



124™

**DIGITAL LIVE
CONSOLE**

OWNER'S MANUAL ADDENDUM VERSION 1.7 UPDATE

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- Visit www.mackie.com and click Support to find: FAQs, manuals, addendums, and other useful information.
- Email us at: techmail@mackie.com.
- Telephone 1-800-898-3211 to speak with one of our splendid technical support chaps, (Monday through Friday, from 7 a.m. to 5 p.m. PST).

1. Introduction

This document describes the new features available in TT24 software upgrade version 1.7.

New Features Summary

- Full Input/Output Routing via TT Control Software.
- Front of House / Monitor Linked Consoles with a single DS3232 Digital Snake.
- Support for the optional LP48 card that offers complex Loudspeaker Processing and Insert EQ Processing.

The details of the LP48 card are not shown in this document. Please see the LP48 card user's guide, available from our website: www.mackie.com.

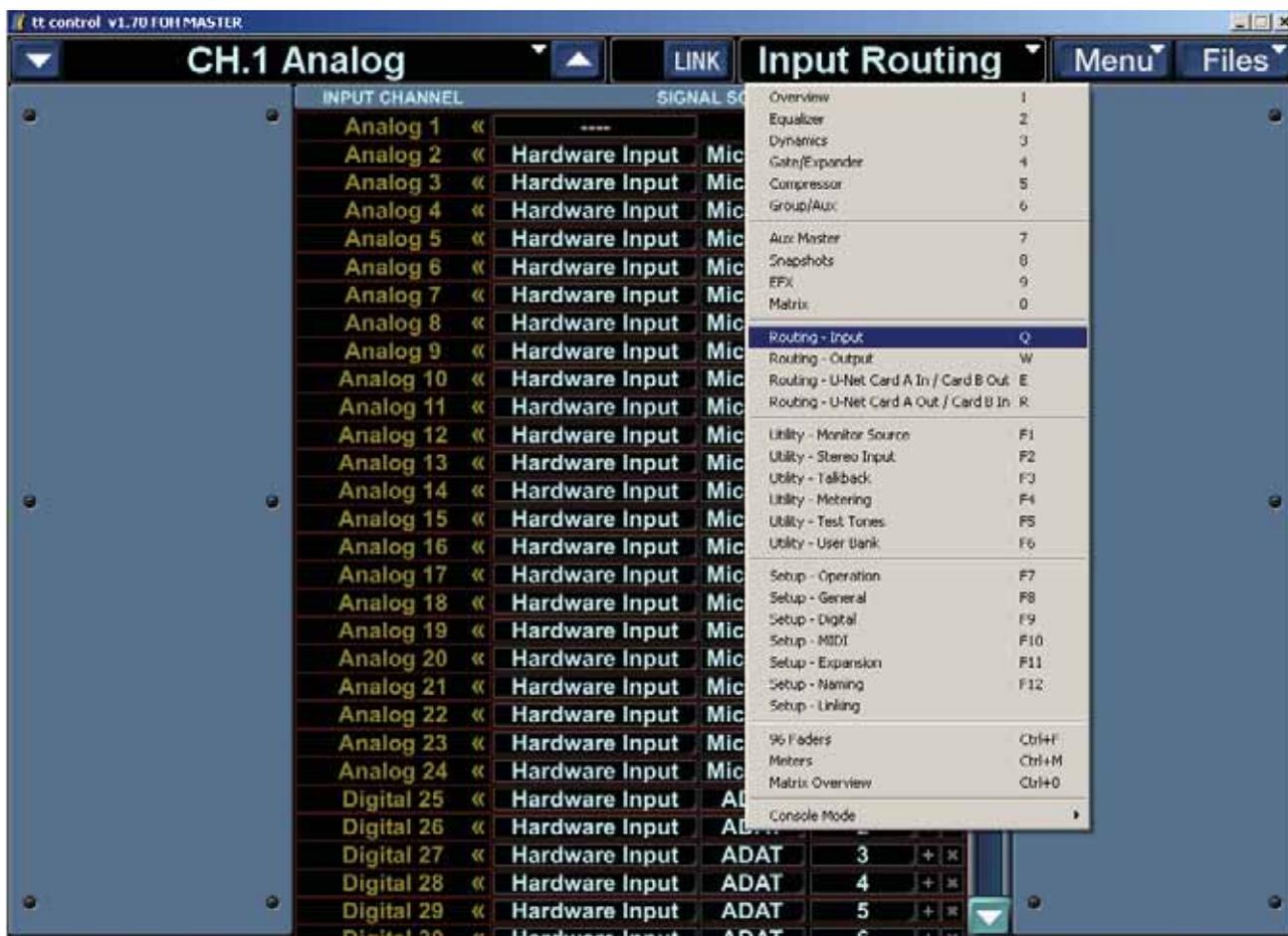
2. Full I/O Routing

A full routing matrix has been added to allow complete input and output routing to and from the input and output channels, as well as direct output routing and hardware I/O patching. This includes analog and digital inputs and outputs. In addition, you can route U-Net signals in and out of the console and between two U100 cards in a single console. These functions are only accessed via the TT Control software. Routing changes cannot be done from the control surface except by loading presets or snapshots.

The routing screens allow things such as ADAT input routing to input channels on bank 1 or Aux Output Routing to the AES output. In addition it allows detailed routing of the signals from each card slot, doing away with the basic bank-based routing used for the DS3232 prior to software version 1.7.

The routing screen is accessible from the central drop-down menu in the TT control PC application. Choose one of the four new Routing options:

- Routing – Input
- Routing – Output
- Routing – U-Net Card A In / Card B Out
- Routing – U-Net Card B In / Card A Out



The Input and Output routing screens are similar, and are comprised of a source and destination list; the destinations are listed vertically on the left column followed by rows of drop down lists to their right allowing source selection. Assign a source by selecting items from the drop down lists from left to right. This process is described in more detail in the routing sections below.



Hovering over a row will highlight the row, making navigation around the list easier. There are more destinations than can fit on the screen at once, so the list can be scrolled vertically using the scroll bar along the right side or using the mouse wheel.

Additionally, each routing row has two buttons to the right side, Clear Routing (X) and Increment Routing (+):

- Clear Routing – This clears the routing assignment for that row. It can be dragged up or down to clear the routing on adjacent rows.
- Increment Routing – When dragged up or down, this will set the next or previous row to the incrementally next or previous assignment, allowing you to set consecutive assignments with ease.

NOTE: If the computer CTRL key is held while dragging, the rows will be filled with the same source signal giving you a fast way to assign the same signal to multiple channels or outputs.

Unassigned source rows will be shown as pale red dashes to indicate they are not currently in use. Once assigned, source rows are color coded to differentiate the main source type selected. These are described for input and output routing in the sections below.

U-Net Card Slot routing screens are slightly different. They still use the same source-destination method except that both the source and destination are selectable U-Net signals. This is described in detail in U-Net Routing section on page 28.

Using a single TT24 and a DS3232 Digital Snake

If you have a single TT24 console and a DS3232 digital snake, then you need only be concerned with the Input and Output Routing described below. There is generally no need to look into the U-Net Card A In / Card B Out, and U-Net Card B In / Card A Out routing, unless you are linking multiple consoles. Even then, there are factory presets that will cover most typical systems.

Routing Presets

The following sections on Input Routing, Output Routing, and U-Net Routing are shown in some detail, with screenshots thrown in from all directions.

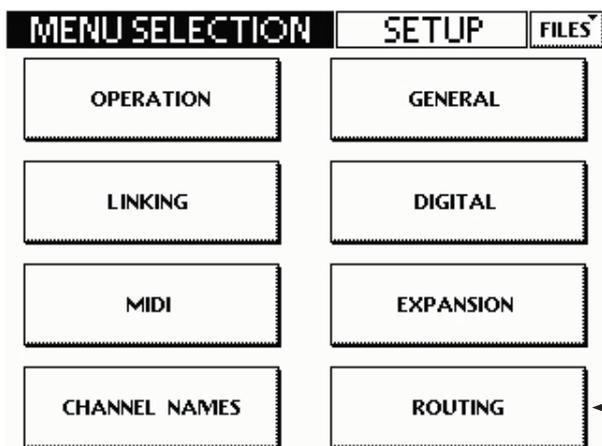
To make it easier for you, we have created routing presets that cover typical applications. You may find that the presets do all that you need, so you do not have to fret over the details.

Routing Presets are stored on the console and can be loaded from the console but not saved. They can be both saved and loaded from the TT Control software. This is because routing changes can only be made from the control software. In the TT Control software, routing presets are saved and loaded from the File menu as usual.

These presets are available in four types from the TT Control FILES dropdown menu, and from the console's routing setup screen:

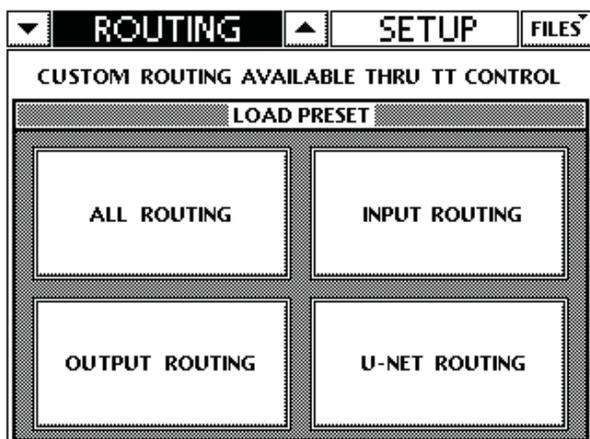
- Global All Routing: Presets that affect all routing (Input, Output, and U-Net)
- Input Routing: Presets that affect the Input Routing
- Output Routing: Presets that affect the Output Routing
- U-Net Routing: Presets that affect the U-Net Routing

You can also save your own presets using TT Control, and then recall these from the console as needed.



TT24 Console Setup Screen

Press "Routing" to bring up the Routing Setup Screen.



TT24 Console Routing Setup Screen

Press these to bring up a list of available presets for each one.

Factory Global All Routing Presets

Preset	Description
Default	Default input, output, and U-Net routing
Snake A Analog	DS3232 connected to a U100 in card slot A, routed to the Analog bank
Snake A Digital	DS3232 connected to a U100 in card slot A, routed to the Digital bank
Snake B Analog	DS3232 connected to a U100 in card slot B, routed to the Analog bank
Snake B Digital	DS3232 connected to a U100 in card slot B, routed to the Digital bank
MON Snake LP48	DS3232 connected to a U100 in card slot B, routed to the Analog bank and to FOH via a second U100 in card slot A. Aux sends are routed back to the snake. Configure console as MON Master on setup>linking screen.
FOH Snake LP48	FOH console linked to MON via a U100 in card slot A. DS3232 is routed to the Analog bank through the monitor console. An LP48 in card slot B is routed back to the snake along with Main and Group outputs. Configure console as FOH Master on setup>linking screen.

Factory Input Routing Presets

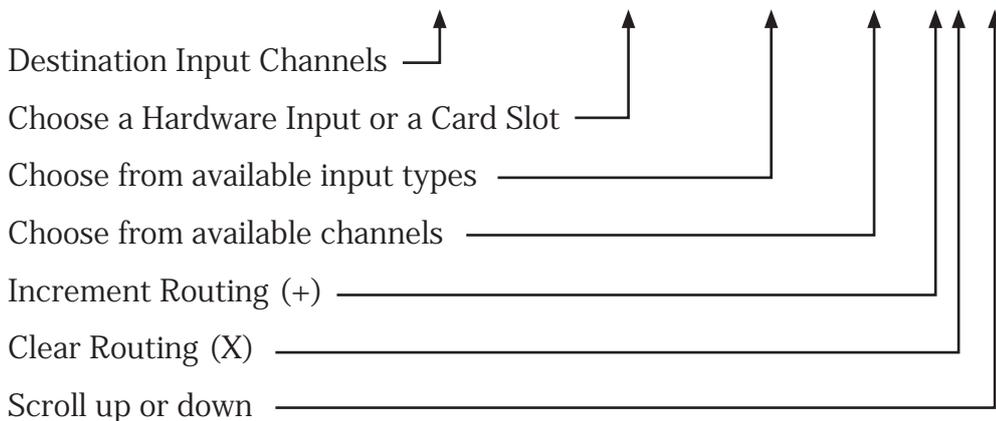
Preset	Description
Default	Default input routing
ADAT B1 Mic B2	ADAT inputs routed to Analog bank and Mic inputs routed to Digital bank
Mic B1 B2	Mic inputs routed to Analog and Digital banks for independant control of FOH and monitor mixes from a single console
ADAT B1 B2	ADAT inputs routed to Analog and Digital banks for independant control of FOH and monitor mixes from a single console

Factory Output Routing Presets

Preset	Description
Default	Default output routing. Group outputs 1-8 to rear panel Group/Matrix outputs.
Default Matrix	Default output routing. Matrix outputs A-H to rear panel Group/Matrix outputs.
Default LP48 A	Default output routing. LP48 in card slot A, speaker processor outputs 1-8 to rear panel Group/Matrix outputs.
Default LP48 B	Default output routing. LP48 in card slot B, speaker processor outputs 1-8 to rear panel Group/Matrix outputs.
Pre DSP ADAT Out	ADAT outputs routed from the Pre DSP channel signal
ADAT In to Out	ADAT inputs routed directly to the ADAT outputs for recording
Aux on Group XLR	Aux 1-8 routed to the rear panel Group/Matrix XLR outputs for easy connection to an analog snake

2.1 Input Routing

The Input Routing screen shows a routing list displaying Destination Input Channels down the left column. Choose Hardware Inputs or Card Slot signals to be routed to one or more input channels.



The vertical list along the left shows input channels which will be listed by their default name and custom name (if assigned). Double clicking a channel will allow you to edit the channel custom name. These names are in a single column with the following contents:

Destination Input Channels	Notes
Analog 1-24	Full DSP when assigned to Analog Bank
Digital 25-48	Full DSP when assigned to Digital Bank or when UFXII installed
Line In 1-8	EQ Only
Card 1-8	Full DSP

Input Routing Table

Signal Sources		
Drop Down 1	Drop Down 2	Drop Down 3
Hardware Input	Mic/Line	1-24
	ADAT	1-24 (44.1 or 48k) 1-12 (88.2 or 96k)
	Line	1-8
	AES/SPDIF (one will be shown depending on the selection in setup>digital)	L-R
	Tape	A L-R, B L-R
	Talkback	In
Card Slot	A - Card Type (e.g. U100)	1-32 (note LP48 shows additional detail, see below)
	B - Card Type (e.g. U100)	1-32 (note LP48 shows additional detail, see below)

Each Source Row is color coded according to the selection made in the Drop Down 1 source type:

- Hardware Input – Pale Blue (a very-faintly blue-tint of white)
- Card Slot - Pale Yellow

For each destination input you can select the source to be either a Hardware Input or a Card Slot:

INPUT CHANNEL	Source	
Analog 1 - SAX 1	Hardware Input	1
Analog 2	Card Slot	2
Analog 3	Hardware Input Mic/Line	3
Analog 4	Hardware Input Mic/Line	4
Analog 5	Hardware Input Mic/Line	5
Analog 6	Hardware Input Mic/Line	6
Analog 7	Hardware Input Mic/Line	7

Choose from Hardware Input or Card Slot

For the Hardware Input, you can choose the source type:

INPUT CHANNEL	SIGN	Type	
Analog 1 - SAX 1	«	Mic/Line	1
Analog 2	«	ADAT	2
Analog 3	«	Line	3
Analog 4	«	SPDIF	4
Analog 5	«	Tape	5
Analog 6	«	Talkback	6
Analog 7	«	Mic/Line	7

Choose from Mic/Line, ADAT, Line, SPDIF, Tape, or Talkback

and choose from the "Instance" or available channels from that source type:

AL SOURCE	Instance
Mic/Line	1
Mic/Line	2
Mic/Line	3
Mic/Line	4
Mic/Line	5
Mic/Line	6
Mic/Line	7
Mic/Line	8
Mic/Line	9
Mic/Line	10
Mic/Line	11
Mic/Line	12
Mic/Line	13
Mic/Line	14
Mic/Line	15
Mic/Line	16
Mic/Line	17
Mic/Line	18
Mic/Line	19
Mic/Line	20
Mic/Line	21
Mic/Line	22
Mic/Line	23
Mic/Line	24

Mic/Line has 1 to 24 choices

AL SOURCE	Instance
ADAT	1
Mic/Line	2
Mic/Line	3
Mic/Line	4
Mic/Line	5
Mic/Line	6
Mic/Line	7
Mic/Line	8
Mic/Line	9
Mic/Line	10
Mic/Line	11
Mic/Line	12
Mic/Line	13
Mic/Line	14
Mic/Line	15
Mic/Line	16
Mic/Line	17
Mic/Line	18
Mic/Line	19
Mic/Line	20
Mic/Line	21
Mic/Line	22
Mic/Line	23
Mic/Line	24

ADAT has 1 to 12 choices at 88.2 or 96k, and 1-24 at 44.1 or 48k

AL SOURCE	Instance
SPDIF	L
Mic/Line	R
Mic/Line	3

SPDIF has L, R choices. (Shows AES if configured.)

AL SOURCE	Instance
Talkback	In
Mic/Line	2

Talkback has IN only

AL SOURCE	Instance
Tape	A L
Mic/Line	A R
Mic/Line	B L
Mic/Line	B R
Mic/Line	5

Tape has A L, A R, B L and B R choices

For the Card Slot, you can select from the installed cards:

INPUT CHANNEL	SIGNAL SOURCE	Type	
Analog 1 - SAX 1	Card Slot	A - UNet	
Analog 2	Hardware Input	B - LP48	2 + x
Analog 3	Hardware Input	Mic/Line	3 + x
Analog 4	Hardware Input	Mic/Line	4 + x
Analog 5	Hardware Input	Mic/Line	5 + x

Choose from UNet card or, in this example, an LP48 card

For the UNet Card Slot, you can choose from 1 to 32:

INPUT CHANNEL	SIGNAL SOURCE	Instance
Analog 1 - SAX 1	Card Slot	A - UNet 1
Analog 2	Hardware Input	Mic/Line 2
Analog 3	Hardware Input	Mic/Line 3
Analog 4	Hardware Input	Mic/Line 4
Analog 5	Hardware Input	Mic/Line 5
Analog 6	Hardware Input	Mic/Line 6
Analog 7	Hardware Input	Mic/Line 7
Analog 8	Hardware Input	Mic/Line 8
Analog 9	Hardware Input	Mic/Line 9
Analog 10	Hardware Input	Mic/Line 10
Analog 11	Hardware Input	Mic/Line 11
Analog 12	Hardware Input	Mic/Line 12
Analog 13	Hardware Input	Mic/Line 13
Analog 14	Hardware Input	Mic/Line 14
Analog 15	Hardware Input	Mic/Line 15
Analog 16	Hardware Input	Mic/Line 16
Analog 17	Hardware Input	Mic/Line 17
Analog 18	Hardware Input	Mic/Line 18
Analog 19	Hardware Input	Mic/Line 19
Analog 20	Hardware Input	Mic/Line 20
Analog 21	Hardware Input	Mic/Line 21
Analog 22	Hardware Input	Mic/Line 22
Analog 23	Hardware Input	Mic/Line 23
Analog 24	Hardware Input	Mic/Line 24
Digital 25	Hardware Input	Line 25
Digital 26	Hardware Input	Line 26
Digital 27	Hardware Input	Line 27
Digital 28	Hardware Input	Line 28
Digital 29	Hardware Input	Line 29
Digital 30	Hardware Input	Line 30
		31
		32

Choose from Instances 1 to 32

The following section describes the options if you choose the LP48 card. If you don't have one, skip the next two pages.

LP48 Card Routing

INPUT CHANNEL	SIGNAL SOURCE	Instance
Analog 1	Card Slot B - LP48	1 - Insert 1
Analog 2	Hardware Input Mic/Line	2 - Insert 2
Analog 3	Hardware Input Mic/Line	3 - Insert 3
Analog 4	Hardware Input Mic/Line	4 - Insert 4
Analog 5	Hardware Input Mic/Line	5 - Insert 5
Analog 6	Hardware Input Mic/Line	6
Analog 7	Hardware Input Mic/Line	7
Analog 8	Hardware Input Mic/Line	8
Analog 9	Hardware Input Mic/Line	9 - Spkr 1
Analog 10	Hardware Input Mic/Line	10 - Spkr 2
Analog 11	Hardware Input Mic/Line	11 - Spkr 3
Analog 12	Hardware Input Mic/Line	12 - Spkr 4
Analog 13	Hardware Input Mic/Line	13
Analog 14	Hardware Input Mic/Line	14
Analog 15	Hardware Input Mic/Line	15
Analog 16	Hardware Input Mic/Line	16
Analog 17	Hardware Input Mic/Line	17 - Insert 6
Analog 18	Hardware Input Mic/Line	18 - Insert 7
Analog 19	Hardware Input Mic/Line	19 - Insert 8
Analog 20	Hardware Input Mic/Line	20 - Insert 9
Analog 21	Hardware Input Mic/Line	21 - Insert 10
Analog 22	Hardware Input Mic/Line	22
Analog 23	Hardware Input Mic/Line	23
Analog 24	Hardware Input Mic/Line	24
Digital 25	Hardware Input Line	25 - Spkr 5
Digital 26	Hardware Input Line	26 - Spkr 6
Digital 27	Hardware Input Line	27 - Spkr 7
Digital 28	Hardware Input Line	28 - Spkr 8
Digital 29	Hardware Input Line	29
Digital 30	Hardware Input Line	30
Digital 31	Hardware Input Line	31
Digital 32	Hardware Input Line	32

If you choose the LP48 card as the source, then the available Instances to choose from will depend on the configuration of the LP48 card.

The LP48 card has three different modes of operation:

10 Insert EQs

4 x 8 Loudspeaker Processor

5 Insert EQs and 2 x 4 Loudspeaker Processor (This is a split mode of the above two modes.)

Insert EQs

In Insert EQ mode, the signals assigned to the LP48 card go into the card, are EQ'd and return to the insert point only. They are not available for any other routing. The output from the LP48 card shows greyed-out labels for the inserts.

Loudspeaker Processors

In Loudspeaker Processor mode, there will either be 4 outputs (in LP48 split mode) or 8 outputs (in 4 x 8 mode).

The available outputs from the Loudspeaker Processor are the outputs from each crossover, such as a low, high for a two-way crossover, low, mid, and high for a three-way, and a low, low-mid, high-mid, and high for a four way crossover.

These can be routed to any input channel, or to any physical output,

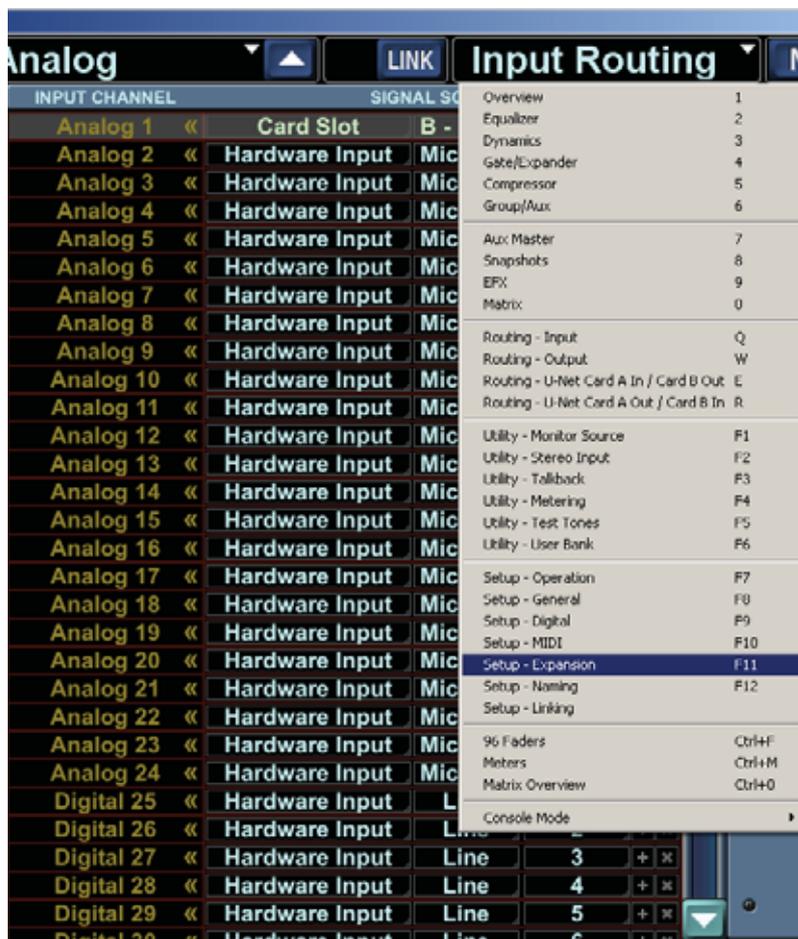
Instance	Instance
1 - Insert 1	1 - Insert 1
2 - Insert 2	2 - Insert 2
3 - Insert 3	3 - Insert 3
4 - Insert 4	4 - Insert 4
5 - Insert 5	5 - Insert 5
6	6
7	7
8	8
9 - Spkr 1	9 - Spkr 1
10 - Spkr 2	10 - Spkr 2
11 - Spkr 3	11 - Spkr 3
12 - Spkr 4	12 - Spkr 4
13	13
14	14
15	15
16	
17 - Insert 6	
18 - Insert 7	
19 - Insert 8	
20 - Insert 9	
21 - Insert 10	
22	
23	
24	
25 - Spkr 5	
26 - Spkr 6	
27 - Spkr 7	
28 - Spkr 8	
29	
30	
31	
32	

LP48 in Split mode

LP48 in 4 x 8 mode

Changing the mode of the LP48 Card

To change the mode of the LP48 card use the Expansion menu, or press F11:



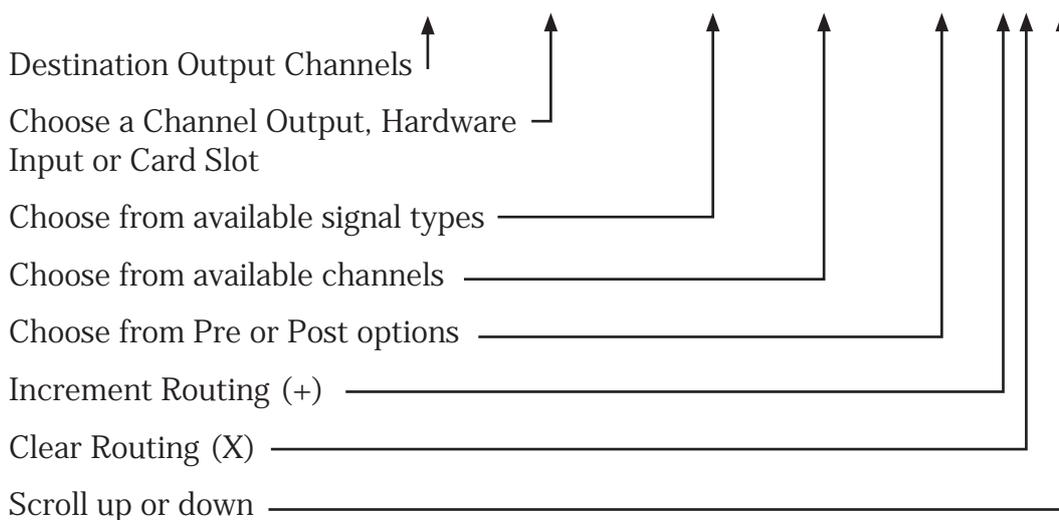
Select Inputs here

Crossover Outputs (4 x 8 mode shown)

Select Mode here

2.2 Output Routing

The Output Routing screen shows a routing list displaying Destination Hardware Outputs down in the left column. Choose from the Channel Outputs, Hardware Inputs, or Card Slot signals to be routed to one or more hardware outputs.



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The vertical list along the left shows hardware outputs. These names appear in a single column with the following contents:

Destination Hardware Outputs	Notes
Aux 1-12	
Group/Matrix 1-8	
Main L, R, C	
Monitor L, R, C	
Tape A L, R	
AES/SPDIF L, R	
ADAT 1-24	
Card A – Card Name 1-32	The card name is replaced with the name of the card in the slot if one is present
Card B – Card Name 1-32	The card name is replaced with the name of the card in the slot if one is present

Choose sources for each output by selecting an item from each of four drop down lists:

Signal Sources				
Drop Down 1	Drop Down 2	Drop Down 3	Drop Down 4	
Channel Output	Main	L, R, C (or M)	Pre DSP, Post Fader	
	Aux	1-12	Pre DSP, Post Fader	
	Group	1-8	Post Fader	
	Matrix	A-H	Pre Delay, Post Delay	
	Monitor	L-R		
	Ch Bank 1	1-24	Pre DSP, Pre Fader	
	Ch Bank 2	25-48	Pre DSP, Pre Fader	
	Ch Line	1-8	Pre DSP, Pre Fader	
	Ch FX	1 L-4 R	Pre Fader	
	Ch Card	1-8	Pre DSP, Pre Fader	
	Hardware Input	Mic/Line	1-24	-
		ADAT	1-24 (44.1 or 48k)	1-12 if sample rate is 88.2 or 96k
		Line	1-8	-
AES/SPDIF		L-R	AES or SPDIF, whichever is installed	
Tape		A L-R, B L-R	-	
Talkback		In		
Card Slot	A - Card Type (e.g. U100)	1-32 (note LP48 shows additional detail, see below)	-	
	B - Card Type (e.g. U100)	1-32 (note LP48 shows additional detail, see below)	-	

Each source row is faintly color-coded according to the selection made in the Drop Down 1 source type:

- Channel Output - Pale Blue-ish white
- Hardware Input – Pale Orange
- Card Slot - Pale Yellow

Output Routing Examples

For each destination output, you can select the source to be either a Channel Output, Hardware Input, or a Card Slot.



Choose from Channel Output, Hardware Input or Card Slot

Channel Outputs



If you choose a Channel Output, you can select from the following Types:

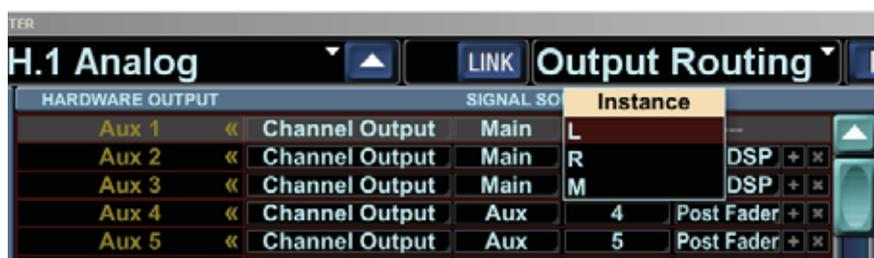
Choose from Main, Aux, Group, Matrix, Monitor, Ch Bank 1, Ch Bank 2, Ch Line, Ch FX, and Ch Card. Each of these has other options, shown on the next few pages of this guide.

Channel Outputs: Main

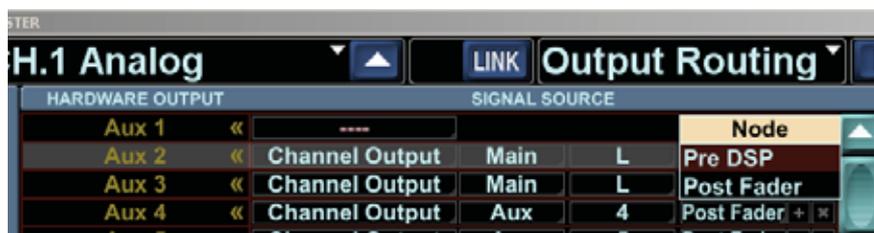
If you choose Main from the channel output options, you can select from L, R or M, and each can be Pre DSP or Post Fader:



Choose Main

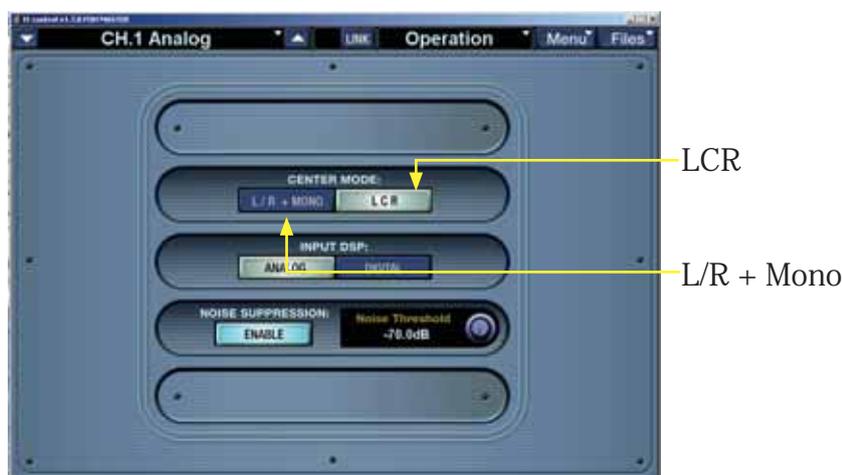


Choose L, R, and M (or L, R, C)*

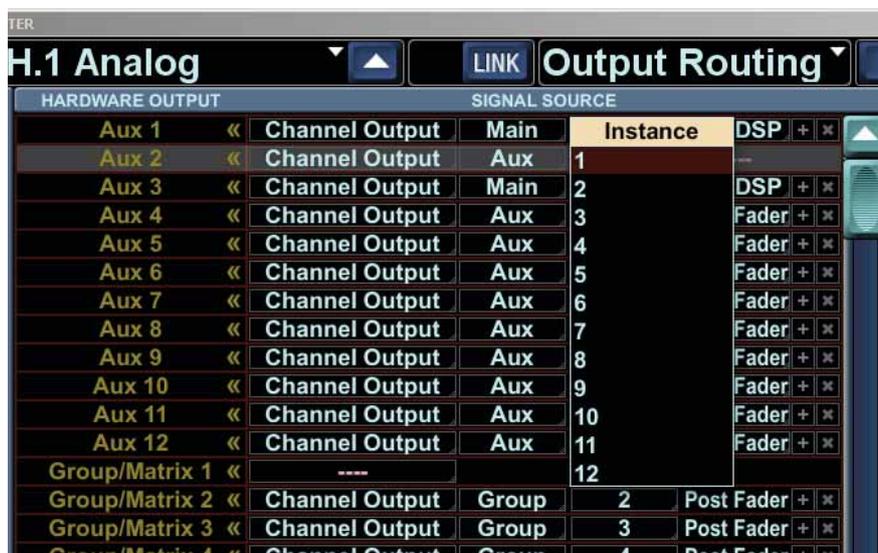


Choose Pre DSP or Post Fader

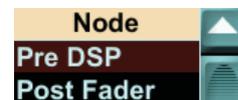
* Instead of L, R, and M, it could be L, R, and C depending upon the setting in the Operations menu:



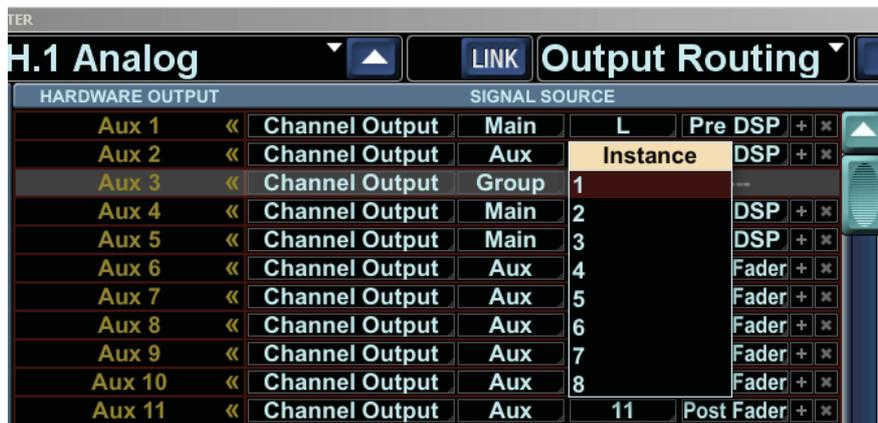
Channel Outputs: Aux



Aux: Choices 1 to 12, and each can be Pre DSP or Post Fader:

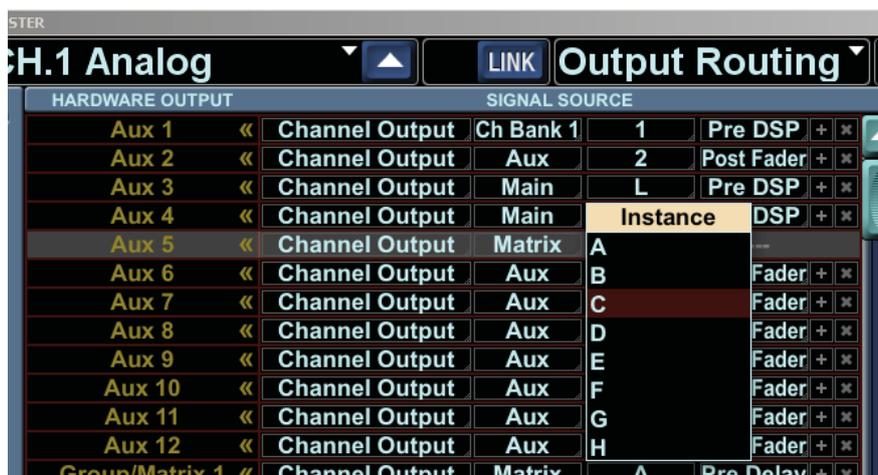


Channel Outputs: Group

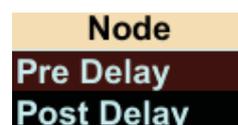


Group: Choices 1 to 8, Post Fader. Note that these Group options will be greyed-out if Matrix is enabled in the console.

Channel Outputs: Matrix



Matrix: Choices A to H, and each can be Pre Delay or Post Delay: Note that these Matrix options will be greyed-out if Group is enabled in the console.



Note: The Group/Matrix hardware outputs are changed automatically when enabling or disabling the matrix if they are set to the defaults. If they have been changed to a custom configuration, they will not be changed as the matrix is enabled or disabled.

Channel Outputs: Monitor

HARDWARE OUTPUT		SIGNAL SOURCE		
Aux 1	Channel Output	Main	L	Pre DSP
Aux 2	Channel Output	Aux	1	Pre DSP
Aux 3	Channel Output	Group	1	Post Fader
Aux 4	Channel Output	Group	1	Post Fader
Aux 5	Channel Output	Monitor	Instance	
Aux 6	Channel Output	Monitor	L	
Aux 7	Channel Output	Aux	R	Fader
Aux 8	Channel Output	Aux	8	Post Fader

Monitor: Choices L and R

Channel Outputs: Channel Bank 1

HARDWARE OUTPUT		SIGNAL SOURCE		
Aux 1	Channel Output	Main	L	Pre DSP
Aux 2	Channel Output	Aux	1	Pre DSP
Aux 3	Channel Output	Group	1	Post Fader
Aux 4	Channel Output	Group	1	Post Fader
Aux 5	Channel Output	Group	2	Post Fader
Aux 6	Channel Output	Group	Instance	Fader
Aux 7	Channel Output	Ch Bank 1	1	
Aux 8	Channel Output	Aux	2	Fader
Aux 9	Channel Output	Aux	3	Fader
Aux 10	Channel Output	Aux	4	Fader
Aux 11	Channel Output	Aux	5	Fader
Aux 12	Channel Output	Aux	6	Fader
Group/Matrix 1	Channel Output	Group	7	Fader
Group/Matrix 2	Channel Output	Group	8	Fader
Group/Matrix 3	Channel Output	Group	9	Fader
Group/Matrix 4	Channel Output	Group	10	Fader
Group/Matrix 5	Channel Output	Group	11	Fader
Group/Matrix 6	Channel Output	Group	12	Fader
Group/Matrix 7	Channel Output	Group	13	Fader
Group/Matrix 8	Channel Output	Group	14	Fader
Main L	Channel Output	Main	15	Fader
Main R	Channel Output	Main	16	Fader
Main C	Channel Output	Main	17	Fader
Monitor L	Channel Output	Monitor	18	
Monitor R	Channel Output	Monitor	19	
Tape A L	Channel Output	Main	20	DSP
Tape A R	Channel Output	Main	21	DSP
AES/SPDIF L	Channel Output	Main	22	DSP
AES/SPDIF R	Channel Output	Main	23	DSP
ADAT 1	Channel Output	Ch Bank 1	24	Fader

Channel Bank 1: Choices 1 to 24.
Each can be Pre DSP or Pre Fader:

Node
Pre DSP
Pre Fader

Channel Outputs: Channel Bank 2

HARDWARE OUTPUT	SIGNAL SOURCE	Instance	Node
Aux 1	Channel Output Main	L	Pre DSP
Aux 2	Channel Output Aux		DSP
Aux 3	Channel Output Ch Bank 2	25	
Aux 4	Channel Output Group	26	Fader
Aux 5	Channel Output Group	27	Fader
Aux 6	Channel Output Group	28	Fader
Aux 7	Channel Output Group	29	Fader
Aux 8	Channel Output Group	30	Fader
Aux 9	Channel Output Group	31	Fader
Aux 10	Channel Output Aux	32	Fader
Aux 11	Channel Output Aux	33	Fader
Aux 12	Channel Output Aux	34	Fader
Group/Matrix 1	Channel Output Group	35	Fader
Group/Matrix 2	Channel Output Group	36	Fader
Group/Matrix 3	Channel Output Group	37	Fader
Group/Matrix 4	Channel Output Group	38	Fader
Group/Matrix 5	Channel Output Group	39	Fader
Group/Matrix 6	Channel Output Group	40	Fader
Group/Matrix 7	Channel Output Group	41	Fader
Group/Matrix 8	Channel Output Group	42	Fader
Main L	Channel Output Main	43	Fader
Main R	Channel Output Main	44	Fader
Main C	Channel Output Main	45	Fader
Monitor L	Channel Output Monitor	46	
Monitor R	Channel Output Monitor	47	
Tape A L	Channel Output Main	48	DSP
Tape A R	Channel Output Main	R	Pre DSP
AFS/SPDIF I	Channel Output Main	L	Pre DSP

Channel Bank 2: Choices 25 to 48.
Each can be Pre DSP or Pre Fader:

Node
Pre DSP
Pre Fader

Channel Outputs: Channel Line

HARDWARE OUTPUT	SIGNAL SOURCE	Instance	Node
Aux 1	Channel Output Main	L	Pre DSP
Aux 2	Channel Output Aux	1	Pre DSP
Aux 3	Channel Output Aux	2	Pre DSP
Aux 4	Channel Output Aux	3	Pre DSP
Aux 5	Channel Output Aux	4	Pre DSP
Aux 6	Channel Output Aux		DSP
Aux 7	Channel Output Ch Line	1	
Aux 8	Channel Output Aux	2	DSP
Aux 9	Channel Output Aux	3	DSP
Aux 10	Channel Output Aux	4	DSP
Aux 11	Channel Output Aux	5	DSP
Aux 12	Channel Output Aux	6	DSP
Group/Matrix 1	Channel Output Aux	7	DSP
Group/Matrix 2	Channel Output Group	8	Fader
Group/Matrix 3	Channel Output Group	3	Post Fader

Channel Line: Choices 1 to 8.
Each can be Pre DSP or Pre Fader:

Node
Pre DSP
Pre Fader

Channel Outputs: Channel FX

HARDWARE OUTPUT	SIGNAL SOURCE	Instance	Node
Aux 1	Channel Output Main	L	Pre DSP
Aux 2	Channel Output Aux	1	Pre DSP
Aux 3	Channel Output Aux	2	Pre DSP
Aux 4	Channel Output Aux	3	Pre DSP
Aux 5	Channel Output Aux		DSP
Aux 6	Channel Output Ch FX	1 L	
Aux 7	Channel Output Aux	1 R	DSP
Aux 8	Channel Output Aux	2 L	DSP
Aux 9	Channel Output Aux	2 R	DSP
Aux 10	Channel Output Aux	3 L	DSP
Aux 11	Channel Output Aux	3 R	DSP
Aux 12	Channel Output Aux	4 L	DSP
Group/Matrix 1	Channel Output Aux	4 R	DSP
Group/Matrix 2	Channel Output Group	2	Post Fader

Channel FX: Choices 1 L, 1 R, 2 L, 2 R, 3 L, 3 R, and 4 L, and 4 R.
Each is Pre Fader.

Channel Outputs: Channel Card

HARDWARE OUTPUT	SIGNAL SOURCE	Instance	Node
Aux 1	Channel Output Main L	Pre DSP	Pre DSP
Aux 2	Channel Output Aux 1	Pre DSP	Pre DSP
Aux 3	Channel Output Aux 2	Pre DSP	Pre DSP
Aux 4	Channel Output Aux 3	Pre DSP	Pre DSP
Aux 5	Channel Output Aux	DSP	Pre DSP
Aux 6	Channel Output Ch Card 1	--	Pre DSP
Aux 7	Channel Output Aux 2	DSP	Pre DSP
Aux 8	Channel Output Aux 3	DSP	Pre DSP
Aux 9	Channel Output Aux 4	DSP	Pre DSP
Aux 10	Channel Output Aux 5	DSP	Pre DSP
Aux 11	Channel Output Aux 6	DSP	Pre DSP
Aux 12	Channel Output Aux 7	DSP	Pre DSP
Group/Matrix 1	Channel Output Aux 8	DSP	Pre DSP
Group/Matrix 2	Channel Output Group 2	Post Fader	Pre DSP

Channel Card: Choices 1 to 8.
Each can be Pre DSP or Pre Fader:

Node
Pre DSP
Pre Fader

Hardware Input

If you choose a Hardware Input instead of a Channel Output:

HARDWARE OUTPUT	Type	Node
Aux 1	Hardware Input Mic/Line	--
Aux 2	Channel Output ADAT 2	Post Fader
Aux 3	Channel Output Line 3	Post Fader
Aux 4	Channel Output SPDIF 4	Post Fader
Aux 5	Channel Output Tape 5	Post Fader
Aux 6	Channel Output Talkback 6	Post Fader
Aux 7	Channel Output Aux 7	Post Fader
Aux 8	Channel Output Aux 8	Post Fader
Aux 9	Channel Output Aux 9	Post Fader
Aux 10	Channel Output Aux 10	Post Fader
Aux 11	Channel Output Aux 11	Post Fader
Aux 12	Channel Output Aux 12	Post Fader
Group/Matrix 1	Channel Output Group 1	Post Fader
Group/Matrix 2	Channel Output Group 2	Post Fader
Group/Matrix 3	Channel Output Group 3	Post Fader
Group/Matrix 4	Channel Output Group 4	Post Fader
Group/Matrix 5	Channel Output Matrix 5	Pre Delay
Group/Matrix 6	Channel Output Matrix 6	Pre Delay
Group/Matrix 7	Channel Output Matrix 7	Pre Delay
Group/Matrix 8	Channel Output Matrix 8	Pre Delay
Main L	Channel Output Main L	Post Fader
Main R	Channel Output Main R	Post Fader
Main C	Channel Output Main M	Post Fader
Monitor L	Channel Output Monitor L	+ x
Monitor R	Channel Output Monitor R	+ x
Tape A L	Channel Output Main L	Pre DSP
Tape A R	Channel Output Main R	Pre DSP
AES/SPDIF L	Channel Output Main L	Pre DSP
AES/SPDIF R	Channel Output Main R	Pre DSP

Hardware Input: Choices are Mic/Line, ADAT, Line, SPDIF, Tape, and Talkback.

Each has its own menus, and these are described in the next few pages:

Hardware Input: Mic/Line

HARDWARE OUTPUT	SIGNAL SOURCE	Instance	
Aux 1	Hardware Input Mic/Line	1	
Aux 2	Channel Output Aux	2	DSP + x
Aux 3	Channel Output Aux	3	DSP + x
Aux 4	Channel Output Aux	4	DSP + x
Aux 5	Channel Output Aux	5	DSP + x
Aux 6	Channel Output Ch Card	6	DSP + x
Aux 7	Channel Output Aux	7	DSP + x
Aux 8	Channel Output Aux	8	DSP + x
Aux 9	Channel Output Aux	9	DSP + x
Aux 10	Channel Output Aux	10	DSP + x
Aux 11	Channel Output Aux	11	DSP + x
Aux 12	Channel Output Aux	12	DSP + x
Group/Matrix 1	Channel Output Aux	13	DSP + x
Group/Matrix 2	Channel Output Group	14	Fader + x
Group/Matrix 3	Channel Output Group	15	Fader + x
Group/Matrix 4	Channel Output Group	16	Fader + x
Group/Matrix 5	Channel Output Group	17	Fader + x
Group/Matrix 6	Channel Output Group	18	Fader + x
Group/Matrix 7	Channel Output Group	19	Fader + x
Group/Matrix 8	Channel Output Group	20	Fader + x
Main L	Channel Output Main	21	Fader + x
Main R	Channel Output Main	22	Fader + x
Main C	Channel Output Main	23	Fader + x
Monitor L	Channel Output Monitor	24	+ x
Monitor R	Channel Output Monitor		+ x

Mic/Line: Choices 1 to 24.

Hardware Input: ADAT

HARDWARE OUTPUT	SIGNAL SOURCE	Instance	
Aux 1	Hardware Input Mic/Line	1	+ x
Aux 2	Hardware Input Mic/Line		+ x
Aux 3	Hardware Input ADAT	1	+ x
Aux 4	Channel Output Aux	2	DSP + x
Aux 5	Channel Output Aux	3	DSP + x
Aux 6	Channel Output Ch Card	4	DSP + x
Aux 7	Channel Output Aux	5	DSP + x
Aux 8	Channel Output Aux	6	DSP + x
Aux 9	Channel Output Aux	7	DSP + x
Aux 10	Channel Output Aux	8	DSP + x
Aux 11	Channel Output Aux	9	DSP + x
Aux 12	Channel Output Aux	10	DSP + x
Group/Matrix 1	Channel Output Aux	11	DSP + x
Group/Matrix 2	Channel Output Group	12	Fader + x
Group/Matrix 3	Channel Output Group	13	Fader + x
Group/Matrix 4	Channel Output Group	14	Fader + x
Group/Matrix 5	Channel Output Group	15	Fader + x
Group/Matrix 6	Channel Output Group	16	Fader + x
Group/Matrix 7	Channel Output Group	17	Fader + x
Group/Matrix 8	Channel Output Group	18	Fader + x
Main L	Channel Output Main	19	Fader + x
Main R	Channel Output Main	20	Fader + x
Main C	Channel Output Main	21	Fader + x
Monitor L	Channel Output Monitor	22	+ x
Monitor R	Channel Output Monitor	23	+ x
Tape A L	Channel Output Main	24	DSP + x
Tape A R	Channel Output Main		DSP + x

ADAT: Choices 1 to 12 if the sample rate is 88.2 or 96k, and 1 to 24 if the sample rate is 44.1 or 48k.

See ADAT card notes on page 27 for more details.

Hardware Input: Line



Line: Choices 1 to 8.

Hardware Input: SPDIF



SPDIF: Choices L, R.

Hardware Input: Tape



Tape: A L, A R, B L, B R.

Hardware Input: Talkback



Talkback: In.

Card Slot

If you choose a Card Slot, you can select from the available cards:



Card Slot: Here we can choose between the LP48 card or the UNet card.

Card Slot: UNet Card



UNet card: Choices 1 to 32.

Card Slot: LP48 Card



LP48 card: If the LP48 card has been configured as a 4x8 Loudspeaker Processor, there will be 8 outputs available as shown. These represent the outputs of the various crossovers of the LP48. The Insert EQs are greyed-out and not available for selection.

Note that the inputs to the LP48 card are chosen in the Setup>Expansion screen.



LP48 card: If the LP48 card has been configured as a 2x4 Loudspeaker Processor, with 5 Insert EQs, there will be 4 outputs available as shown. These represent the outputs of the various crossovers. The Insert EQs are greyed-out and not available for selection. (The EQ'd signals return to the insert point they came from.)

See additional details described in the input routing screen on page 12.

Card Slot Routing Notes

Routing to and from I/O cards is handled by signal paths. The TT24 supports a maximum of 32 input and output signal paths to/from each card slot. But, some cards (LP48) have less than 32 inputs/outputs. Routing signals to these cards on these higher number signal paths will have no effect, but because the routing is stored in the console, we support routing to all signal path numbers no matter which card is installed.

If a card is installed that does not support all 32 signal paths, the unused signal path numbers are shown in grey indicating that they are still routable but that they are unused. Routing to/from these greyed out signal paths will therefore have no effect although you can still make the routing connections. Similarly, if the console does not have a card in slot A and/or B, the signal paths for the entire card slot will appear in grey again indicating that they are still routable although no audio will pass. Finally, some cards (LP48) have multiple operating modes and each mode may have a different number of supported signal paths. Therefore as the mode is changed, the number of available versus greyed-out source and destination signal path numbers will change.

The above method allows for cards to be changed while keeping routings intact. Existing routings will just use the corresponding input/output from the newly added card. This gives consistent and reliable operation that you can easily understand and change when a new card is installed and you hear audio you are not expecting.

This operation is described for each card type and mode combination in the table below; note that each card slot is handled independently:

Card Type	Mode	Available Card Signal Path Sources	Available Card Signal Path Destinations
UFXII	Channel 25-48 DSP	None (routing done automatically)	None (routing done automatically)
LP48	4 x 8 Speaker Processor	9-12 (Spkr 1-4) 25-28 (Spkr 5-8)	None (routing done on the Setup>Expansion screen)
	2 x 4 Speaker Processor / 10 Insert EQ	9-12 (Spkr 1-4)	None (routing done on the Setup>Expansion screen)
	10 Insert EQ	None (routing done on the setup>Expansion screen)	None (routing done on the Setup>Expansion screen)
ADAT	44.1/48 kHz	None, see ADAT table below	None, see ADAT table below
	88.2/96 kHz	None, see ADAT table below	None, see ADAT table below
U100	FOH/MON/DS3232 U-Net Routing	32	32
	Master/Slave Console Linking	None	None

ADAT Notes

The available built-in ADAT I/O and an ADAT card interact with sample rate changes. Because the channel-count halves for ADAT at the 88.2/96k rating, special consideration must be taken.

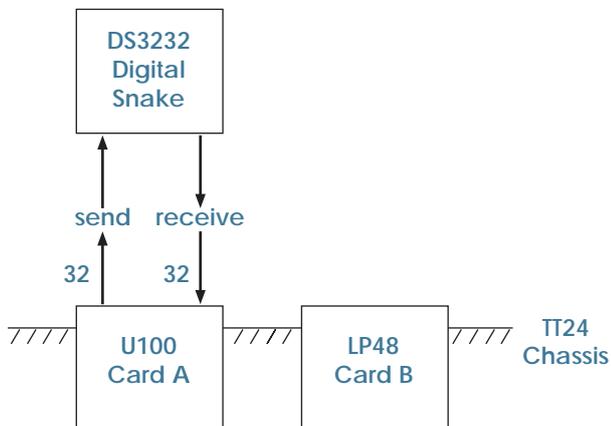
At 88.2/96k, half of the built in ADAT I/O (sources and destinations) will be unavailable and will be shown with the destination number in the output routing table greyed out. Similarly, the input and output routing will also indicate half of the sources are unavailable at 88.2/96k but still routable. Again, these will still be assignable and will resume functionality if you changes back to 44.1/48k. Adding an ADAT card will allow all the sources and destinations to be available at all sample rates. The card automatically routes the inputs and outputs for the additional I/O. It therefore is not directly routable.

There are three supported ADAT related configurations: no cards, 1 card in Slot A, and 1 card in Slot B. Two ADAT cards are not supported at this time. The various scenarios are summarized in this table:

ADAT	Mode	Built In ADAT I/O
Built-In	44.1/48 kHz	1-24
	88.2/96 kHz	1-12
Built-In and Card A	44.1/48 kHz	1-24
	88.2/96 kHz	1-24
Built-In and Card B	44.1/48 kHz	1-24
	88.2/96 kHz	1-24

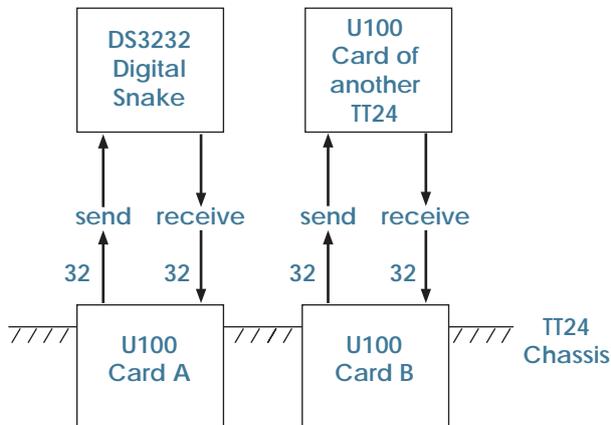
2.3 U-Net Routing Screens

If you are using a single TT24 and a DS3232, then the input and output routing screens allow you to choose the desired input and output routing. These two screens are generally all that you will need.



For a single TT24 console, the input and output screens allow you to configure all the TT24 console inputs and outputs, including the U100 card. For example, the U100 card in slot A can receive the inputs from your digital snake, and it can also send the LP48 card loudspeaker crossover outputs to the snake and off to your power amplifiers.

If you have two TT24 consoles and wish to link them together, you can use the U-Net routing screens. These are required to pass signals between the two U-Net cards. Using the U-Net routing allows the two TT24 consoles to share the digital snake and other inputs. For example, microphones connected to the snake can pass their signals to both consoles.



You must manually route up to 32 signals to and from each card using the Output Routing screen as described previously. You must also choose up to 32 signals to route-into and another 32 to route-out of the card slot signal paths themselves. This is done using the U-Net Routing screens. Because this can be labor intensive, many common setups are provided as routing presets. See Routing Presets on page 6.

Two U-Net Routing screens are available: one to route from Card Slot A to B and one to route from Card Slot B to A. Routing from Card Slot A to B involves choosing up to 32 U-Net signals to receive at the input of card A. You can then choose to route these signals directly back out of U-Net card B. This complements the Card A to B patching that can be done in the Output Routing screen. Routing from Card B to A is essentially the inverse. Otherwise, the two screens are the same. If one or the other U-Net cards is not installed, the routings will appear greyed out, but they can still be viewed and edited allowing configuration of the network in advance without a U100 card.

Each Routing screen is in two parts: receive and send. For example, the UNetAB Routing screen shows:

- U100 Slot A: Receive
- U100 Slot B: Send

In the Slot A Receive section, you can select the particular signals that you want to come into the Slot A Card either from a DS3232 or another TT24.

In the Slot B Send area, you can select the output signals available from the Card Slot B Sends.

Each U-Net Routing screen has a column of fields to select the 32 receive signals, and another one for the 32 send signals. An arrow at the top helps to illustrate the send/return relationship. Double clicking on this arrow is a shortcut to switch between the two U-Net routing screens. For each send or receive signal path, five drop down menus are necessary to define a U-Net signal:

- Product (DS3232 or TT24)
- Instance
- Signal Type
- Number
- Node

Double-click the arrow to see the next Routing screen

U100 SLOT A: RECEIVE					U100 SLOT B: SEND					
DS3232	1	Input	1	Pre Fader + x	1	TT24	FOH Master	Out Main	L	Pre DSP + x
DS3232	1	Input	2	Pre Fader + x	2	TT24	FOH Master	Out Main	R	Pre DSP + x
DS3232	1	Input	3	Pre Fader + x	3	TT24	FOH Master	Out Main	C/M	Pre DSP + x
DS3232	1	Input	4	Pre Fader + x	4					
DS3232	1	Input	5	Pre Fader + x	5					

Labels for U100 Slot A: RECEIVE:

- Product (points to DS3232)
- Instance (points to 1)
- Signal Type (points to Input)
- Number (points to 1)
- Node (points to Pre Fader)
- Increment Routing (+) (points to +)
- Clear Routing (X) (points to x)

Labels for U100 Slot B: SEND:

- Product (points to TT24)
- Instance (points to 1)
- Signal (points to FOH Master)
- Number (points to 1)
- Node (points to Out Main)
- Increment Routing (+) (points to +)
- Clear Routing (X) (points to x)
- Scroll up or down (points to the vertical arrow on the right)

U100 Slot A: RECEIVE U100 Slot B: SEND

Card signals 1 to 32

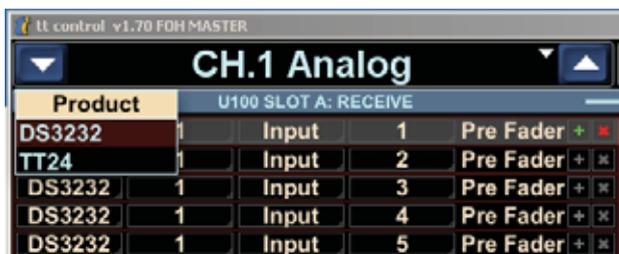
If you choose a U-Net Send signal that originates in the console (e.g. TT FOH Master Aux 1), the corresponding connection will also be automatically made in the Output Routing screen. The inverse is also true: if you route a signal to a Card Slot output signal path on the Output Routing screen, the corresponding signal will automatically be chosen in the appropriate U-Net Routing screen.

If you choose a U-Net Send signal that does not originate in the console (e.g. DS3232 Input 1), it will automatically be requested at the opposite card's U-Net Receive input. The corresponding Output Routing screen selection routing one card to the other will also be made.

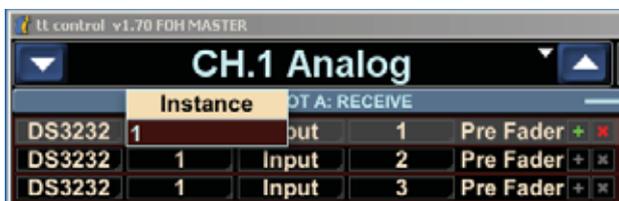
Selection Examples

The following example screens show the selections available for U100 Slot A: Receive. The same choices are available on Slot A or B, Send or Receive.

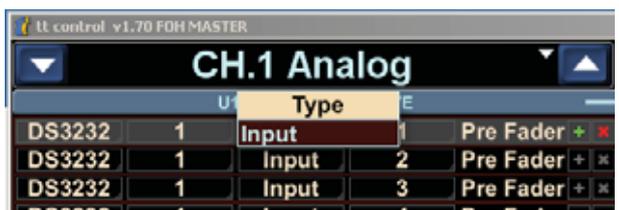
DS3232 on a Card Slot RECEIVE



Select DS3232.



Select 1.



Select Input.



Select Instance from 1 to 32.

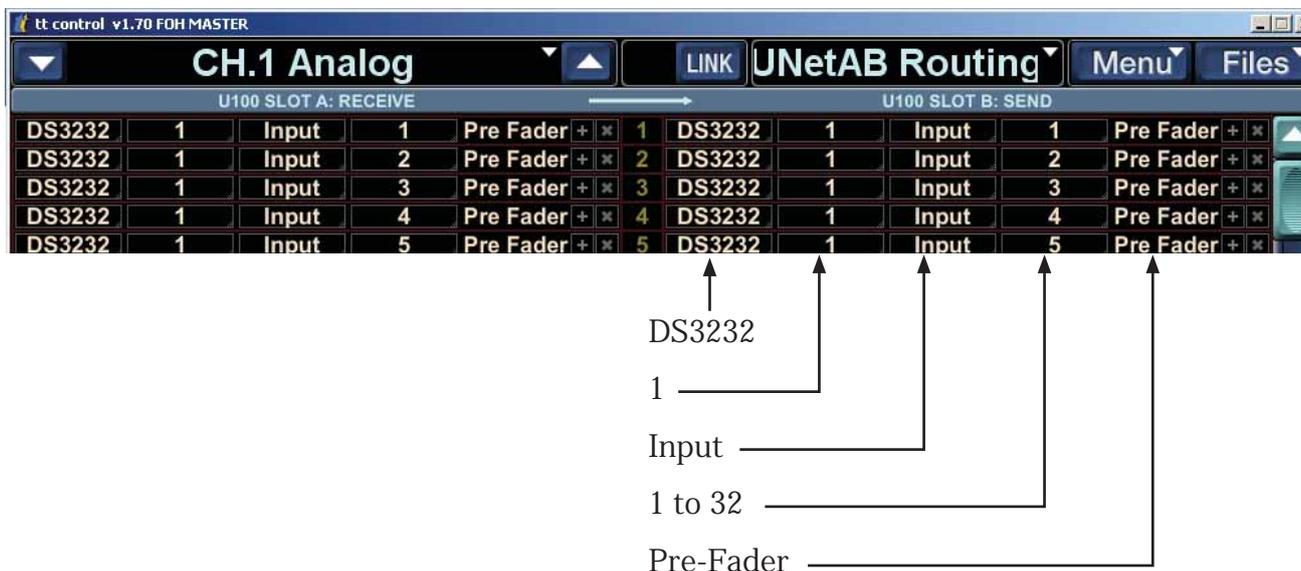
All are Pre Fader

So, what have we done?

We have set the U100 Slot A channel 1 so it will receive only the DS3232 signal that corresponds to our selections. The card is ready for the input of a DS3232 signal 1 on Card A receive channel 1. The card inputs can now be routed to input channels.

DS3232 on a Card Slot SEND

This example shows the U100 Slot B: SEND set up in exactly the same way as shown for the Slot A: RECEIVE on the previous page:



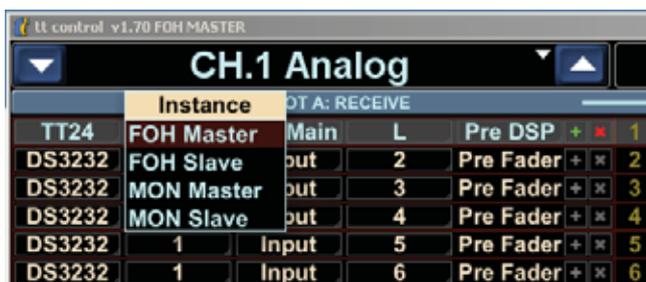
Here we have set the U100 Slot B SEND channel 1 so it will send only the DS3232 signal that corresponds to our selections. Card B Send is ready to send DS3232 signals.

You will notice that when you select the DS3232 as your send, the U100 Slot A: RECEIVE will also change to the same selection. This auto-filling-in of the receive side only happens if you are sending a signal that does not originate from the host console. The card is getting ready to send something it doesn't have, so it gets ready to find and receive it first.

If, in the above example, we now set up to send a TT24 FOH Master signal, the receive side will not change:



TT24 on a Card Slot RECEIVE



Select TT24, and the choices are:

- FOH Master
- FOH Slave
- MON Master
- MON Slave

Each of these selections has extra menus as shown in the next few pages of this guide:



Select FOH Master, and the choices are: Out Main, Out Aux, Out Group, Out Matrix, Out Monitor, Ch Bank 1, Ch Bank 2, Ch Line, Ch FX, Ch Card, In Mic/Line, In ADAT, In Line, In AES/SPDIF, In Tape, In Talkback, Card A and Card B.

Main: Choose L, R, C (or M)



The choices are: Pre DSP or Post Fader



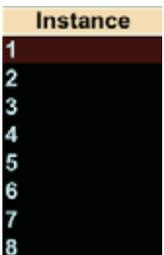
Aux: Choose 1 to 12



The choices are: Pre DSP or Post Fader



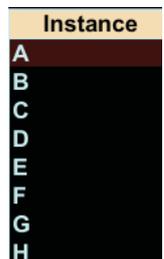
Group: Choose 1 to 8



The choices are: Pre DSP or Post Fader

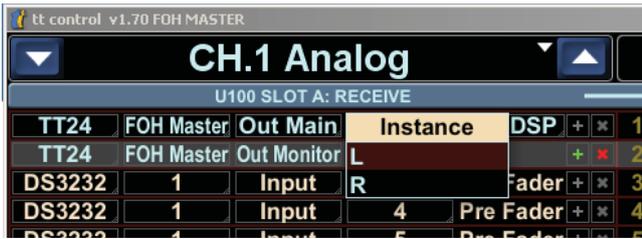


Matrix: Choose A to H



The choices are: Pre Delay or Post Delay





Monitor: Choices are L or R.

Instance
1
2
3
4
5
6
7
8
9
10
11
12
13
14
15
16
17
18
19
20
21
22
23
24

Ch Bank 1: Choices are 1 to 24 (Pre Fader)

Instance
25
26
27
28
29
30
31
32
33
34
35
36
37
38
39
40
41
42
43
44
45
46
47
48

Ch Bank 2: Choices are 25 to 48 (Pre Fader)

Instance
1
2
3
4
5
6
7
8

Ch Line:
Choices are 1 to 8
(Pre Fader)

Instance
1
2
3
4
5
6
7
8
9
10
11
12
13
14
15
16
17
18
19
20
21
22
23
24

In Mic/Line:
Choices are 1 to 24
(Input)

Instance
1 L
1 R
2 L
2 R
3 L
3 R
4 L
4 R

Ch FX:
Choices are 1L, 1R, 2L, 2R, 3L, 3R, 4L and 4R (Pre Fader)

Instance
1
2
3
4
5
6
7
8
9
10
11
12
13
14
15
16
17
18
19
20
21
22
23
24

ADAT:
Choices are 1 to 24
(Input)

Instance
1
2
3
4
5
6
7
8

In Line:
Choices are 1 to 8
(Input)

Instance
L
R

In AES/SPDIF:
Choices are L, R (Input)

Instance
In

Talkback: In (Input)

Instance
A L
A R
B L
B R

In Tape:
Choices are A L, A R, B L, or B R
(Input)

Instance
1
2
3
4
5
6
7
8
9
10
11
12
13
14
15
16
17
18
19
20
21
22
23
24
25
26
27
28
29
30
31
32

Card A:
Choices are 1 to 32

Instance
1
2
3
4
5
6
7
8
9
10
11
12
13
14
15
16
17
18
19
20
21
22
23
24
25
26
27
28
29
30
31
32

Card B:
Choices are 1 to 32

TT24 on a Card Slot SEND

This example shows the U100 Slot B: SEND set up to send the FOH Master main outputs. As these signals originate in the host console (set to FOH Master) setting the sends does not affect the receive settings (here the dashes show no assignment).

U100 SLOT A: RECEIVE					U100 SLOT B: SEND				
1	TT24	FOH Master	Out Main	L	Pre DSP	+	×		
2	TT24	FOH Master	Out Main	R	Pre DSP	+	×		
3	TT24	FOH Master	Out Main	C/M	Pre DSP	+	×		

Labels and arrows in the diagram point to:

- TT24 (Channel Name)
- FOH Master (Source)
- Ch Bank 11 to 32 (Output)
- 1-24 (Pan)
- Pre Fader (DSP)

U100 Slot B: SEND

If we change the Send signals to MON Master, these signals do not originate on the host console (configured as a FOH Master), so they must be found. In this case, the Receives have all been set automatically to the MON Master settings.

U100 SLOT A: RECEIVE					U100 SLOT B: SEND				
1	TT24	MON Master	Out Main	L	Pre DSP	+	×		
2	TT24	MON Master	Out Main	R	Pre DSP	+	×		
3	TT24	MON Master	Out Main	C/M	Pre DSP	+	×		
4									
5									
6									
7									
8									

Labels and arrows in the diagram point to:

- MON Master (Source for Receive)
- MON Master (Source for Send)

2.4 DS3232 Routing Macros

Because you can now use input routing to freely assign any hardware input to any channel, you can assign some snake signals to bank 1, some to bank 2, and still others to bank 3. This can be done per channel as previously described. Additionally, three macro buttons makes quick routing possible for all snake inputs.



On the Expansion screen with a DS3232 present, if ANALOG is pressed, the analog snake inputs are automatically assigned in the Input Routing screen to the analog inputs:

Analog



In the Input Routing screen, the analog inputs are automatically filled with the card slot U-Net card inputs.



In the Expansion screen, if DIGITAL is pressed, the digital snake inputs are automatically assigned in the Input Routing screen to the digital inputs as shown on the next page of this guide.

Digital

INPUT CHANNEL	SIGNAL SOURCE		
Analog 23	Hardware Input	Mic/Line	23
Analog 24	Hardware Input	Mic/Line	24
Digital 25	Card Slot	A - UNet	1
Digital 26	Card Slot	A - UNet	2
Digital 27	Card Slot	A - UNet	3
Digital 28	Card Slot	A - UNet	4
Digital 29	Card Slot	A - UNet	5
Digital 30	Card Slot	A - UNet	6
Digital 31	Card Slot	A - UNet	7
Digital 32	Card Slot	A - UNet	8
Digital 33	Card Slot	A - UNet	9
Digital 34	Card Slot	A - UNet	10
Digital 35	Card Slot	A - UNet	11
Digital 36	Card Slot	A - UNet	12
Digital 37	Card Slot	A - UNet	13
Digital 38	Card Slot	A - UNet	14
Digital 39	Card Slot	A - UNet	15
Digital 40	Card Slot	A - UNet	16
Digital 41	Card Slot	A - UNet	17
Digital 42	Card Slot	A - UNet	18
Digital 43	Card Slot	A - UNet	19
Digital 44	Card Slot	A - UNet	20
Digital 45	Card Slot	A - UNet	21
Digital 46	Card Slot	A - UNet	22
Digital 47	Card Slot	A - UNet	23
Digital 48	Card Slot	A - UNet	24
Line In 1	Hardware Input	Line	1
Line In 2	Hardware Input	Line	2
Line In 3	Hardware Input	Line	3

In the Input Routing screen, the digital inputs are automatically filled with the card slot U-Net card inputs.



In the Expansion screen, if NONE is selected, then no Snake inputs are automatically assigned to the U-Net cards:

None

INPUT CHANNEL	SIGNAL SOURCE		
Analog 1	Hardware Input	Mic/Line	1
Analog 2	Hardware Input	Mic/Line	2
Analog 3	Hardware Input	Mic/Line	3
Analog 4	Hardware Input	Mic/Line	4
Analog 5	Hardware Input	Mic/Line	5
Analog 6	Hardware Input	Mic/Line	6
Analog 7	Hardware Input	Mic/Line	7
Analog 8	Hardware Input	Mic/Line	8
Analog 9	Hardware Input	Mic/Line	9
Analog 10	Hardware Input	Mic/Line	10
Analog 11	Hardware Input	Mic/Line	11
Analog 12	Hardware Input	Mic/Line	12
Analog 13	Hardware Input	Mic/Line	13
Analog 14	Hardware Input	Mic/Line	14
Analog 15	Hardware Input	Mic/Line	15
Analog 16	Hardware Input	Mic/Line	16
Analog 17	Hardware Input	Mic/Line	17
Analog 18	Hardware Input	Mic/Line	18
Analog 19	Hardware Input	Mic/Line	19

INPUT CHANNEL	SIGNAL SOURCE		
Digital 36	Hardware Input	ADAT	
Digital 37	Hardware Input	ADAT	
Digital 38	Hardware Input	ADAT	
Digital 39	Hardware Input	ADAT	
Digital 40	Hardware Input	ADAT	
Digital 41	Hardware Input	ADAT	
Digital 42	Hardware Input	ADAT	
Digital 43	Hardware Input	ADAT	
Digital 44	Hardware Input	ADAT	
Digital 45	Hardware Input	ADAT	
Digital 46	Hardware Input	ADAT	
Digital 47	Hardware Input	ADAT	
Digital 48	Hardware Input	ADAT	
Line In 1	Hardware Input	Line	
Line In 2	Hardware Input	Line	
Line In 3	Hardware Input	Line	
Line In 4	Hardware Input	Line	
Line In 5	Hardware Input	Line	
Line In 6	Hardware Input	Line	

DS3232 Routing Macros output routing

HARDWARE OUTPUT		SIGNAL SOURCE		
Card A - UNet 3	Channel Output	Aux	3	Post Fader
Card A - UNet 4	Channel Output	Aux	4	Post Fader
Card A - UNet 5	Channel Output	Aux	5	Post Fader
Card A - UNet 6	Channel Output	Aux	6	Post Fader
Card A - UNet 7	Channel Output	Aux	7	Post Fader
Card A - UNet 8	Channel Output	Aux	8	Post Fader
Card A - UNet 9	Channel Output	Aux	9	Post Fader
Card A - UNet 10	Channel Output	Aux	10	Post Fader
Card A - UNet 11	Channel Output	Aux	11	Post Fader
Card A - UNet 12	Channel Output	Aux	12	Post Fader
Card A - UNet 13	Channel Output	Group	1	Post Fader
Card A - UNet 14	Channel Output	Group	2	Post Fader
Card A - UNet 15	Channel Output	Group	3	Post Fader
Card A - UNet 16	Channel Output	Group	4	Post Fader
Card A - UNet 17	Channel Output	Group	5	Post Fader
Card A - UNet 18	Channel Output	Group	6	Post Fader
Card A - UNet 19	Channel Output	Group	7	Post Fader
Card A - UNet 20	Channel Output	Group	8	Post Fader
Card A - UNet 21	Card Slot	B - LP48	9 - Spkr 1	
Card A - UNet 22	Card Slot	B - LP48	10 - Spkr 2	
Card A - UNet 23	Card Slot	B - LP48	11 - Spkr 3	
Card A - UNet 24	Card Slot	B - LP48	12 - Spkr 4	
Card A - UNet 25	Card Slot	B - LP48	25 - Spkr 5	
Card A - UNet 26	Card Slot	B - LP48	26 - Spkr 6	
Card A - UNet 27	Card Slot	B - LP48	27 - Spkr 7	
Card A - UNet 28	Card Slot	B - LP48	28 - Spkr 8	
Card A - UNet 29	Hardware Input	Talkback	In	
Card A - UNet 30	Channel Output	Main	L	Post Fader
Card A - UNet 31	Channel Output	Main	R	Post Fader
Card A - UNet 32	Channel Output	Main	M	Post Fader

In the Expansion screen, if DIGITAL or ANALOG are pressed, the outputs to the digital snake are automatically assigned in the Output Routing screen.

The outputs to the snake are:

Auxes 1 to 12, post fader.

Groups 1 to 8, or Matrix instead if assigned.

LP48 outputs (if LP48 card present) or Auxes again (if LP48 not present). Some or all of the LP48 outputs may be greyed-out and so not available, depending on the mode of the LP48 card.

Talkback.

Main outputs L, R, and M (or C).

Additional Controls

Snake assigned to channel

If a snake is assigned to an input channel, the following additional controls are shown for the channel overview:



Mic Pre Gain

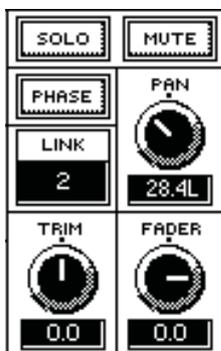
Pad

48V

Snake Digital Trim (note that this is a different parameter than the digital trim shown when a snake is not assigned – see gain compensation in section 3.1 or more information)



Snake not assigned to channel



If a single snake input is routed to more than one channel, then all channels fed by the snake input will see and have control of the same mic pre. This is because there is only a single mic pre, pad, and +48V for each input in the snake, and it therefore will affect all channels fed by that snake input. All snake digital trims are independent per channel.

2.5 Snapshot Filtering

Routing

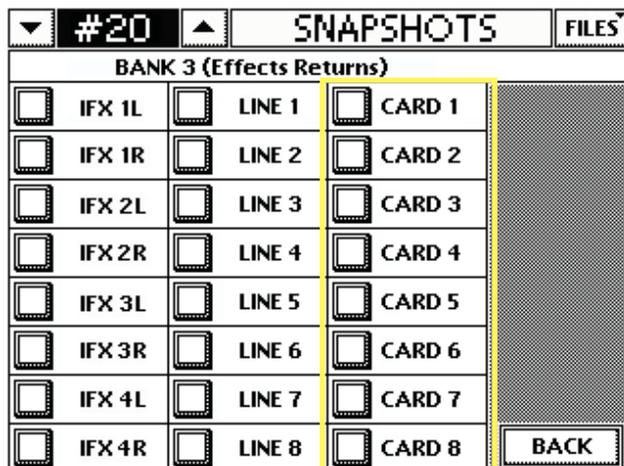
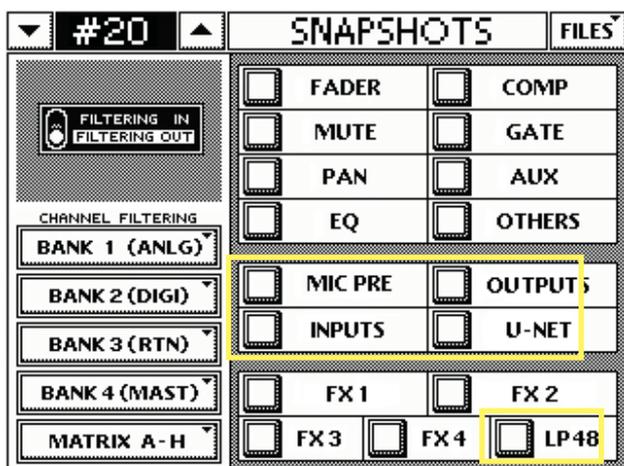
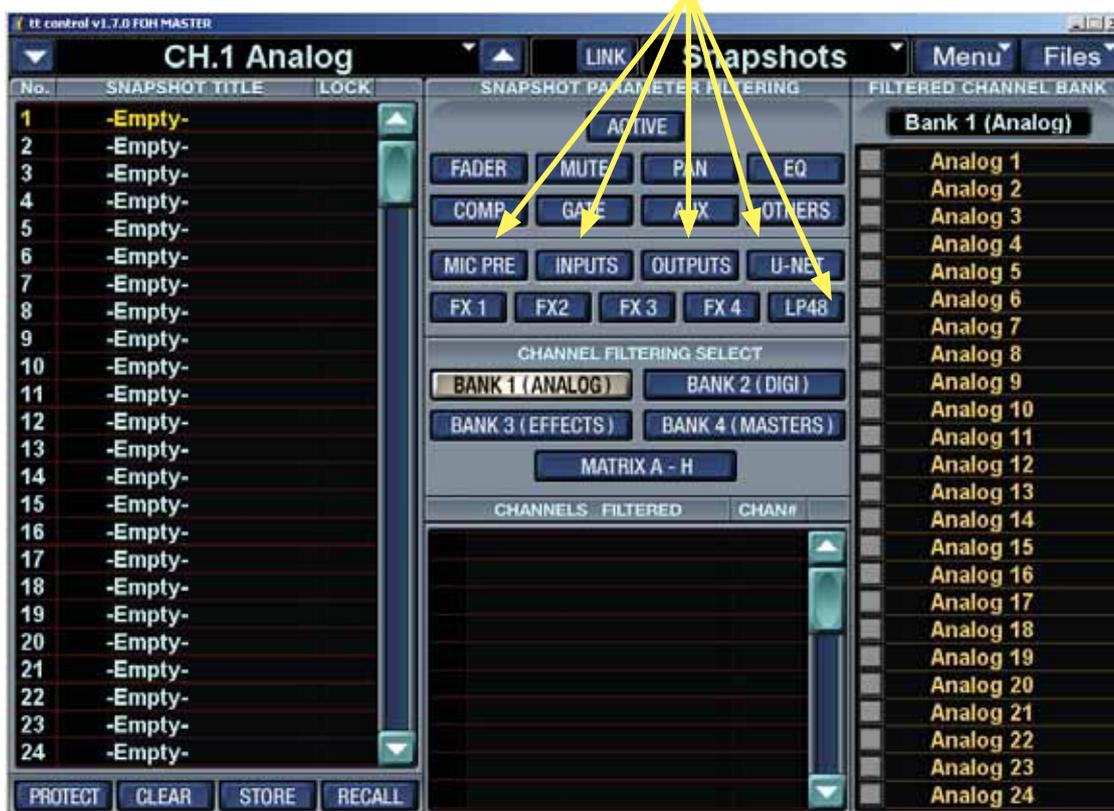
Routing is saved in a snapshot and is filtered with three new “routing” filter buttons:

- Inputs – Filters out all Input routing changes
- Outputs – Filters out all Output routing changes
- U-Net – Filters out all U-Net (A to B and B to A) routing changes.

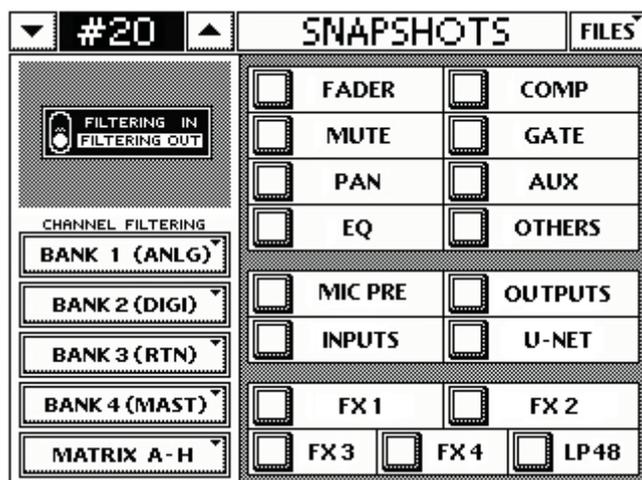
Unless filtered, loading a snapshot will change the entire console routing to that stored in the snapshot. When filtered, a snapshot will not recall any corresponding routing changes. Note that when doing U-Net routing, it usually makes sense to either enable or disable all three routing buttons at once since U-Net routing generally requires input and output routing changes as well. Two other (non-routing) filters are also added: LP48 and Mic Pre, and you can filter the card channels on Bank 3.

New Snapshot Filters

Snapshot Filtering Example



Snapshot Filtering – Mic Pre



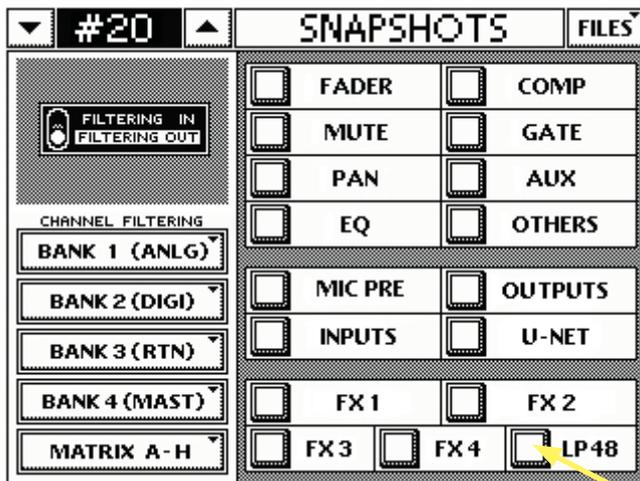
Snapshot filtering has a new filter added called MIC PRE. The following parameters will not be recalled when this filter is enabled:

- Snake Gain
- Snake Pad
- Snake 48v
- Digital Trim (previously filtered by Others and available even if a snake is not connected)
- Polarity Inverse (previously filtered by Others and available even if a snake is not connected)



Snapshot Filtering – LP48 card

Snapshot filtering has a new filter added called LP48. All the parameters of the LP48 card and the card setup found on Setup>Expansion will not be recalled when this filter is enabled:



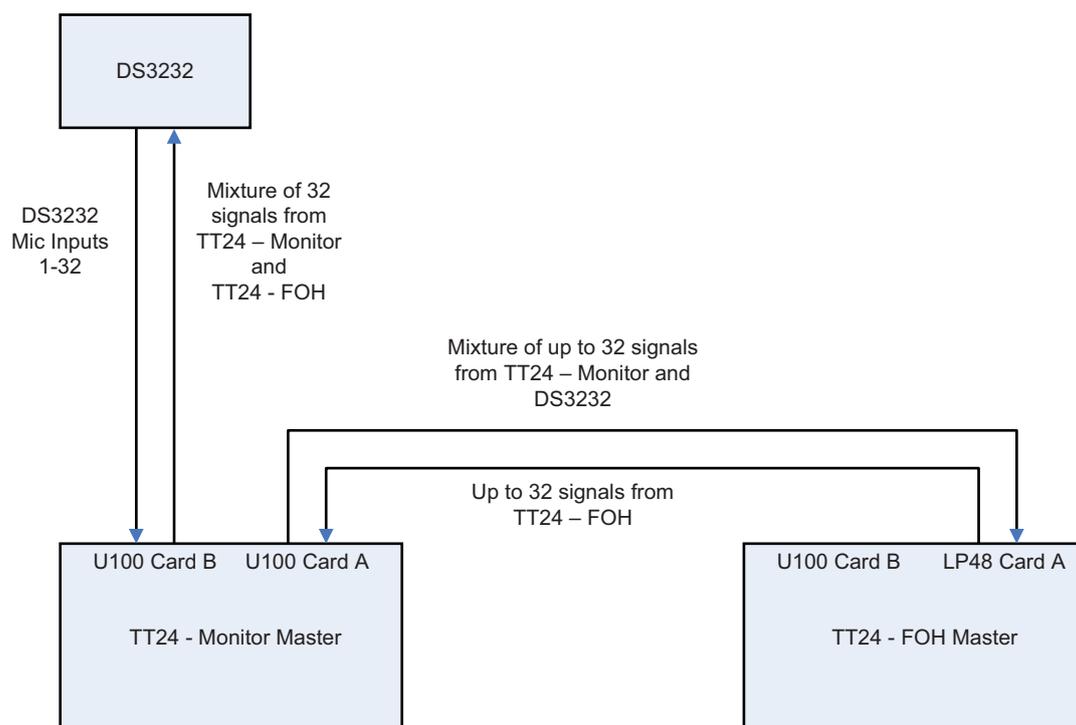
LP48 Snapshot Filter



3.0 Front of House / Monitor Console Linking

Front of House / Monitor Console Linking allows the support of two TT24s (one in front of house master position and the other in monitor master position) and one DS3232.

The Snake is connected to a U100 in the Monitor Master TT24 while a second U100 card in the monitor TT24 connects to a third U100 card in the FOH Master TT24. The second slot in the FOH TT24 can utilize any other card, such as the LP48. This allows both TT24s to use the inputs from the DS3232 and then both TT24s can feed their outputs to the DS3232 as well. This is accomplished via the routing pages. The terms FOH and Monitor are purely for convenience; the roles can be reversed and the snake can be connected to the FOH console if desired.



Linked TT24 FOH-Master and Monitor-Master with DS3232

This configuration could even be used with a conventional analog snake connected to one of the consoles. In addition, linked consoles at FOH and MON and two snake configurations can also be supported.

When the TT24s are linked in this fashion, the two will send signal between each other just as they send and receive signals to and from a DS3232. This is also done from the U-Net routing screens. Since the Monitor console is directly connected to the snake, the FOH console must route signal to the monitor console. The monitor engineer can then route that signal on to the snake as one of the 32 available channels using the U-Net and Output routing screens.

When linking consoles, either for the new FOH-MON linking or the existing FOH-FOH linking, you must configure the following new controls from the Setup>Linking screen:



- Console Link Mode – This configures how the console behaves with other linked consoles. Only one can be selected at a time on any console.
 - FOH Master (Default)
 - MON Master
 - FOH Slave
 - MON Slave
- Linked Clock Mode – This configures which unit will be used as the clock master when linked. Only one TT24 on a linked chain should be configured as Linked Clock Master. All other TT24s should be configured as clock slave. The clock master must still configure internal or external clock on the setup>digital page. This setting effectively sets the TT24 to the highest clock priority which will cause it to be chosen as the U-Net clock master by the U-Net clock negotiation. If more than one console is configured as master (or none are configured as Master), the system will automatically choose one using standard U-Net clock negotiation; this may lead to clocking issues or unexpected clocking behavior. By default this is set to Slave.
- Mic Pre Mode– This configures whether the console is master or slave for Mic Pre Gain Compensation. This only has an effect if consoles are connected to a DS3232. If only one console is connected to a snake, it should be set to Mic Pre Master. See the rest of this section for details on Gain Compensation. By default it is set to Master.
- Mic Pre Trim Compensate On/Off – This configures whether the console is compensating for gain changes. Any combination of Mic Pre Masters or Slaves can be configured to compensate as desired. See the rest of this section for details on Gain Compensation. By default it is set to Off.

Note that if more than one console is configured for the same Console Link Mode on a network, then a U-Net Network error will occur.

Also, note that only one Clock master should be present, but many Clock slaves can be connected.

Note also that the Console Master Slave setting is no longer present on the setup operation screen as this is now configured on the new Setup>Linking screen described above.

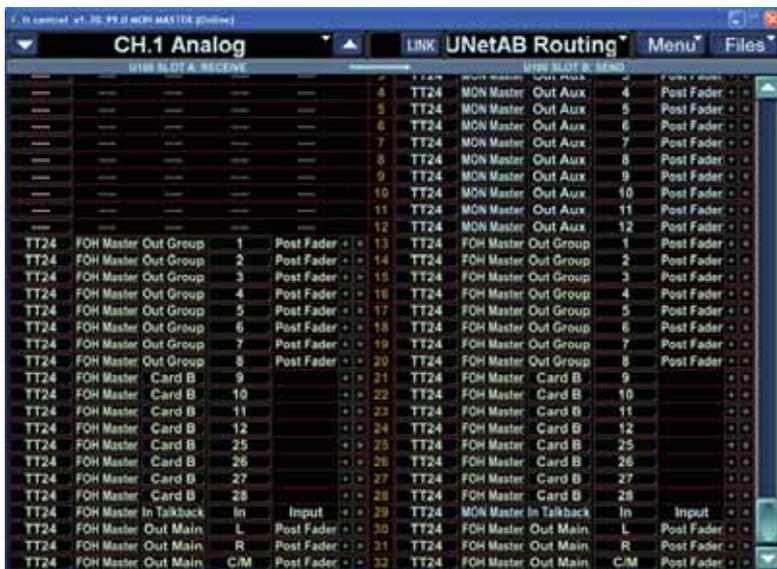
Setup – Operation Screen



Note that the input DSP option is still available for backwards compatibility and convenience. The equivalent can now be achieved with much more flexibility by leaving the input DSP on the analog bank and changing the input routing as desired.

Monitor Master U-Net Routing

For this setup, the U-Net Routing Screens for the Monitor Master TT24 console are setup as follows, using the factory preset "MON Snake LP48."



U-Net AB Routing

Card A is set to receive the FOH Group outputs, the FOH Card B outputs (such as from an LP48), FOH Talkback, and the FOH main outputs.

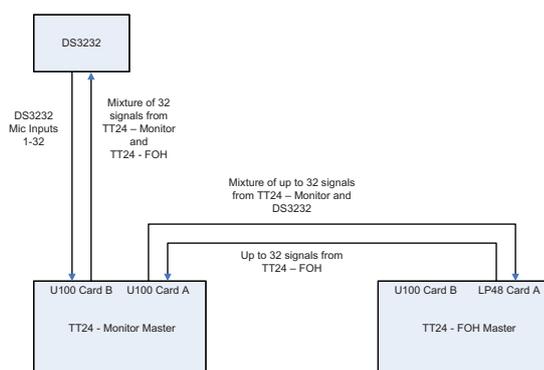
Card B is set to send the Mon Master Auxes, FOH Groups, Card B, Mon Talkback, FOH mains.



U-Net BA Routing

Card A is set to send along the DS3232 signals (to the FOH Master).

Card B is set to receive the DS3232 signals from the DS3232.



Linked TT24 FOH-Master and Monitor-Master with DS3232

FOH Master U-Net Routing

For this setup, the U-Net Routing Screens for the FOH Master TT24 console are setup as follows, using the factory preset "MON Snake LP48."



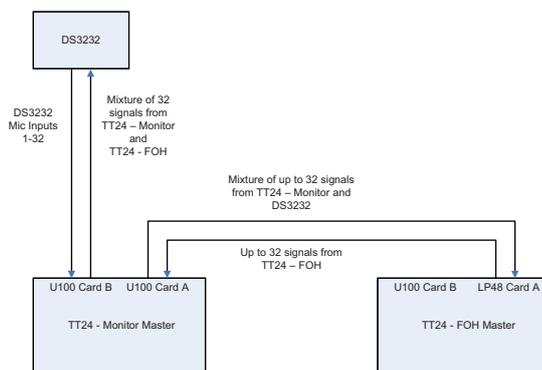
U-Net AB Routing

Card A is set to receive the DS3232 signals (via the TT24 Monitor Master).



U-Net BA Routing

Card A is set to send the FOH Master Group outputs (post fader) the Card B outputs (such as from an LP48), Talkback, and the main outputs.



Linked TT24 FOH-Master and Monitor-Master with DS3232

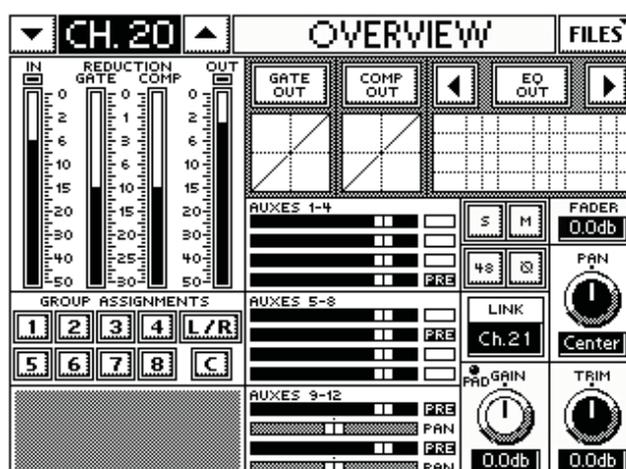
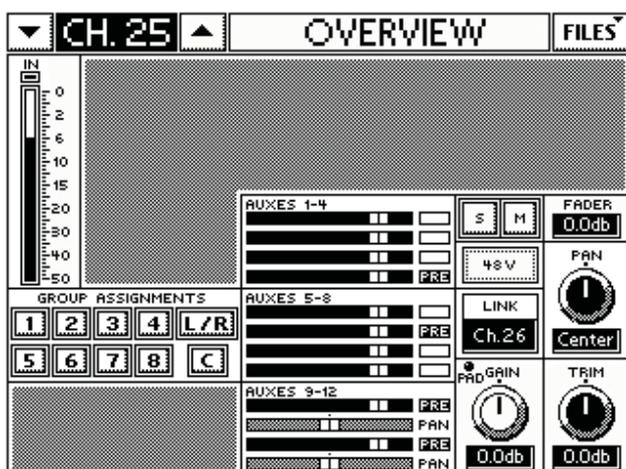
3.1 Gain Compensation

When FOH-MON linking, since both consoles will be using inputs from the same mic pre, it may be desirable to have gain changes made to the mic pre not affect the other console. This is done using the digital trim on one console to compensate for changes made to the gain from the other. For this to work, at least one console must be designated to actually control the mic pre by enabling Mic Pre Master on the Setup>Linking screen as described above.

Note: In the details below, the term “slave” and “master” refer to if the consoles are set to be a mic-pre slave or mic-pre master, not a FOH or Mon slave or a FOH or Mon master.

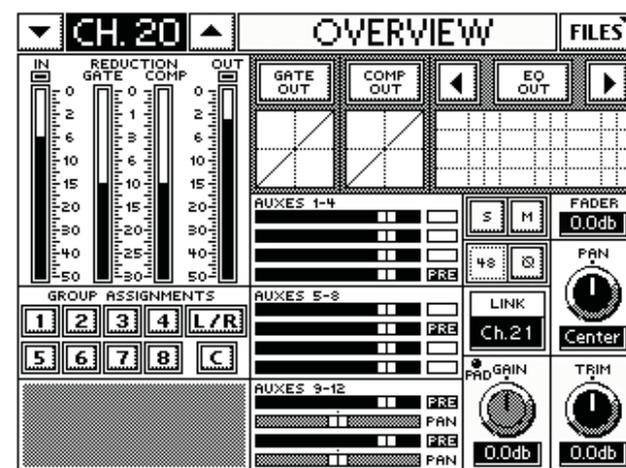
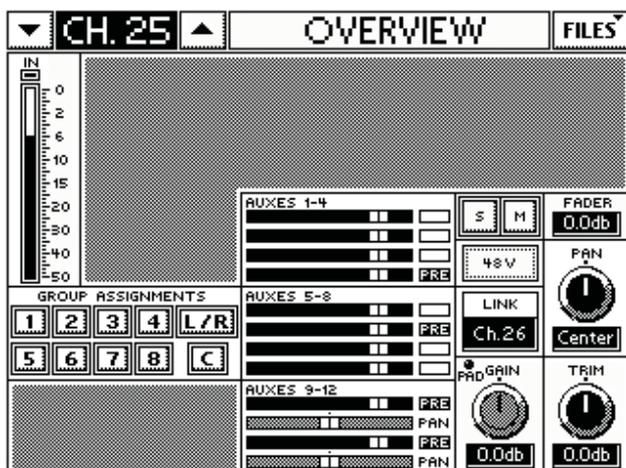
Mic Pre Master

With a Mic Pre Master console and a Mic Pre Slave console connected to a DS3232, the mic-pre-master console has control of the Mic Pre gain, 48V, and Pad. The Trim V-Pot control shows Mic Pre gain on the channel V-Pots



Mic Pre Slave

The mic-pre-slave console can display all true values of Mic Pre Gain, 48v, and Pad, but the Digital Trim is the only control that is adjustable. The Trim V-Pot control shows digital trim on the channel V-Pots.



When the mic-pre-master turns up or down the gain or adjusts the pad, the mic-pre-slave console with Mic Pre Gain Compensation enabled, automatically compensates by adjusting its digital trim. For instance, if the mic-pre-master turns up the mic pre gain by 5 dB, the mic-pre-slave's digital trim gets turned down 5 dB. This works fine once things are up and running, but it would be weird for initial setup; you may end up with some very strange settings on the mic-pre-slave trims during initial setup as all of the mic gains are being turned up. For this reason, the mic-pre-slave console can stop compensation during initial set up by disabling the "Trim Compensate" button. When compensating, the console Trim V-Pot control flashes when selected, and pressing the console Ctrl + TRIM toggles this selection.

Also, + or - 15 dB of trim might not be enough to compensate in all situations especially because the pad requires 23 dB of compensation, therefore, the digital trim which normally only allows +-15dB of trim control without a snake attached, changes to +-60 dB when a snake is attached. Note that if the mic-pre-slave needs to compensate with more than +-15 dB of digital trim, something is probably wrong in the gain structure somewhere.

Another complex issue arises when snapshots are used. Both the mic-pre-master and mic-pre-slave console need to be able to use snapshots to recall gain settings.

The mic-pre-master console actually controls the Mic Pre Controls, so it can store and recall these settings. Doing so will send the information to the mic-pre-slave console which will compensate for these snapshot induced changes, just as it does when you manually adjusts these parameters.

When the mic-pre-slave engineer recalls a snapshot, their goal is to have an absolute gain stored in the snapshot recalled and applied to the input. But, since the mic-pre-slave console cannot directly control the snake mic pres, it must calculate the compensated gain from the snapshot (stored mic pre gain + pad + stored digital trim), and then adjust the digital trim to achieve the same level of compensation with the current mic pre gain setting set by the mic-pre-master.

For example, let us assume the mic-pre-master and mic-pre-slave are configured as shown in the columns below:

	Master	Slave Before	Slave Snapshot	Slave Recall
Gain	40 dB	Na	45 dB	Na
Pad	Out	Na	Out	Na
48v	Out	Na	Out	Na
Trim	0 dB	-10 dB	0 dB	5 dB
Compensated Gain	40 dB	30 dB	45 dB	45 dB

The snake gain is set to 40 dB by the mic-pre-master console. The mic-pre-slave console engineer has adjusted the trim to -10 dB giving a compensated gain to 30 dB into the corresponding slave console channel. If the master were to increase the Gain in the snake, the mic-pre-slave would decrease the digital trim by the same amount. The mic-pre-slave snapshot (which could have been created when the console was a mic-pre-master or mic-pre-slave – see below) has 45 dB of compensated Gain through the mic pre and digital trim. Since the current state of the snake mic pre is 40 dB, to achieve 45 dB of compensated gain, the mic-pre-slave console must adjust the digital trim to 5 dB as shown in the Slave Recall column.

This allows snapshots to be shared between mic-pre-master and mic-pre-slave console and to allow the console to operate in either configuration while keeping their snapshots working properly. This is because when storing snapshots on the mic-pre-slave, the master gain, pad, 48v are stored in the snapshot along with the snake digital trim, so the snapshot is no different then that on a mic-pre-master console.

4.0 Appendix Ramblings: Front of House and Monitor Mixer Utilizing a Snake

In larger venues, a separate front of house and monitor console are used. In an all-analog world, the inputs from stage are sent to both the front of house and monitor console by using a splitter snake. Each console then receives the signal and each engineer has an independent mic pre for the same input signal. This is easy to use but the engineers must deal with extremely long and heavy cable runs making set up/tear down and audio quality an issue.

In a digital world, a digital snake is often used to convert the signal to digital and send it to each console over a light weight digital connection. This also improves audio quality since there is no signal loss over the long distance.

Shared I/O from a Single Analog Snake

A splitter snake can be rendered unnecessary if the two digital consoles used at FOH and Monitor can pass signal between them digitally. If this is the case, only one of the two consoles needs to be connected to the analog snake and it can pass these signals (or any other for that matter) on to the other remote console. Effectively, the console and digital connection are acting as a splitter snake. The remote console can also use the digital connection to send outputs back to the console connected to the snake, which can then route the analog outputs back to the snake. Thus either console can receive inputs from and send outputs to the analog snake.

Shared Mic Pres from a Single Digital Snake

The digital snake solution is not always perfect, especially in a scenario where FOH and Monitor desks are connected to the same snake. Only a single mic pre is used and it is instead positioned in the flow before the signal is sent to both consoles. Thus the engineers don't have independent control of the mic pre, as changing the gain from one console will affect the other. This can create havoc. For instance, if the monitor engineer were to reduce the gain of the lead vocalists mic pre because they are getting feedback with the monitors, this would lower the level of the vocalist at FOH making it hard for the audience to hear them.

To remove the symptoms of this problem, adjusting the gain from one console should not affect the other consoles gain structure, and vice versa. Engineers have come to expect this "gain compensation" when working with digital desks sharing a digital snake. This is often done by designating one console a master and the other a slave.

Shared Outputs from a Single Digital Snake

On the opposite side of things, the two consoles must be able to both feed signals to the single digital snake with a fixed number of outputs. Each engineer must be able to send signals to the snake for patching to amps/active speakers.

Snapshot Recall of Mic Pre Parameters

Because the mic pre parameters can be stored and recalled via snapshots, this can cause some problems with sharing the mic pre. Ultimately, the users of both consoles want to be able to store and recall snapshots with gain changes and want changes made from one to not affect the other.

Features of the 1.7 upgrade

- Two consoles can share the input signals from a single analog or digital snake
- Gain Compensation allows two consoles sharing the same mic pre from a digital snake to have apparently independent control of their gain structure by using a master/slave structure.
- When the master console recalls a snapshot with gain changes, the slave console can compensate.
- When the slave console recalls a snapshot with gain changes, the master console is unaffected.

Recording at Front of House

In today's live sound environment, the need to record shows off the front of house desk is becoming increasingly important. This is especially true at churches and other fixed installs which may need to deliver a variety of different performances to a variety of different recording and broadcast mediums.

Stereo Recording

The most simple delivery medium is a stereo recording to CD. This is commonly done straight off the LR mix buss with the exact same signal and level. It would be more beneficial if this stereo mix could be at a slightly different level allowing you to hit the CD converters as loud as possible to achieve the best possible SNR. In an advanced scenario, the engineer may wish to use a different combination of levels from each channel for recording than is used at FOH. Finally, in addition to creating a different mix of levels for recording, the engineer would ideally like to have a different set of DSP available for the recording mix essentially having a completely different mix for recording than for the live output.

Multi Track Recording

If an engineer doesn't have the time to make a perfect recording mix, or wants to endlessly perfect the recording mix after the fact, then they may choose to record individual inputs or groups of inputs onto a multi track device. Sometimes they may choose to record the inputs without any processing, completely independent of the main mix they are doing for the live show. Other times they may wish to use the process signals for the recording. Either way, the recording process should require very little attention during the show. When the show is complete and they are comfortably in their mixing studio, they can take the time to create the perfect mix.

Features of the 1.7 upgrade

- A console can route any output signal to any physical output or outputs
- A console can route any physical input signal to any input channel or channels
- A console can route any input channel's pre-DSP or post-DSP signal to any physical output or outputs
- A console can route any output channel's pre-DSP or post-DSP signal to any physical output or outputs

Front of house and monitor mixing from the same mixer

Many users don't require a separate monitor console and front of house console. This means that they must mix both from the same console.

Separate Channels for FOH and Monitor

Users with enough input channels may want to use a separate channel for FOH and Monitor mixing. This essentially gives them the benefit of two mixers in one.

Features of the 1.7 upgrade

- A console can route any physical input signal to any input channel or channels

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