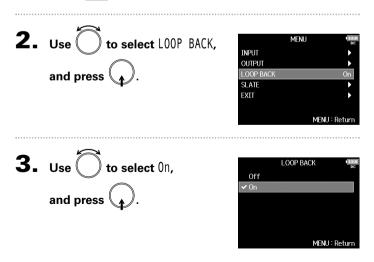
You can adjust the mix balance of input signals sent to the computer or iOS device. When using Multitrack mode, the individual inputs will be sent. When using Stereo Mix mode, the mixed stereo signal will be sent.

• Open the mixer on the Home

Screen. (\rightarrow P.11)



Press MENU

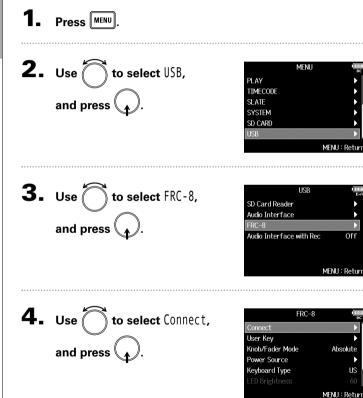


2. Adjust the parameter settings.

See "Adjusting the input signal monitoring balance" (\rightarrow P.75) for how to change settings.

Using an FRC-8 as a controller (Connect)

By connecting an **FRC-8** to the **FB**, you can use it to adjust trim, fader and pan settings, for example.



5. Use a USB cable to connect the $\mathbf{F}\mathbf{B}$ and

the FRC-8.

6. Turn the FRC-8 power ON.

NOTE When disconnecting the **FRC-8**, select "Disconnect" before unplugging the USB cable.

Setting the type of keyboard connected to the FRC-8 (Keyboard Type)

Setting the type of keyboard connected to the FRC-8 (Keyboard Type)

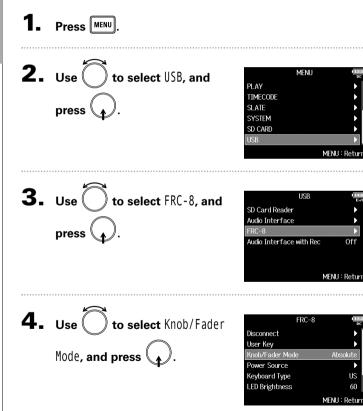
You can connect a PC keyboard to the **FRC-8** and use it to input characters. Set the type of PC keyboard connected to the **FRC-8**.

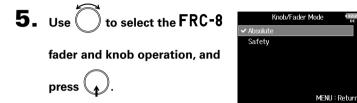
 Press IND. Use to select USB, and press to. 	Type and press (.).	FRC-8 User Disconnect > User Key > Knob/Fader Mode Absolute Power Source > Keyboard Type US LED Brightness 60 MENU : Return Keyboard Type US JP US
3. Use to select FRC-8, and press . MENU: F	turn US Use for English-language k	

Setting value	Explanation	
US	Use for English-language keyboards.	
JP	Use for Japanese keyboards.	

Setting FRC-8 fader and knob operation (Knob/Fader Mode)

How the **FRC-8** faders and TRIM/PAN knobs operate when their positions differ from actual parameter values can be set.





Setting value	Explanation	
Absolute	When a knob or fader is operated, the parameter value will change to the value shown by that knob or fader.	
Safety	When a knob or fader is operated, the parameter value will not change until the knob or fader first matches that value.	

NOTE

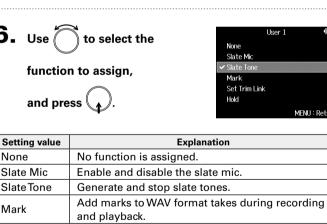
The **FRC-8** headphone volume operation cannot be changed.

Using an FRC-8

Setting user keys for the FRC-8 (User Key)

You can assign functions to the **FRC-8** user keys.

1. Press MENU.		5. Use
2. Use to select USB,	MENU (1828) PC	which t
and press \bigcirc .	PLAY	and pre
	MENU : Return	6. Use
3. Use to select FRC-8,	USB (Internet	function
and press \bigcirc .	SD Card Reader ▶ Audio Interface ▶ FRC-8 ▶ Audio Interface with Rec Off	and pre
		Setting value
	MENU : Return	None
		Slate Mic
		Slate Tone
4. Use to select User Key,	FRC-8	Mark
	User Key Knob/Fader Mode Absolute	Set Trim Link
and press 🔔.	Power Source	Hold
	Keyboard Type US LED Brightness 60	Clear Clip Indicator
	MENU : Return	Circled



Open the MENU > INPUT > Trim Link screen.

Clear the level meter clipping indicators.

Circle the currently selected take.

Use to disable the keys set with "Key HoldTarget".

to select the key to

which to assign a function,

and press

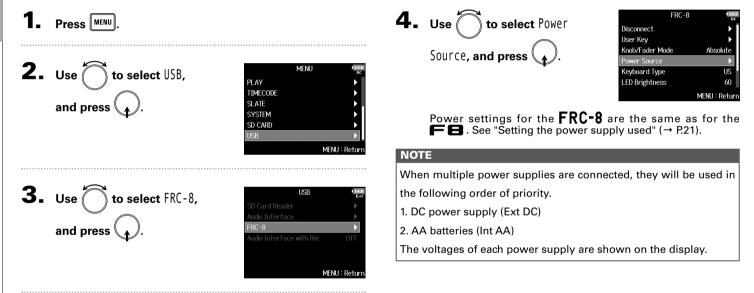




Setting user keys for the FRC-8 (User Key)

Setting the power supply used by the FRC-8 (Power Source)

Set the DC power supply shutdown voltage, nominal voltage and type of batteries so that the remaining power supply charge can be shown accurately. On this menu page, you can check the voltage of each power supply and the remaining battery capacity.



Setting the FRC-8 LED brightness (LED Brightness)

Setting the FRC-8 LED brightness (LED Brightness)

You can adjust the brightness of the LEDs on the **FRC-8**.

1. Press MENU.	5. Use	$\mathbf{\tilde{\mathbf{C}}}$
2. Use to select USB, and press .	bright MENU PLAY TIMECODE SLATE SYSTEM SD CARD USB MENU: Return This can be s	
3. Use to select FRC-8, and press .	USB For SD Card Reader Audio Interface FRC-8 Audio Interface with Rec Off MENU : Return	
4. Use to select LED Brightness, and press .	FRC-8 Disconnect User Key Knob/Fader Mode Absolute Power Source Keyboard Type LED Brightness 60 MENU : Return	



from 5 to 100.

Updating the FRC-8 firmware

You can check the **FRC-8** firmware version and update it to the latest version. The latest update file can be downloaded from the ZOOM website (www.zoom.co.jp).

1. See "Using an FRC-8 as a controller" (\rightarrow P.136), and

connect the FE and the FRC-8.

NOTE

Updating is not possible if the remaining battery or DC power supply charge is low. In this case, replace the batteries with new ones or use a charged DC power supply.

2. Copy the update file to the root directory on an SD

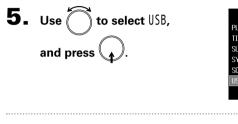
card.

3. Load the SD card into the SD CARD 1 slot.

NOTE

If an SD card is loaded in the SD CARD 2 slot, eject it.







6. Use to select FRC-8, and press



Continue to one of the following procedures.

Checking the firmware version	P.159
Updating the firmware	P.159

Updating the FRC-8 firmware

Checking the firmware version

7. Use to select Firmware Version, and press

FRC-8 Disconnect User Key Power Source Keyboard Type LED Brightness 60 irmware Version MENU : Return **Firmware Version** System Version : 1.00 Boot Version : 1.00

MENU : Return

Updating the firmware

7. Use to select Update Firmware, and press .



 $1.00 \rightarrow 1.10$

Are you sure?

No

MENU : Return



NOTE

Do not turn the power off, remove an SD card or disconnect the USB cable during an update. Doing so could cause the **FRC-8** to become unstartable.

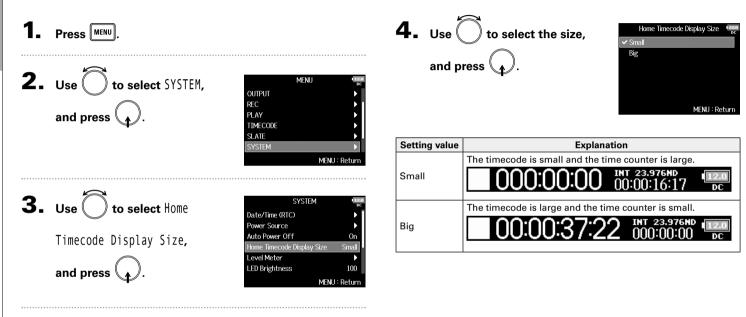
9. After the update completes,

turn the FRC-8 power off.



Setting how timecode is shown (Home Timecode Display Size)

You can change the size of the timecode display on the Home Screen.



Setting level meter appearance (Level Meter)

You can set how the level meters appear on the display.

1.	Press MENU.	You can set whether the level meter
2.	Use to select SYSTEM, and press .	4. Use to select Type, and press .
3.	Use to select Level Meter, and press .	and press .
►	Continue to one of the following procedures.	
	Setting the type	
	Setting the peak hold time	P.162
	Setting the level meter resolution	P.163
	Setting which track level meters are shown	
	on the Home Screen	P.163
	Showing track names on level meters	P.164
	Setting the level meter reference	P.164

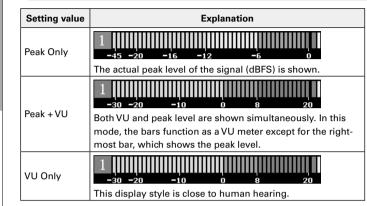
Setting the type

ther the level meters use VU or Peak style.



	Туре	12. DC
✓ Peak Only		
Peak + VU		
VU Only		
		MENU : Return

Setting level meter appearance (Level Meter) (continued)



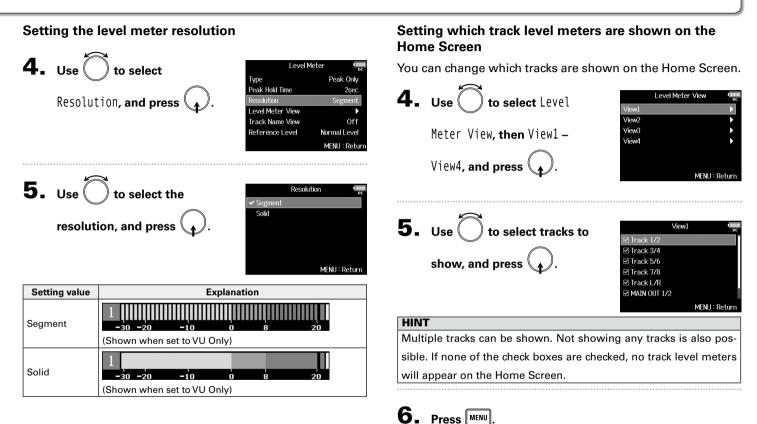
Setting the peak hold time

4. Use to select Peak Hold Time, and press

Level	Aeter Iza
Туре	Peak Only
Peak Hold Time	2sec
Resolution	Segment
Level Meter View	•
Track Name View	Off
Reference Level	Normal Level
	MENU : Return

5. Use to adjust the peak hold time, and press .

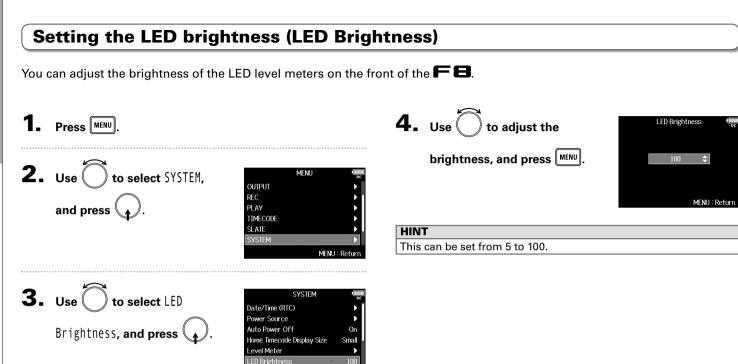




Setting level meter appearance (Level Meter) (continued)

Showing track names on level meters Setting the level meter reference **4.** Use to select Track Name View, and press . **4.** Use to select Reference Level, and press . Level Meter Level Meter Peak Only Туре Peak Only Туре Peak Hold Time Peak Hold Time 2sec 2sec Resolution Resolution Segment Segment Level Meter View Level Meter View Track Name View Off Track Name View Reference Level Normal Leve Reference Level Normal Level MENU : Retur MFNH: Retur 5. Use to select On, and press . Track Name View **5.** Use to select the Off Reference Level Normal Level Low Level reference level setting, and press (MENU : Return MENU: Return Setting value Explanation 000:00:00 00:42:29:15 The track names are not shown Off on the level meters. 000:00:00 The track names set with the On "Track Name" setting (\rightarrow P. 48) are shown on the level meters.

	Explanation		
Setting value	When level meter type is	When level meter type is	
	Peak Only	Peak + VU or VU Only	
	-45 −20 −16 −12 −6 0	-30 -20 -10 0 8 20	
	The center of the level	The center of the level meter	
Normal Level	meter is -12 dBFS. Clear	is 0 VU (-20 dBFS). Clear mon-	
	monitoring of levels	itoring of levels higher than 0	
	higher than -12 dBFS is	VU (-20 dBFS) is possible.	
	possible.		
Low Level	10000000000000000000000000000000000000	11111111111111111111111111111111111111	
	The center of the level	The center of the level meter is	
	meter is -20 dBFS. Clear	-10 VU (-30 dBFS). Clear moni-	
	monitoring of levels	toring of levels lower than -10	
	lower than -20 dBFS is	VU (-30 dBFS) is possible.	
	possible.		

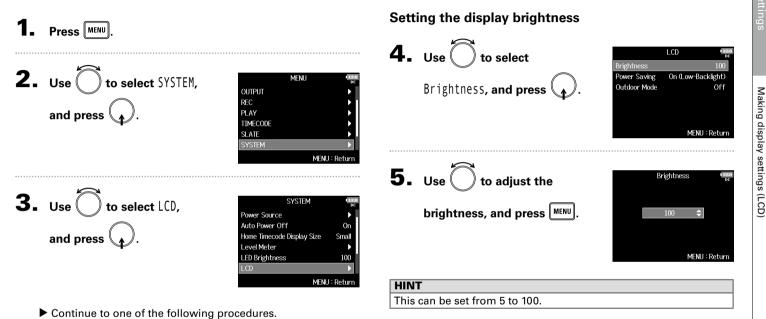


MENU : Return

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Making display settings (LCD)

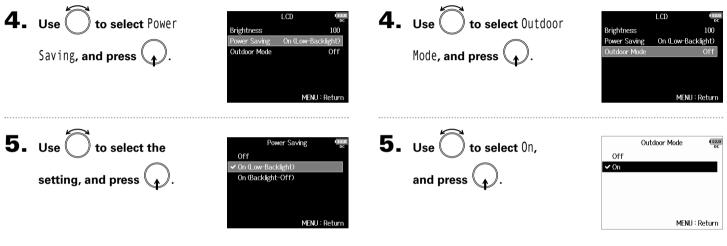
You can make settings related to the display.



Making display settings (LCD) (continued)

Changing the display backlight setting

You can set the display backlight to dim after 30 seconds without use.

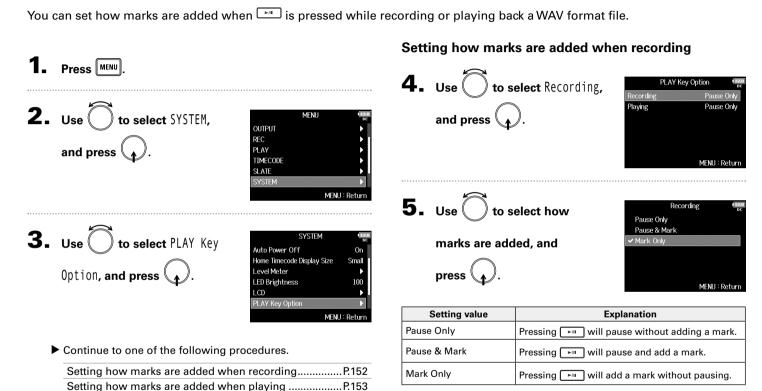


	MENU : Return	
Setting value	Explanation	
Off	The backlight brightness does not change even after time passes without use.	
On (Low-Backlight)	The backlight dims after time without use.	
On (Backlight-Off)	The backlight turns off after time without use.	

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.

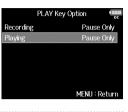
Making display settings (LCD



Adding marks when pausing (PLAY Key Option)

Setting how marks are added when playing







marks are added, and

press

Playing	[<u>12.0</u>] DC
✔ Pause Only	
Pause & Mark	
Mark Only	
	MENU : Return

Setting value	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.

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Various setting

Setting the keys held (Key Hold Target)

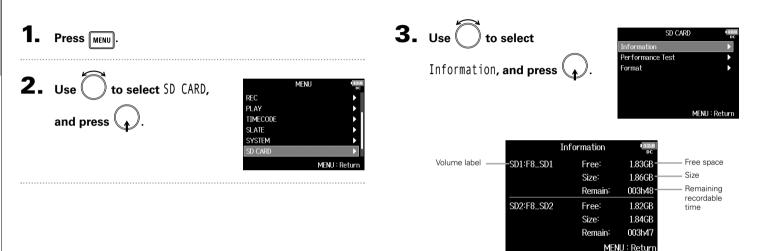
Setting the keys held (Key Hold Target)

Use the hold function to prevent misoperation during recording. Press - + 8 to turn it on/off. Follow these instructions to set which keys are disabled by the hold function.

1. Press MENU.	4. Use to select the keys
2. Use to select SYSTEM, OUTPUT and press .	held, and press .
SI ATF	HINT
SYSTEM >	You can select Track 1-8, PFL 1-8, Trim Knob 1–8, Slate Mic, Slate
MENU : Return	Tone, Encoder, MENU, HP Volume, REW, STOP, FF, PLAY and REC.
3. Use to select Key Hold	5. Press MENU.
Target, and press ↓ LCD Brightness ↓ 100 PLAY Key Option ↓	
Track Knob Option Fader Key Hold Target ► MENU : Return	 Even when hold is on for "STOP" and "Track1-8", you can press + I to turn the hold function off. Operation using the FRC-8 and F8 Control is possible even when the hold function is on.

Checking SD card information (Information)

You can check the size and free space of SD cards.



Testing SD card performance (Performance Test)

You can test whether an SD card can be used with the FB. The Quick Test is basic, and the Full Test checks the entire SD card.

Continue to one of the following procedures. Press MENU Conducting a guick test.....P.156 Conducting a full testP.157 **2.** Use **to select** SD CARD, MENU Conducting a guick test REC PLAY and press 5. Use to select Quick TIMECODE SD1 SLATE Quick Test SYSTEM Full Test Test, and press (**3.** Use to select SD CARD MENU : Return Information erformance Test Performance Test. Format 6. Use (**to select** Yes, Quick Test and press Execute Quick Test. Are you sure? MENU : Return and press 🔰 The card performance test will No start. The test should take about **4.** Use () to select the SD MENU : Return Performance Test 30 seconds. SD2 card to test, and press

MENU : Return

Testing SD card performance (Performance Test) (continued)

Conducting a full test 7. Performance Test The test completes. Result : OK The result of the evaluation will **5.** Use to select Full Test, SD1 50% 100% be shown. Quick Test and press . MENIL: Return The amount of time required for MENU : Return the full test will be shown. **8.** Press MENU to stop the test. 6. Use to select Yes, and press . NOTE Full Test Even if a performance test result is "OK", there is no guarantee Execute Full Test. Are you sure? that writing errors will not occur. This information is just to provide (Estimated Time : 0h 14m) guidance. No MENU : Return Performance Test

7. The test completes.

The result of the evaluation will be shown.

If the access rate MAX reaches 100%, the card will fail (NG).

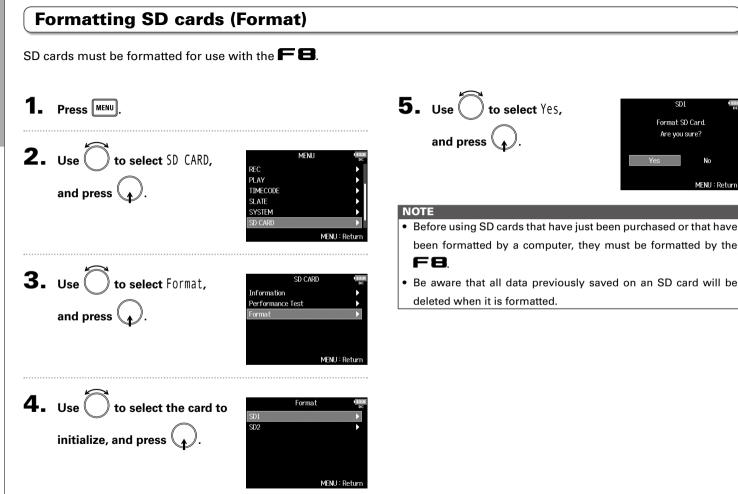


Testing SD card performance (Performance Test)

8. Press MENU to stop the test.

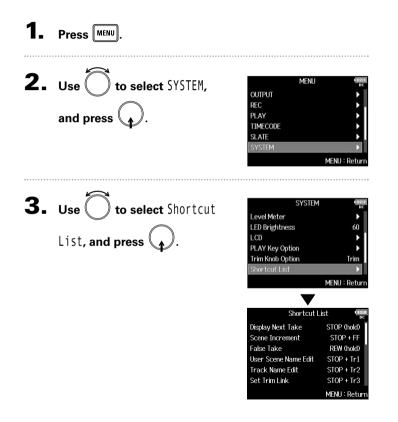
NOTE

- You can press **I** to pause and resume the test.
- Even if a performance test result is "OK", there is no guarantee that writing errors will not occur. This information is just to provide guidance.



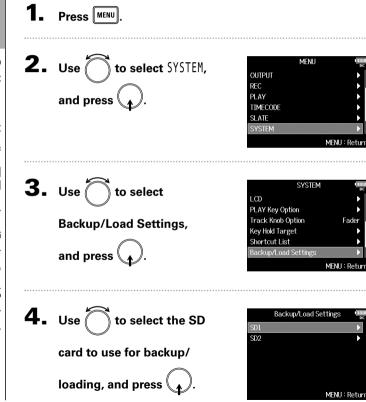
Checking the FB Shortcut List

The **FB** has a shortcut feature that allows quick access to various functions. See the "List of shortcuts" (\rightarrow P. 173) for information about the shortcut functions.



Backing up and loading **F B** settings (Backup/Load Settings)

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



Continue to one of the following procedures.

Backing up	178
LoadingP.	179

Backing up

This saves a backup file to the "F8_SETTINGS" folder in the root directory of the SD card.





6. Edit the name of the file

saved.

See "Character input screen" $(\rightarrow P.13)$ for how to input characters.

HINT

Fader

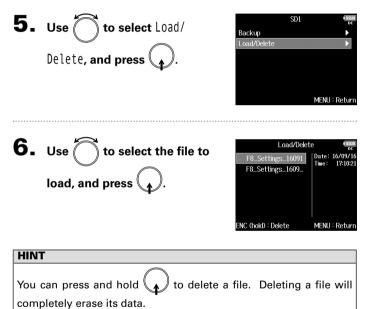
The extension of the saved backup file is ".ZSF".



Backing up and loading FB settings (Backup/Load Settings) (continued)

Loading

You can load a backup file that is saved in the "F8_SETTINGS" folder in the root directory of the SD card.



7. Use (**to select** Yes, and press (



Restoring default setting values (Factory Reset)

You can restore the factory default settings.

1.	Press MENU.	
2.	Use \bigcirc to select SYSTEM, and press \bigcirc .	MENU OUIPUT REC PLAY TIMECODE SLATE SYSTEM
		MEN
3.	Use to select Factory Reset, and press .	SYSTEM PLAY Key Option Track Knob Option Key Hold Target Shortcut List Backup/Load Settings Factory Reset
		MF

4. Use to select Yes, and press .

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

Factory Reset Reset all settings. Are you sure? Yes No MENU : Return

NOTE

Return

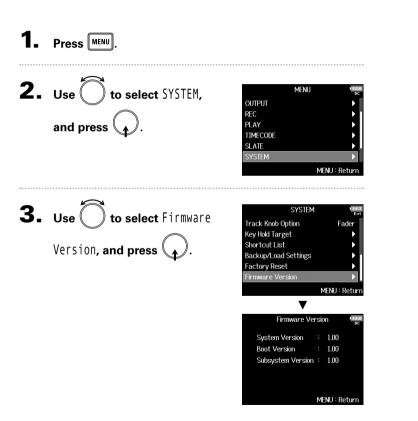
Fader

J:Returr

Input volume knob settings will not be reset.

Checking the firmware version (Firmware Version)

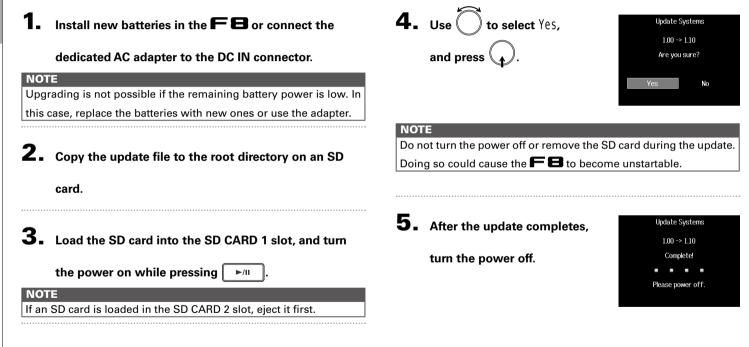
You can check the firmware version.



Updating the firmware

The **FB** firmware can be updated to the latest versions.

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



Troubleshooting

If you think that the **FB** is operating incorrectly, check the following items first.

Recording/playback trouble

- There is no sound or output is very quiet
- Check the connections to your monitoring system and its volume setting.
- · Confirm that the volume of the \blacksquare is not too low. (\rightarrow P.75)

• No sound from connected equipment or inputs or it is very quiet

- \cdot If you are using a mic capsule, confirm that it is oriented correctly.
- \cdot Check the input level settings. (\rightarrow P.27)
- 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. (\rightarrow P.75)
- \cdot Check the phantom power and plug-in power settings. (\rightarrow P.90, P.93)
- · Check the headphone, MAIN OUT 1/2 and SUB OUT 1/2 routing settings. (\rightarrow P.108, P.122-123)
- Recording is not possible
- \cdot Confirm that track keys are lit red.
- \cdot Confirm that the SD card has free space. (\rightarrow P.172)
- \cdot Confirm that an SD card is loaded properly in a card slot.
- If "Card Protected!" appears on the display, the SD card writeprotection is enabled. Slide the lock switch on the SD card to disable write-protection.

- The recorded sound cannot be heard or is very quiet
- \cdot Confirm that the volume levels of the tracks are not too low. (\rightarrow P.52)
- \cdot Confirm that track keys are lit green during playback.

Other trouble

- \bullet Computer does not recognize the $\ensuremath{\hbox{\rm FB}}$ even though it is connected to the USB port
- · Confirm that the operating system is compatible. (\rightarrow P.144)

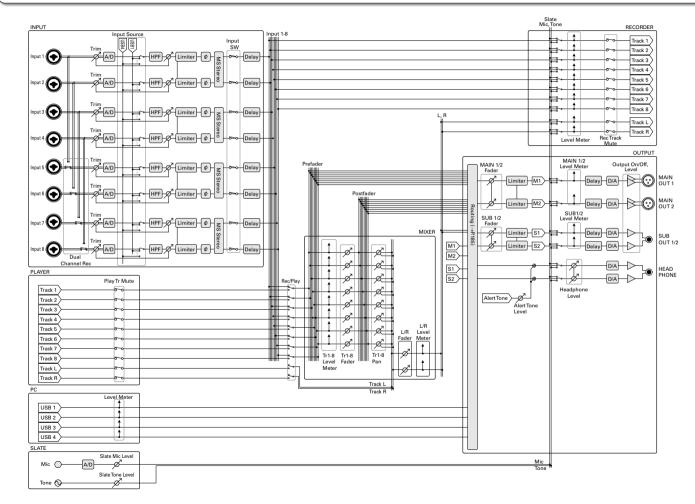
The operation mode must be set on the FB to allow the computer to recognize the FB. (\rightarrow P145)

• Battery operation time is short

Making the following settings increase the battery operation time.

- \cdot Set the power supply used correctly. (\rightarrow P.22)
- \cdot Turn unnecessary tracks off. (\rightarrow P.27)
- · Turn unnecessary outputs off. (\rightarrow P.114)
- · Set the phantom power voltage to 24V. (\rightarrow P.91)
- · Disable phantom power during playback. (\rightarrow P.92)
- \cdot Turn timecode off if not using it. (\rightarrow P.127)
- · Reduce the LED brightness.(\rightarrow P.166)
- \cdot Reduce the display brightness. (\rightarrow P.167)
- \cdot Set the display to dim when not used for some time. (\rightarrow P.168)
- \cdot Reduce the sampling rate used to record files. (\rightarrow P.30)
- Due to their characteristics, using nickel metal hydride batteries (especially high-capacity ones) or lithium batteries should enable longer use than alkaline batteries.

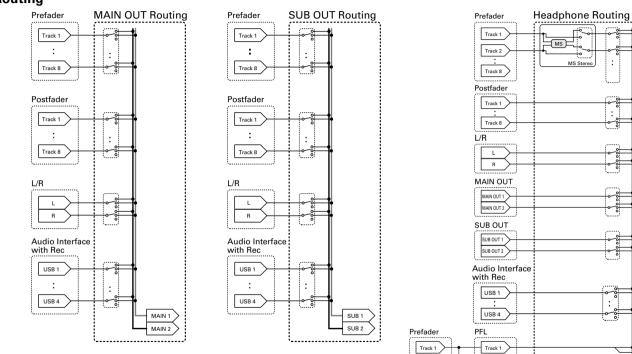
Detailed product diagrams



MS Stereo

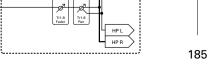
.





Track 8

Track 8 SOLO Track 1 Track 8



ň

8

Metadata list

Metadata contained in BEXT chunks in WAV files

Metadata list

Tag	Explanation	Remarks
SPEED=	Frame rate	MENU >TIMECODE >Timecode > FPS
TAKE=	Take number	
UBITS=	User bits	MENU >TIMECODE >Timecode > Ubits
SCENE=	Scene name	MENU > META DATA (for NextTake) > Scene Name Mode MENU > META DATA (for NextTake) > User Scene Name MENU > FINDER > Option >Meta Data Edit > Scene MENU > FINDER > Option > Rename
TAPE=	Name of recording destination folder	MENU > FINDER (Recording destination folder name) MENU > FINDER > Option > Meta Data Edit > Folder (Tape) Name
CIRCLED=	Circled take	MENU > FINDER > Option > Meta Data Edit > Circle
TRL=	Left track name	
TRR=	Right track name	
TR1=	Track 1 name	
TR2=	Track 2 name	Track names are written as follows.
TR3=	Track 3 name	TRL = left track, TRR = right track
TR4=	Track 4 name	TR1 = track 1, TR2 = track 2TR8 = track 8
TR5=	Track 5 name	During dual channel recording, tracks 1–4 are written to tracks 5–8.
TR6=	Track 6 name	
TR7=	Track 7 name	
TR8=	Track 8 name	
NOTE=	Take note	MENU > META DATA (for NextTake) > Note MENU > FINDER > Option > Meta Data Edit > Note

Metadata contained in iXML chunks in WAV files

iXML master tag	iXML sub tag	Written	Read	Remarks
				MENU > FINDER (SD card root folder)
<project></project>		0	0	MENU > FINDER > Option > Meta Data Edit > Project
				Name
				MENU > META DATA (for NextTake) > Scene Name Mode
<scene></scene>		0	0	MENU > META DATA (for NextTake) > User Scene Name
<scene></scene>		0	0	MENU > FINDER > Option > Meta Data Edit > Scene
				MENU > FINDER > Option > Rename
<take></take>		0	0	MENU > FINDER > Option > Meta Data Edit > Take
< IARE>		0	0	MENU > FINDER > Option > Rename
				MENU > FINDER (recording destination folder name)
<tape></tape>		0	0	MENU > FINDER > Option > Meta Data Edit > Folder
				(Tape) Name
<circled></circled>		0	0	MENU > FINDER > Option > Meta Data Edit > Circle
<wildtrack></wildtrack>		×	×	
<false start=""></false>		×	×	
<no good=""></no>		×	×	
<file_uid></file_uid>		0	×	
<ubits></ubits>		0	×	MENU > TIMECODE > Timecode > Ubits
NOTE		_	_	MENU > META DATA (for NextTake) > Note
<note></note>		0	0	MENU > FINDER > Option > Meta Data Edit > Note
<bext></bext>		×	×	
<user></user>		×	×	

Metadata list (continued)

iXML master tag	iXML sub tag	Written	Read	Remarks
<speed></speed>				
<speed></speed>	<note></note>	0	×	
<speed></speed>	<master_speed></master_speed>	0	0	MENU > TIMECODE > Timecode > FPS
<speed></speed>	<current_speed></current_speed>	0	×	MENU > TIMECODE > Timecode > FPS
<speed></speed>	<timecode_rate></timecode_rate>	0	×	MENU > TIMECODE > Timecode > FPS
<speed></speed>	<timecode_flag></timecode_flag>	0	×	MENU > TIMECODE > Timecode > FPS
<speed></speed>	<file_sample_rate></file_sample_rate>	0	×	MENU > REC > Sample Rate
<speed></speed>	<audio_bit_depth></audio_bit_depth>	0	×	MENU > REC > WAV Bit Depth
<speed></speed>	<digitizer_sample_rate></digitizer_sample_rate>	0	×	MENU > REC > Sample Rate
<speed></speed>	<timestamp_samples_since_midnight_hi></timestamp_samples_since_midnight_hi>	0	×	
<speed></speed>	<timestamp_samples_since_midnight_lo></timestamp_samples_since_midnight_lo>	0	×	
<speed></speed>	<timestamp_sample_rate></timestamp_sample_rate>	0	×	MENU > REC > Sample Rate

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>	0	×	
<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>	0	×	
<file_set></file_set>	<family_uid></family_uid>	0	×	
<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>	0	×	

iXML master tag	iXML sub tag	Written	Read	Remarks
<track_list></track_list>				
<track_list></track_list>	<track_count></track_count>	0	×	
<track/>	<channel_index></channel_index>	0	×	
<track/>	<interleave_index></interleave_index>	0	×	
<track/>	<name></name>	0	0	MENU > META DATA (for NextTake) > Track Name MENU > FINDER > Option > Meta Data Edit > Track Name
<track/>	<function></function>	×	×	

 $\circ = YES \times = NO$

Metadata and ID3 fields contained in MP3 files

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

List of shortcuts

Home Screen

Shortcut	Explanation
	Show the name that will be given to the next
Press and hold	take recorded.
	Example: Scene001-T002
	Advance the scene number by 1 (when the
	Home Screen is open).
	Move the previously recorded take to the
Press and hold 🖼	FALSETAKE folder (when the Home Screen
	is open).
	The number given to the next recorded take
Press and hold 🛏	can be increased or decreased by one when
	the Home Screen is open.
• + 1	Open the MENU > META DATA (for Next
	Take) > User Scene Name screen.
	Open the MENU > META DATA (for Next
• + 2	Take) > Track Name screen.
	During recording, the • key does not need to be used.
• + 3	Open the MENU > INPUT > Trim Link screen. During recording, the • key does not need
	to be used.
	Open the MENU > META DATA (for Next
	Take) > Note screen.
• + 4	During recording, the I key does not need
	to be used.
	Clear the level meter clipping indicators.
• + 5	During recording, the skey does not need
	to be used.
	Open the L/R track fader settings screen.
• + 6	During recording, the 🔳 key does not need
	to be used.

Shortcut	Explanation
• + 7	Open the MENU > OUTPUT > Headphone > Headphone Routing screen. During recording, the • key does not need to be used.
• + 8	Use to disable the keys set with "Key Hold Target". During recording, the • key does not need to be used.
• + FFL (Track 1)	Circle the currently selected take.
• + FF (Track 2)	Open MENU > TIMECODE > Timecode screen.

Shortcut	Explanation
Press and hold 😱	Reset the selected pan/fader to the default value (when the Home Screen mixer is open). If already set to its default value, selecting a fader mutes the track.

Character input screen

Shortcut	Explanation
Press and turn	Move the cursor vertically on a character input screen keyboard.
• + ••	Delete a character on the character input screen.
• + ••	Move the cursor to "Enter" on the character input screen keyboard.

Routing screen

Shortcut	Explanation
Press and turn	Move the cursor vertically

Specifications

Recor	ding media	Dual SD card slots support 16MB–2GB SD cards, 4GB–32GB SDHC cards and 64GB–512GB SDXC cards		
	INPUT 1–8	Connectors	XLR/TRS combo jacks (XLR: 2 hot, TRS: TIP hot)	
	XLR inputs (MIC)	Input gain	+10 - +75 dB	
		Input impedance	3.3 kΩ	
		Maximum input level	+14 dBu (at 0 dBFS, limiter ON)	
		Phantom power	+24/+48V 10mA maximum for each channel	
	TRS inputs (LINE)	Input gain	-10 – +55 dB	
		Input impedance	28 kΩ	
Inputs		Maximum input level	+24 dBu (at 0 dBFS, limiter ON)	
lnp	Equivalent input noise	-127 dBu or less (A-weighted, +75dB input gain, 150Ω input)		
	Frequency characteristics	10 Hz – 80 kHz +0.5dB/–1dB (192kHz sampling rate)		
	A/D dynamic range	120 dB typ (-60dBFS input, A-weighted)		
	Crosstalk	–90 dB or less (between adjacent channels, 1kHz)		
	MIC IN	ZOOM mic capsule input (use disables Inputs 1/2)		
	SLATE MIC	Built-in mic for voice memos can be assigned to tracks freely		
	MAIN OUT 1/2	Connectors	TA3 connectors, balanced output (2: hot)	
		Output impedance	150 Ω or less	
		Reference output level	–10 dBV (normal output level), –40 dBV (mic output level), 1 kHz, 600 Ω load	
		Maximum output level	+10 dBV (normal output level), -20 dBV (mic output level), 1 kHz, 600 Ω load	
ŝ	SUB OUT 1/2	Connector	3.5mm stereo mini unbalanced output jack	
put		Output impedance	100 Ω or less	
Outputs		Reference output level	–10 dBV (normal output level), –40 dBV (mic output level), 1 kHz, 10 k\Omega load	
0		Maximum output level	+10 dBV (normal output level), -20 dBV (mic output level), 1 kHz, 10 k\Omega load	
	HEADPHONE	Connector	1/4" unbalanced stereo output jack	
		Output impedance	15 Ω or less	
		Maximum output level	100mW + 100mW (32Ω load)	
	D/A dynamic range	106 dB typ (-60dBFS input, A-weighted)		

	When WAV selected			
	Supported formats	44.1/47.952/48/48.048/88.2/96/192kHz, 16/24-bit, mono/stereo//2-10ch poly, BWF and iXML		
	Maximum simultaneous	10 (8 inputs + stereo mix)		
Descuding formats	recording tracks	8 (at 192kHz sampling rate)		
Recording formats	When MP3 selected			
	Supported formats	128/192/320kbps, 44.1/48kHz, ID3v1 tags		
	Maximum simultaneous			
	recording tracks	2		
	Using a 32GB card	Using a 32GB card		
Recording time	30:51:00 (48kHz/24-bit stereo WAV)			
	7:42:00 (192kHz/24-bit stereo WAV)			
	Connector	BNC		
	Modes	Off, Int Free Run, Int Record Run, Int RTC Run, Ext,		
	Modes	Ext Auto Rec (audio clock can be synchronized to timecode)		
	Frame rates	23.976ND, 24ND, 25ND, 29.97ND, 29.97D, 30ND, 30D		
Timecode	Precision	±0.2 ppm		
	Supported input levels	0.2 – 5.0 Vpp		
	Input impedance	4.6 kΩ		
	Output level	3.3 Vpp		
	Output impedance	50 Ω or less		
	Batteries: 8 AA			
Power supplies	AC adapter: AD-19 DC12V 2A (center plus)			
	External DC power supply: H	External DC power supply: HIROSE HR10A-7R-4S 4-pin connector (1 pin: –, 4 pin: +), 9–16 V		

Specifications (continued)

	When recording 2 channels at 48kHz/16-bit to SD1 with MAIN/SUB OUT OFF, TIME CODE OFF, LED/LCD Brightness 5, 32Ω head- phones, PHANTOM OFF		
	Alkaline batteries	8.5 hours or more	
	NiMH (2450mAh)	10 hours or more	
	Lithium batteries	12.5 hours or more	
		48kHz/24-bit to SD1 with MAIN/SUB OUT OFF, TIME CODE OFF, LED/LCD Brightness 5, 32Ω head-	
	phones, PHANTOM OFF		
Continuous recording time	Alkaline batteries	4.5 hours or more	
j	NiMH (2450mAh)	6 hours or more	
	Lithium batteries	8.5 hours or more	
	When recording 8 channels at 192kHz/24-bit to SD1 with MAIN/SUB OUT ON, TIME CODE Int Free Run, LED/LCD Brightness 60,		
	32Ω headphones, PHANTOM 48V		
	Alkaline batteries	1 hour or more	
	NiMH (2450mAh)	2 hours or more	
	Lithium batteries	3 hours or more	
Display	2.4" full-color LCD (320x240)		
	Mass storage operation		
	Class	USB 2.0 High Speed	
	Multi Track audio interface operation (driver required for Windows, not required for Mac)		
	Class	USB 2.0 High Speed	
	Specifications	44.1/48/88.2/96kHz sampling rate, 16/24-bit bit rate, 8 in/4 out	
USB	Stereo Mix audio interface operation (no driver required)		
036	Class	USB 2.0 Full Speed	
	Specifications	44.1/48kHz sampling rate, 16-bit bit rate, 2 in/2 out	
	Audio Interface with Rec: (driver required for Windows, not required for Mac)		
	Class	USB 2.0 High Speed	
	Specifications	44.1/48 kHz sampling rate, 16/24-bit bit rate, 10 in/4 out	
	Note: iOS device audio interface operation supported (stereo mode only)		
Power consumption	12 W		
External dimensions	Main unit: 7.0 in. (W) × 5.5 in. (D) × 2.1 in. (H) 178.2 mm (W) × 140.3 mm (D) × 54.3 mm (H)		
Weight (main unit only)	2.1 pounds (960 g)		

For U.S.A. FCC regulation warning

This equipment has been tested and found to comply with the limits for a Class B digital device, pursuant to Part 15 of the FCC Bules. These limits are designed to provide reasonable protection against harmful interference in a residential installation. This equipment generates, uses and can radiate radio frequency energy and, if not installed and used in accordance with the instructions. may cause harmful interference to radio communications. However, there is no guarantee that interference will not occur in a particular installation. If this equipment does cause harmful interference to radio or television reception. which can be determined by turning the equipment off and on, the user is encouraged to try to correct the interference by one or more of the following measures:

- Reorient or relocate the receiving antenna.
- Increase the separation between the equipment and receiver.
- Connect the equipment to an outlet on a circuit different from that to which the receiver is connected.
- Consult the dealer or an experienced radio/TV technician for help.

FCC CAUTION

Changes or modifications not expressly approved by the party responsible for compliance could void the user's authority to operate the equipment. This transmitter must not be co-located or operated in conjunction with any other antenna or transmitter.

For EU Countries -

Declaration of Conformity

For U.S.A. and CANADA

This device complies with part 15 of the FCC Rules and Industry Canada licenseexempt RSS standard(s). Operation is subject to the following two conditions: (1) This device may not cause harmful interference, and (2) this device must accept any interference received including interference that may cause undesired operation.

This equipment complies with FCC/IC radiation exposure limits set forth for an uncontrolled environment and meets the FCC radio frequency (RF) Exposure Guidelines and RSS-102 of the IC radio frequency (RF) Exposure rules. This equipment has very low levels of RF energy that are deemed to comply without testing of specific absorption ratio (SAR).

For CANADA

Le présent appareil est conforme aux CNR d'Industrie Canada applicables aux appareils radio exempts de licence. L'exploitation est autorisée aux deux conditions suivantes : (1) l'appareil ne doit pas produire de brouillage, et (2) l'utilisateur de l'appareil doit accepter tout brouillage radioélectrique subi. même si le brouillage est susceptible d'en compromettre le fonctionnement.

Cet équipement est conforme aux limites d'exposition aux ravonnements énoncées pour un environnement non contrôlé et respecte les règles d'exposition aux fréquences radioélectriques (RF) CNR-102 de l'IC. Cet équipement émet une énergie RF très faible qui est considérée conforme sans évaluation du débit d'absorption spécifique (DAS).

Label is located at the bottom of the unit



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