



# TECHNICAL SALES DATA ISSUE 14

Calrec Audio Ltd Nutclough Mill Hebden Bridge West Yorkshire HX7 8EZ England UK

Tel: + 44 (0) 1422 842159 Fax: + 44 (0) 1422 845244 Email: enquiries@calrec.com Website: www.calrec.com



.....



# CONTENTS

OVERVI	EW	5
	INTRODUCTION	-
	PRINCIPAL FEATURES	7
	IMPORTANT CONCEPTS	8
	PATHS AND PORTS	9
	SIGNAL PATHS	
	INPUTS AND OUTPUTS	
	TYPICAL DIGITAL SYSTEM DIAGRAM	
	OPTIONAL I/O EXPANSION VIA WIDE AREA INTERFACES	13
FRAME	OPTIONS & DIMENSIONS	
	CONTROL SURFACE FRAME SIZES	
	END PROFILE	18
FADER /	AREA	
	"CHANNEL" FADERS	
	"CHANNEL" CONTROL	
	ASSIGNABLE FADER	
	VCA GROUPING	23
ASSIGN	PANELS	
	I/O MATRIX	
	INPUT/OUTPUT CONTROLS	
	FUNCTIONS CONTROLS	
	ROUTING PANEL	
	TALKBACK	
	EQ & FILTERS PANEL	
	DYNAMICS PANEL	
	AUXILIARY CONTROL PANEL	
	AUXILIARY OUTPUT PANELS	
	MAIN OUTPUTS PANEL	
	BROADCAST FACILITIES	
	MONITORING, METER SELECT AND LS CONTROL	
	MOTORISED JOYSTICK PANEL (OPTIONAL)	
	MEMORY CONTROLS	
	INPUT DELAY PANEL (OPTIONAL)	
	OSCILLATOR CONTROLS	
	METERING OPTIONS	
	OPTIONAL THIRD PARTY METERING	

.....





PC SCREEN USAGE & LAYOUT	
ERROR MESSAGES (AWACS) INPUT/OUTPUT PORTS SCREENS	
CHANNEL FUNCTIONS SCREENS	
SYNCHRONISATION SCREEN	
MONITOR I/P & TB SCREEN - TALKBACK INPUT PORTS VIEW	
GENERAL PURPOSE OUTPUTS SCREEN	
GENERAL PURPOSE INPUTS SCREEN	
AUTOMATIC CROSS-FADING	
TECHNICAL INFORMATION	65
TYPICAL RACK LAYOUT	66
RACK SPECIFICATIONS	
POWER SUPPLIES	68
PC INFORMATION	
INTERFACE CONNECTOR PANELS	
MAXIMUM CABLE LENGTHS	
ENVIRONMENTAL CONSIDERATIONS	
SYSTEM SPECIFICATION	72
NOTES	74

### © 2004 Calrec Audio Ltd. All Rights Reserved.

No part of this manual may be reproduced or transmitted in any form or by any means, electronic or mechanical, including photocopying & scanning, for any purpose, without the prior written consent of Calrec Audio Ltd.

Whilst the Company ensures that all details in this document are correct at the time of publication, we reserve the right to alter specifications & equipment without notice. Any changes we make will be reflected in subsequent issues of this document. The latest version will be available upon request.

This publication is for International usage.



. . . . . . . . . . . . . . . .

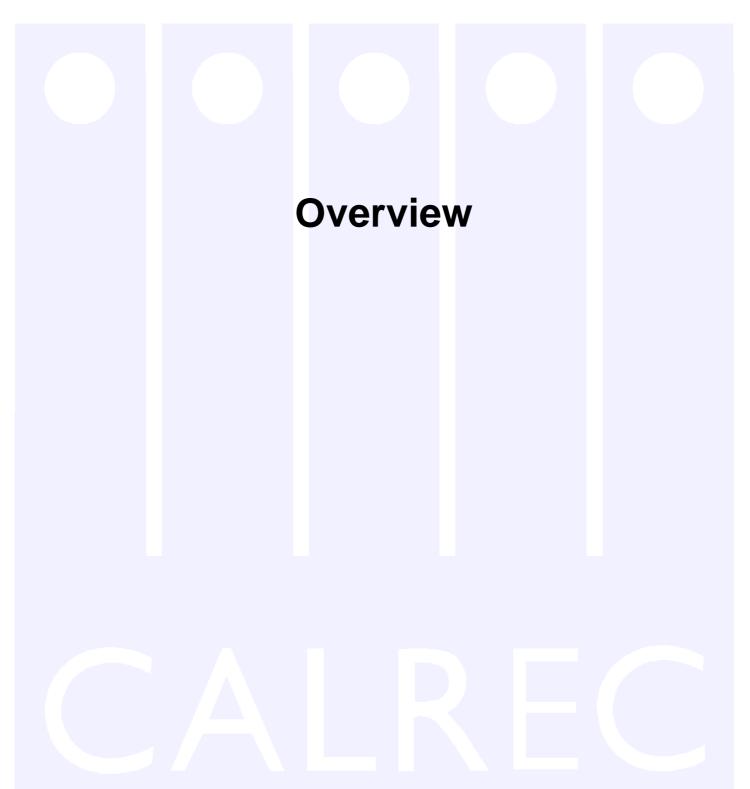






. . . . . . . . . . . . . . . . . .









## INTRODUCTION

The Alpha 100 is a large format digital console designed for the most critical broadcast production and on-air applications. It is a no-compromise design that provides comprehensive features and functionality with sophisticated failure protection systems. The Alpha 100 represents a milestone in digital audio mixing console systems as it offers the reliability associated with analogue technology but with the flexibility of an all-digital system.

The Alpha 100 is the result of over 30 years experience in broadcast console design and is the third generation of Calrec consoles to feature a digital control surface and computer-aided memory system. The introduction of digitally controlled assignable systems in 1980 has allowed for their ergonomics to be continuously refined by user input and the Alpha 100 reflects this in its user interface. In contrast to many other designs, the flexibility offered by digital control has been harnessed to provide greater functionality and ease of use.

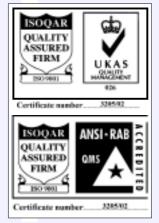
Calrec has a world-wide customer base which includes many of the world's most prestigious broadcasters. By consistently focusing upon purely broadcast products, Calrec offers consoles with the most comprehensive combination of performance and features available. The high level of reliability of all Calrec products, many of which are still in daily use after 20 years service, reflects a clear awareness of the critical nature of the operating environment.

This understanding of the real issues of broadcast operations is one of the many reasons why operators and management alike prefer Calrec. The Alpha 100 is designed to ensure this level of confidence will continue in the digital era.

### ISO 9001 AND RAB REGISTERED

Calrec Audio Ltd has been issued the ISO9001: 2000 standard by the Governing Board of ISOQAR.

The award, for both UKAS and RAB registration, is the most comprehensive of the ISO9000 international standards. Granted in recognition of excellence across design, development, manufacture and after-sales support, the certification follows a rigorous and thorough review of Calrec's internal and external communication and business procedures.









## PRINCIPAL FEATURES

#### Format

Up to 96 faders, with A & B layers of control, plus 4 dedicated main output faders. 226 equivalent channels: Up to 96 stereo or mono channels plus 34 mono channels. Comprehensive surround panning and monitoring with optional motorised joystick. Input Delay and control panel option.

Optional I/O expansion via a wide area interface such as MADI or Hydra, Calrec's sophisticated audio networking system.

#### **Channel / Group Facilities**

All channels have 4-band EQ, 2-band Filters, Compressor/Limiter and Expander/Gate. All groups have Compressor and Expander/Gate.

Up to 20 auxiliary outputs which can be 20 mono or 10 stereo.

There is a pool of assignable inserts and a pool of direct outputs for channels and groups. Pre configured inserts are assignable to any channel or group.

Direct outputs can be from pre EQ, pre fader, or post fader.

Every direct output can be a mix minus feed.

Automatic cross-fading facility, with user-definable fade out and in times

All faders are motorised and touch-sensitive.

A centrally assigned fader allows control of any fader from the optimum listening position.

### Routing

8 stereo or mono audio groups.

Additional VCA style grouping system.

Up to 48 outputs for multi-track or general purpose feeds.

Tracks can be fed from pre EQ, pre fader, post fader or mix minus.

4 main stereo or 5.1 surround outputs with Compressors.

Simultaneous LCRS, stereo and mono outputs available from each 5.1 main output.

Every channel can route to every bus, at the same time, without restrictions.

Direct input available to group, mains, aux and mix-minus busses.

#### System

On board Flash ROM memory system allows 99 full console snaphot or partial memories. PC backup allows an unlimited number of memories.

Console operates independantly of PC.

Sophisticated GPIO facilities.

Independent DSP operation ensures audio continuity even during PC or control reset.

Console & racks boot from power on in less than 20 seconds.

Full control system reset in less than 15 seconds.

Last settings fully restored on power-up or reset.

Automatic change over to hot spares for power supplies, control cards & DSP cards. Hot plugging of every card and module.

Hot plugged cards initialise upon insertion.





#### **IMPORTANT CONCEPTS**

### Layering

Each fader can control two independent audio signal paths, named A and B. These signal paths can be either channels or groups, although for easy reference, the faders are simply known as "Channel Faders". B signal paths are fully equipped with all the same facilities as an A path.

The faders are motorised, so when switching between A and B, the fader will move to the correct position.

Less important signals can be placed on the B layer. Even then, only one button press is required to access them again. Using the ALL A and ALL B buttons is like moving to a different section of a single layer design.

This arrangement allows more channels to be fitted into the space available in the frame. Channels towards the ends of the control surface can be accessed more quickly than on a conventional, single layer design.

#### **Assignable Control**

MR SL ST SS	A	O GP A
MR SL ST SS	В	GP B O

Each fader has an "Assign" button (sometimes called the "Show Me" button) for each audio path. The Assign buttons are labelled A & B for channel or group paths, and M1, M2, M3 or M4 for the main output paths on the main faders. Pressing the Assign button causes the central control panels (the "Assign panels") to display and control the settings for that fader's channel, group or main path.

In this way a large number of controls can be accessed, for each audio path, from the central listening position. As there is less need to move around a large control surface, controls can be accessed more efficiently.

A number of controls and displays are also provided on a per fader basis, to allow important information to be even more easily available.

In addition to the above, the "channel" faders are assignable, in that the operator can choose which faders to use for the mono channels, which for the stereo channels, and which for the groups.

For large consoles, some of the Assign panels may be duplicated to aid user operation.





### PATHS AND PORTS

On an analogue desk, the channel inputs are physical connections to the channel module or card. They are fixed. Channel 1's input is always channel 1's input (even though it may be possible to control channel 1 from a different fader). Every channel will probably have both a mic and a line input, even though most will only use one of them at any one time.

In a digital desk, there are two basic types of input: mic/line and digital. However, it is not necessary to provide both types for every channel, as only one input will be used at any one time. To provide both types for each channel would increase the cost, size and power consumption of the desk unnecessarily.

Instead, a "pool" of each type is available, plus an internal matrix to allow any of them to be connected to any channel. This provides more flexibility than is possible with analogue designs. The matrix can be thought of as an electronic patch-bay with the advantage that any connections made will be stored with the console's memories. A similar matrix and "pool" is provided for the outputs which can also be stored with the memories.

Each channel can select from two inputs (1 and 2), which can be any combination of mic/line and digital. Both inputs can be set up independently, using separate input controls ( input gain, phase reverse, phantom power, etc). The switching between the two inputs takes place after these controls.

The basic terminology is that channels, groups and mains are referred to as "paths" within the digital processing system, and the inputs and outputs are referred to as "ports" through which the audio signals have to pass. Ports are connected to paths via the Matrix.

All ports are optional, including those for the monitoring. The system can be supplied with any combination of mic/line and digital ports. Calrec digital consoles are available in a number of configurations known as Audio Packs, which are a suggested complement of ports. The Audio Pack which most closely matches the requirements of the installation can be chosen, and the port quantities can be fine tuned appropriately.





# SIGNAL PATHS

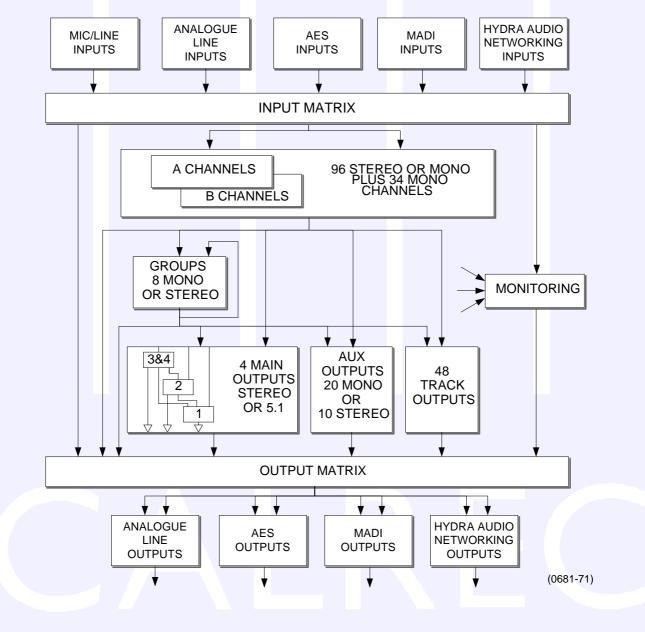
The Alpha 100 system can have 226 equivalent channels: up to 96 stereo or mono, plus 34 mono channels.

The 8 groups can each be designated as stereo or mono. In addition, as many VCA style groups as required can be created.

The 4 main outputs can each be designated as stereo or 5.1 surround. If they are 5.1 surround, a mono rear is derived at the output to allow them to be used as LCRS mains. stereo and mono downmixes of the 5.1 are also produced.

If a channel is panned to both a stereo and 5.1 bus simultaneously, the pan law to each will be correct, as though the other bus did not exist; even though the same control is used to achieve the pan.

The 20 mono auxiliary outputs can be paired up to give up to 10 stereo auxiliary outputs.







## **INPUTS AND OUTPUTS**

There are two types of ANALOGUE INPUT CARD:

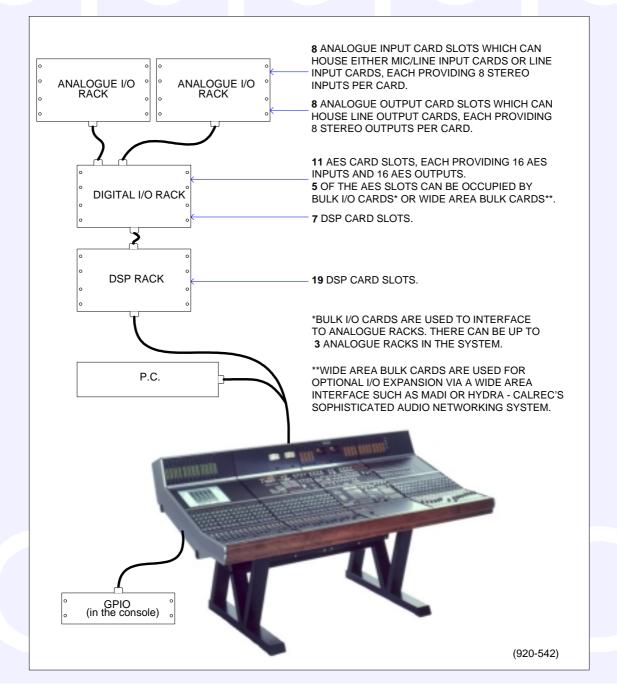
- Mic/Line input card 8 stereo or 16 mono inputs per card.
- Line input card 8 stereo or 16 mono inputs per card.

There is one type of ANALOGUE OUTPUT CARD:

■ Line output card - 8 stereo or 16 mono line outputs per card.

There is a DIGITAL (AES3) INPUT/OUTPUT CARD:

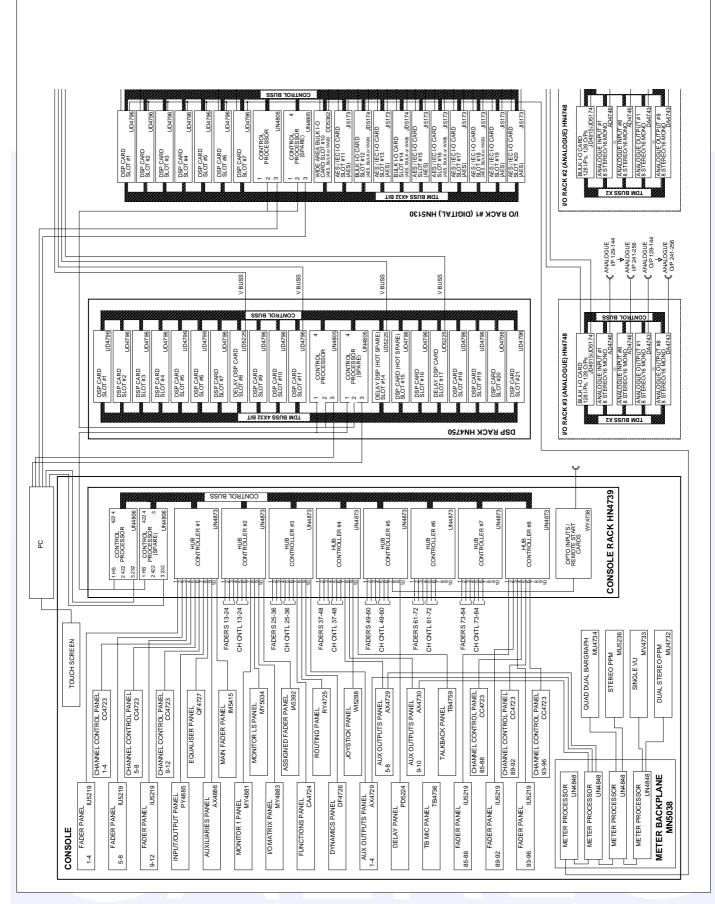
Digital (AES3) input/output card - 16 AES inputs and 16 AES outputs per card. All inputs have switchable sample rate conversion.







#### **TYPICAL DIGITAL SYSTEM DIAGRAM**







# **OPTIONAL I/O EXPANSION VIA WIDE AREA INTERFACES**

#### MADI

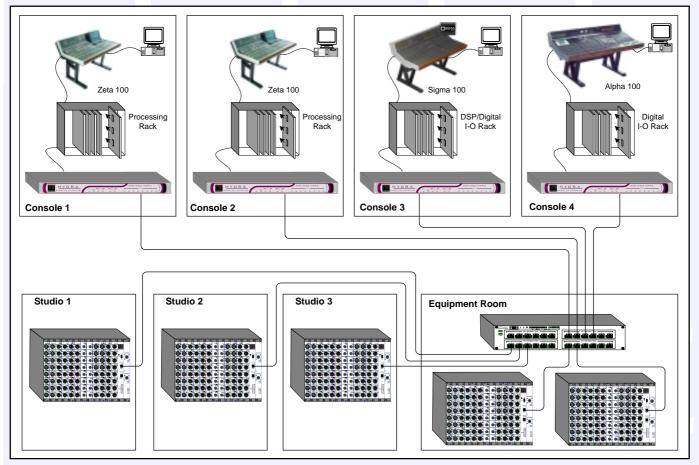


The rack mounted MADI unit contains two independent AES10 MADI compatible interfaces, and is available as an option. The two ports are interfaced to the Alpha 100 system via a Wide Area Bulk (WAB) card, which occupies one of the AES card slots in the Digital I/O Rack. Each MADI interface can operate in either 56 or 64 channel mode and can transmit over a coaxial AND optical medium and receive over a coaxial OR optical medium. A switch allows receiver selection. There is no sample rate conversion available on MADI inputs or outputs therefore, all the equipment connected via MADI must be synchronised to the same source as the console.

### **HYDRA**



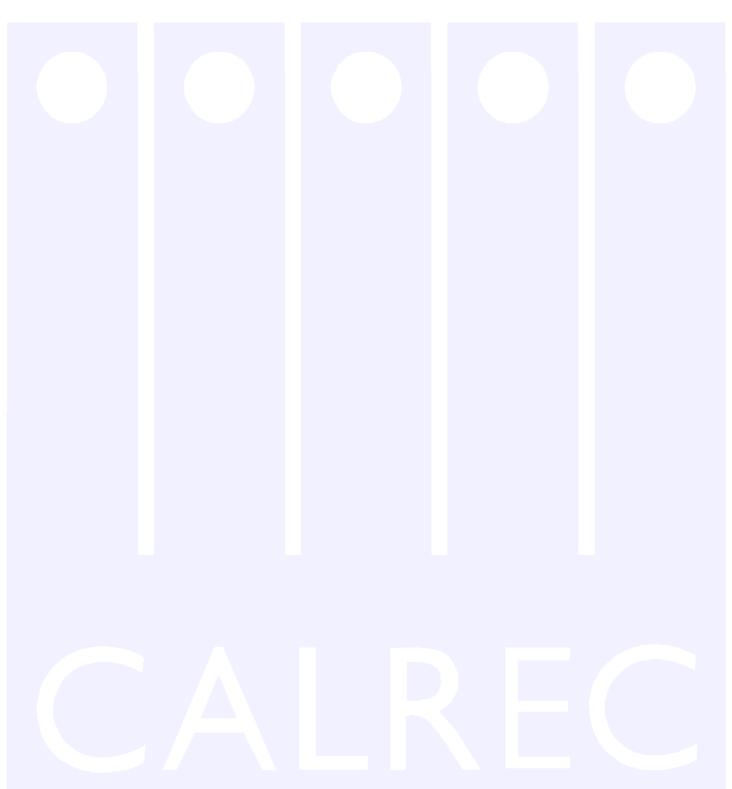
The Hydra audio networking system provides a powerful network for sharing of I/O resources and control data between Calrec digital consoles. Remote I/O units, with up to 96 inputs/outputs, analogue or digital, may be connected onto the network, providing remotely located sources and destinations that can be used by any or all mixing consoles. The Alpha 100 system interfaces to the Hydra gigabit interface unit shown above, via a Wide Area Bulk (WAB) card, which occupies one of the AES card slots in the Digital I/O Rack.





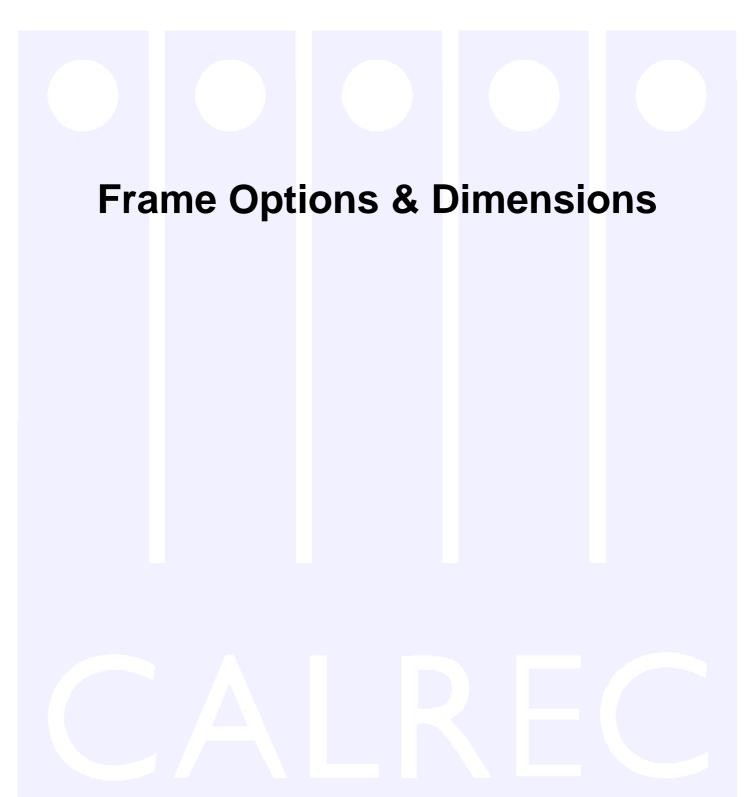
. . . . . . . . . . . . . . .







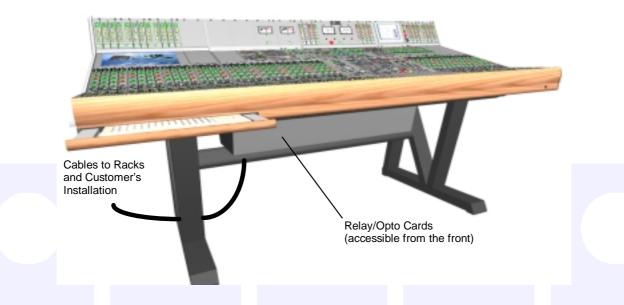








# CONTROL SURFACE FRAME SIZES



Frames are made up of sections which can be 4, 5 or 6 modules wide. This allows many different sizes of console to be achieved using different combinations of different sized sections. Fader modules have 4 faders each, so console size can depend on the number of faders required.

The table below shows the dimensions of the standard frame sizes available. Sections within the frame do not have to be in the order shown. For details of custom frames, with wedge sections etc, please contact Calrec.

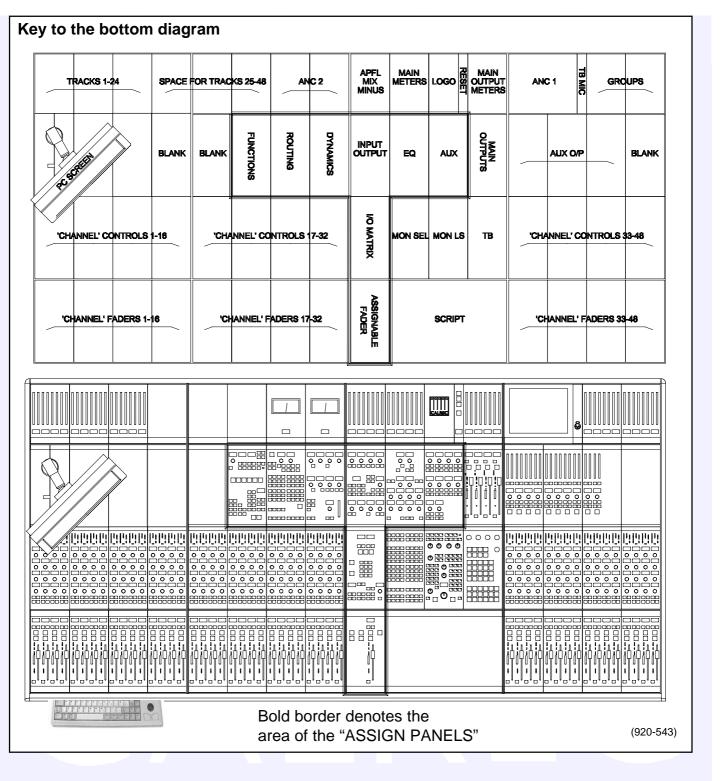
No of Modules	Frame	Leng	gth	Depth		
Wide	Frame	inches	mm	inches	mm	
12	4:4:4	60.9	1547	43.2	1098	
13	4:4:5	65.9	1672	43.2	1098	
14	4:4:6	70.8	1797	43.2	1098	
15	4:5:6	75.7	1922	43.2	1098	
16	4:6:6	80.7	2047	43.2	1098	
17	5:6:6	85.6	2172	43.2	1098	
18	6:6:6 90.5		2297	43.2	1098	
19	4:4:5:6	95.7	2428	43.2	1098	
20	4:4:6:6	100.6	2553	43.2	1098	
21	4:5:6:6	105.5	2678	43.2	1098	
22	4:6:6:6	110.4	2803	43.2	1098	
23	5:6:6:6	115.4	2928	43.2	1098	
24	6:6:6:6	120.3	3053	43.2	1098	
25	4:4:5:6:6	125.4	3184	43.2	1098	
26	4:4:6:6:6	130.4	3309	43.2	1098	
27	4:5:6:6:6	135.3	3434	43.2	1098	
28	4:6:6:6:6	140.2	3559	43.2	1098	





The smallest frame size is 12 modules wide, made up of 3 sections (4:4:4) and usually provides 32 faders with two audio paths on each fader. This allows up to 64 "channel faders" within a frame only 1547mm (60.9 inches) wide. The largest frame size is 28 modules wide, and made up of 5 sections (4:6:6:6:6) which provides 96 faders with two audio paths on each fader within a frame only 3559mm (140.2 inches) wide. This would give more faders than the maximum number of available paths, so some B layer faders would not need to be used.

The example below shows a typical 48 fader layout.



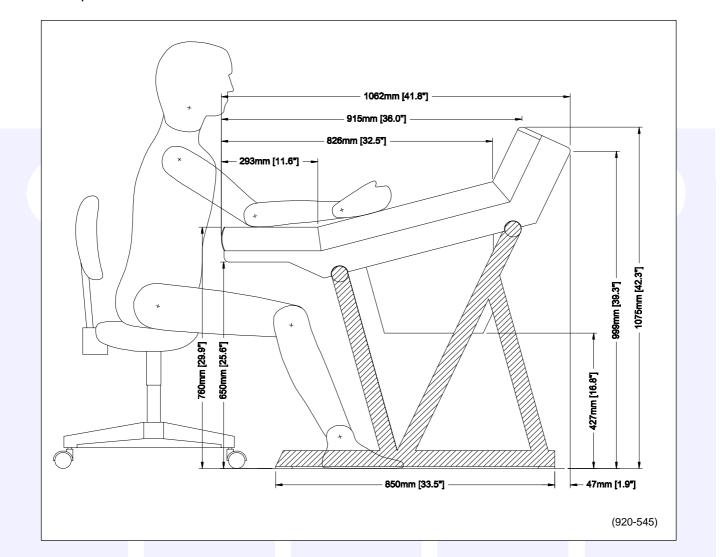
17





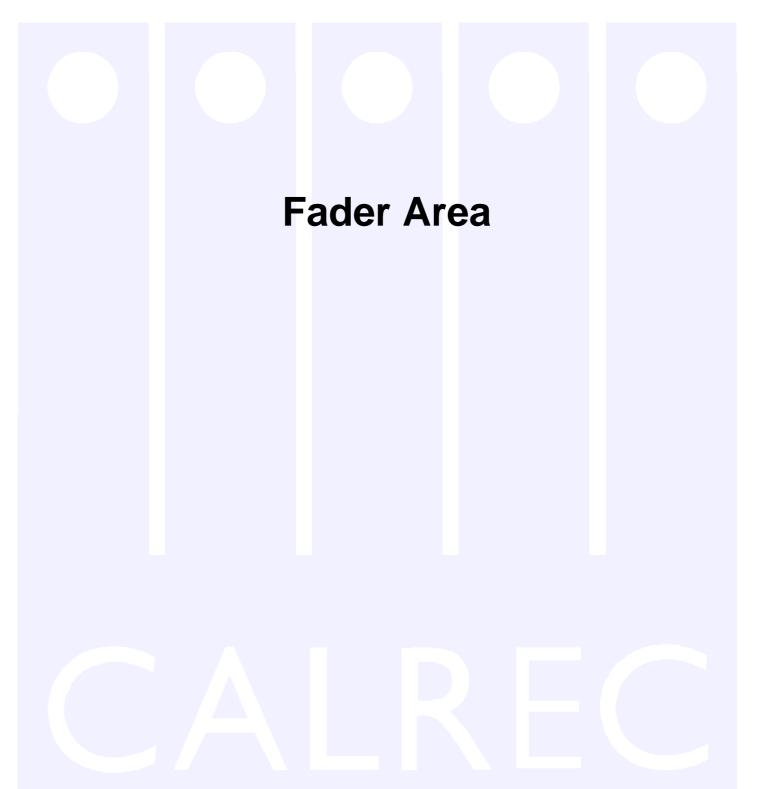
# **END PROFILE**

The end profile dimensions are the same for all frame sizes.









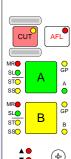




# "CHANNEL" FADERS

 $( \mathbf{F} )$ 

Channel and group paths are controlled by the console's "Channel" faders. Each fader can control two independent audio signal paths, named A and B. Any fader can control any channel or group path. Main output paths are controlled by their dedicated faders on the Main Outputs panel.



**Т** 

O/P O

4 0 E 8 0 T

EQ 🛑 30 FLTR 🔵 30

> 50 60

> > (+)

DYN 🔿 M/S 🛑

PFL

IU5219

10

5

10

Lead

The A & B buttons are used to select either of the two channel paths A & B. Selecting a path will "call" the fader to the Assign panels. Any changes made to the Assign panels will affect the selected path only. When switching between the two paths, the indicative displays and fader position will change to match the settings of each path.

The label in the display is the name associated with the input assigned to the path, or the group number if the path is a group. The input labels default to the Port ID unless a name is entered via the PC. Path A's label is shown in the top half of the display, and path B's label is shown in the bottom half of the display. The colour of the display indicates the active path. If path A is active, the label will be green. If path B is active, the label will be amber.

The CUT button cuts the channel or group. Its effect is the same as fading out the channel or group. Alternatively, there can be ON buttons which switch the channel on.

AFL will be heard through the monitor loudspeakers (main or small). AFL will be heard in surround if surround panning is in use and the loudspeaker system is surround.

### Assign Button LEDs

MR

SL

GP

ST

SS

A B

- The fader path is a Master of a VCA style group.
- The fader path is a slave within a VCA style group.
- A group is assigned to the path.
- The path is a stereo channel or group.
- Not Used.
  - Path A is active
  - Path B is active

The ▲ and ▼ Null LEDs will only illuminate when the position of the fader is not the same as the level of the audio. For example, if a VCA Master is moved away from the `0' posi-

tion, the null leds on the slaves will light. When illuminated they indicate whether the audio is above or below the position of the fader.

The **T** LED indicates that the console has recognised that the fader has been touched. The PEAK LED will illuminate if the channel or group signal is within 3 dB of the clipping level. The ON LED illuminates when the audio level is not at the  $\infty$  position.

The fader bargraph indicates the level at the channel input (post the input gain & switching and the tone switching), the channel direct output, or the gain reduction of the dynamics, indicated by the three LEDs. Selection is made either on the Functions panel, or using the PC.

The EQ, FLTR, DYN and M/S LEDs indicate that these functions are active.

PFL is provided on the fader overpress and on the button. It will be heard on the small LS (or the main LS if PFL to Mon is selected), or PFL LS (depending upon how the monitoring is configured).





## "CHANNEL" CONTROL

The "Channel Control" section is situated directly above the channel fader section. A set of LED's provide indication of :

- Routing to groups and mains
- The currently selected input type (mic, analogue line, or digital)
- If the Sample Rate Convertor (SRC) is switched in (for digital inputs)
- Routing to any track
- Whether the direct output is being fed with a mix minus feed
- The currently active fader path A or B

This section houses four Wild controls per fader. Almost any assign panel rotary control for the selected path can be assigned to a Wild control, including:

- Input Gain
- Aux Send Level
- Direct Output Level
- Pan and Balance Track Output Level
- EQ **Dynamics**

Stereo Width

Wild controls are assigned either using the Functions panel or the USER - CHAN screen.

Once assigned, the four Wild controls "FLIP" with the fader providing the same function for each of the two paths. The A & B faders may also be assigned to a Wild control, in which case it will be the opposite fader which is being controlled. The colour of the Wild control display will show which fader the control is related to: Green for A, Amber for B.

### Wild Control Push-Switch Option

If a wild control has the Aux Send or Front Pan controls assigned to it, the user can control the ON/OFF or IN/OUT status of these controls using the Wild control pushswitch. This feature is optional, so it can be enabled or disabled using the MISC screen.

### **Button Options**

Depending on the options purchased, the two buttons beneath the rotary controls can perform different functions.



#### Option 1

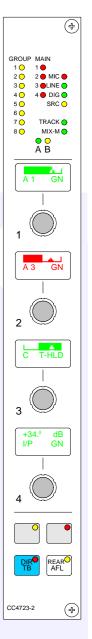
Each channel path can select between two inputs using buttons 1 and 2. There are buttons on the Input/Output panel to allow selection between inputs 1 and 2, but as an option, these buttons can be duplicated on this panel. This can be either two buttons (shown left), or just one (shown below).



#### Option 2

The ALT WILD button allows switching between two complete sets of alternate Wild settings. This would then allow up to 8 available Wild controls per fader. I/P 2 allows selection between inputs 1 and 2 with just one button. Input 1 is selected when the button LED is off, and input 2 is selected with the button LED on.

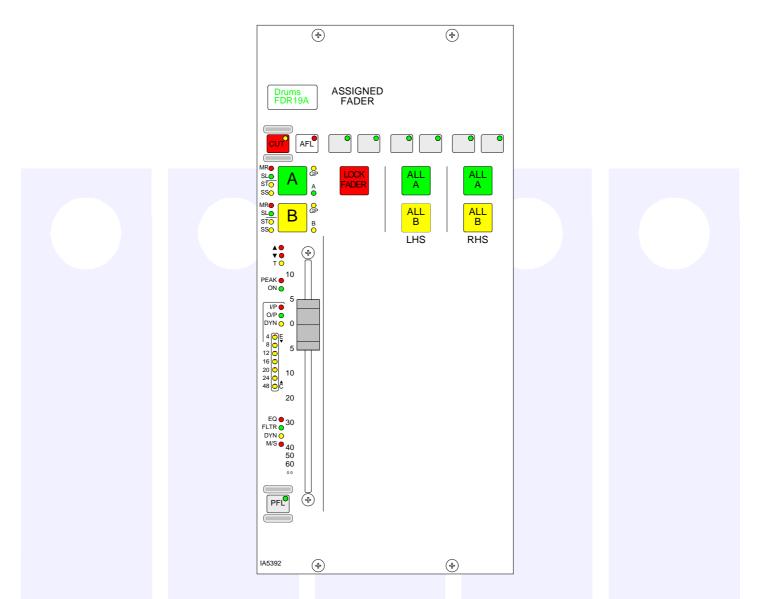
Talkback is available to direct outputs using the DIR TB button. All Talkback buttons are subject to On-Air inhibits, set up on the TX/REH screen.







# **ASSIGNABLE FADER**



The Assignable Fader is positioned towards the centre of the console, in the optimum listening position, and works in parallel with the currently assigned channel or group fader. Alternatively, LOCK FADER allows it to be fixed to a specific path.

The ALL A and ALL B buttons switch all the channel faders to display either their A path or their B path. Using the ALL A and ALL B buttons is like moving to a different section of a single layer design.

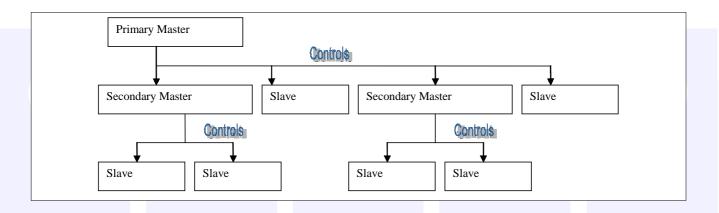






# VCA GROUPING

VCA groups allow the audio level, CUT, AFL and PFL functions of several slave faders to be controlled from one master fader. A VCA group is made or edited by holding down the Assign Button (A or B) of the fader to be master and pressing the Assign Buttons of faders to be added or removed as slaves. It is possible to select a VCA master as a slave of another VCA group. When this happens, the slave master is known as the secondary master, and it's master is known as the primary master.



When the level of a primary master is adjusted it will change the audio level of its slaves and the levels of its secondary master's slaves by the same amount. The CUT, AFL and PFL settings will also be applied to all the slaves, secondary masters and their slaves.

When the level of a secondary master is adjusted, the audio level of all its slaves changes by the same amount. Its adjustment will not affect the level of the primary master or *its* slaves. Changing the CUT, AFL or PFL of a secondary master will also apply the settings to the secondary master's slaves only.

The number of slaves in VCA group with a primary master would include all the primary master's slaves and the slaves of all its secondary masters. There can be up to 48 members of a VCA group.

It is possible to create the primary or secondary group in any order. A slave can be made into a secondary master by adding slaves to it. The path on that fader will be removed from primary master and become a slave of the secondary master. If a slave added to the VCA group is already a master it will become a secondary master.

The MR and SL LED's next to the Assign buttons on the fader panel indicate the VCA group status of that fader. A secondary master fader has both the MR and SL LED lit.

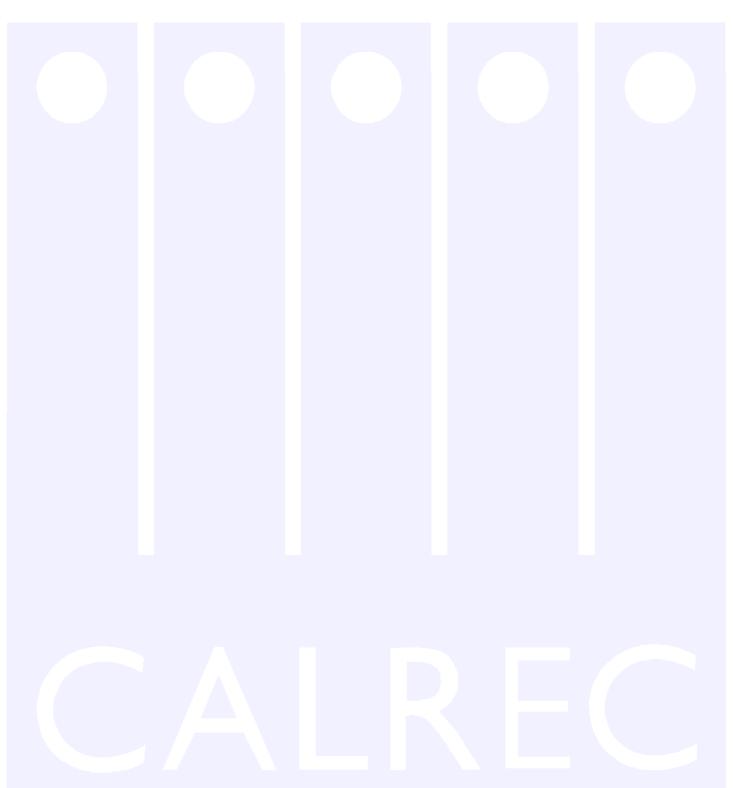
Interrogation provides a clear way of indicating VCA group assignments. Interrogation is performed by holding down the Assign button of a VCA group member. Interrogation of a Primary Master will light the Assign buttons of it's primary slaves and secondary masters. Interrogation of a secondary master will light the Assign buttons of it's secondary slaves, and the primary master's Assign button will flash.

The editing of VCA groups can be enabled and disabled using the PC. This provides protection against accidental changes.



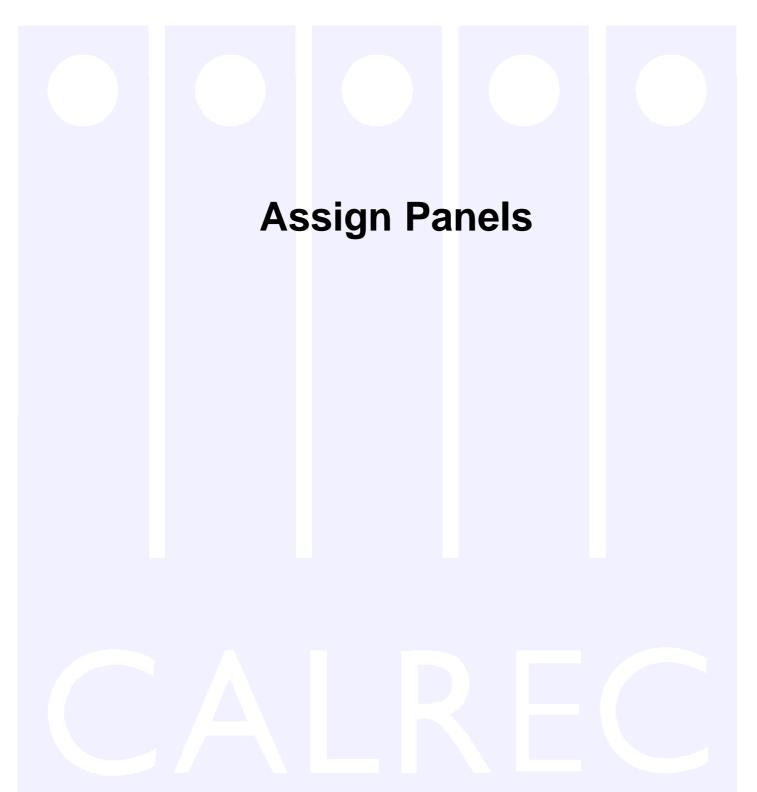
.....















# I/O MATRIX

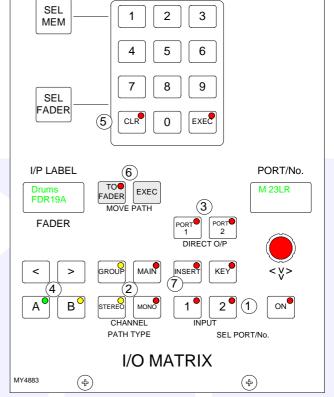
Each channel path can select between two input ports. Ports are assigned to inputs 1 and 2 for the currently assigned fader using the I/O Matrix.

## (1) Port Assignment

- Press 1 or 2 to select an input.
- Use the rotary control to scroll through the lists of available input ports.
- Upon reaching the desired input port, press the ON button to assign the chosen input port to the input 1 or 2.
- Pressing ON again will de-assign the port.

#### Lists

Pressing and turning the rotary control gives access to lists of other types of input port which are set up during installation of the console. Each port can be allocated to one of a number of lists to allow I/O which is wired for similar purposes to be grouped together for selection.



It is possible to determine which lists of input ports appear for selection on the I/O Matrix panel using the Options-Misc screen. Making only the relevant lists available for selection makes it easier to find the ports you want.

Port assignment can also be done using the I/O screens.

### (2) Path Type Selection

The GROUP, STEREO and MONO buttons select the path type for the currently assigned fader. If the path is to be a group, its number is selected using the rotary control & ON button. The path type can also be selected using the USER-CHAN screen.

### (3) Direct Output Ports

Ports can be connected to channel and group direct outputs, first by selecting PORT1 or PORT2, and using the rotary control and ON button to choose and select ports. (Two ports can be connected to each direct output). When scrolling through the lists of output ports, those that are in use will display "IN USE".

### (4) Fader Path Selection

In addition to the Assign buttons on the fader panels (A & B), fader paths can be called to the Assign panels using the nudge buttons to scroll through the faders, and the A and B buttons to choose the path. This is for use when pressing the fader assign button is not convenient, or should a fault develop on the fader strip. Paths can also be selected by pressing SEL FADER and entering the fader number on the keypad.





# (5) Clearing Paths

Channels can be cleared off the fader by pressing SEL FADER, CLR and EXEC. This will clear all settings and port assignments from the channel.

## (6) Moving Paths

Paths can be moved or swapped from one fader to another, using the MOVE PATH buttons. To move paths, select the assign button of the path you wish to move, and press TO FADER (the assign button will flash). Then select the assign button of the destination fader, and press EXEC to move the path. This function is also available using the USER-CHAN screen.

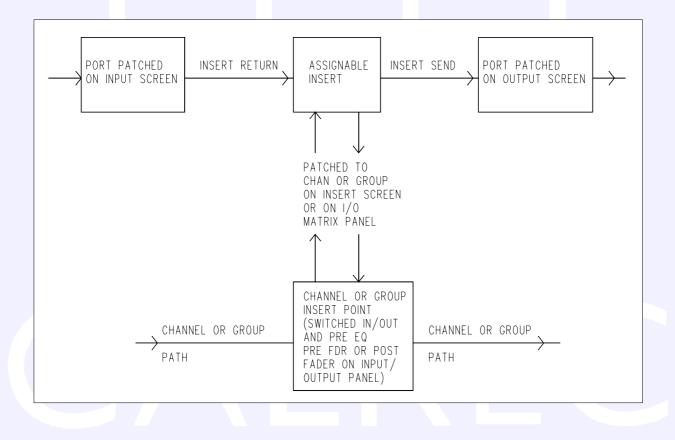
## (7) Channel & Group Inserts

The system provides up to 24 L-R pairs of assignable inserts which can be used in the stereo and mono channels and groups. In addition, the main outputs have their own dedicated inserts.

Assignable inserts are designed to be pre-connected to send and return ports which are in turn pre-wired to insertable devices or to an insert patchbay (normally there would be some assignable inserts of each type). The Input and Output screens allow send and return ports to be set up for the assignable inserts.

Pressing the INSERT button allows the rotary control and ON button to control selection of inserts on channels and groups. This selection can also be made using the Insert screen. The insert is then patched in and out of the channel or group path using the buttons on the Input/ Output panel.

The assignable inserts can be divided into up to 4 lists in a similar way to input and output ports. This separates them for selection on the pot-switch.







# **INPUT/OUTPUT CONTROLS**

The INPUT controls in the Input/Output section allow separate settings for the two channel inputs and gain, and ON/ OFF for the group & main direct inputs.

### (1) Input Settings

Each channel path can switch between two input ports using buttons 1 and 2. Optionally, each fader can have dedicated selection buttons for inputs 1 and 2 on its channel control panel.

SRC switches the sample rate converter on AES inputs.

48L & 48R switch phantom power on Mic/line channel inputs. 48L is used for mono channels.

LB & RB provide Left to Both & Right to Both on stereo channels and groups.

M/S converts a sum & difference (mono/stereo) input to L & R on stereo channels.

ØL and ØR buttons reverse the phase of the channel inputs. ØL is used for mono channels.

The TONE button switches tone to the input of the channel or group, from where it can be routed as required.

### (2) Gain Adjustment

Comprises 2 buttons for coarse ranging plus a knob for fine adjustment. Pressing both buttons at the same time sets the gain to 0 dB. For a group or main path, the controls set

the gain of the direct input. Gain is adjustable from -18dB to +78dB for mic/line inputs, -18dB to +24dB for digital inputs, and  $\infty$  to +10dB for direct inputs.

It is possible to link the gains of inputs 1 and 2. When the gains are linked, if either gain is adjusted, the change in gain is applied to both inputs. The lower and upper level endstops still apply, and are dependent upon the input type. If one of the inputs reaches an endstop during adjustment, this will stop both gains going any lower or higher.

### (3) Balance Control

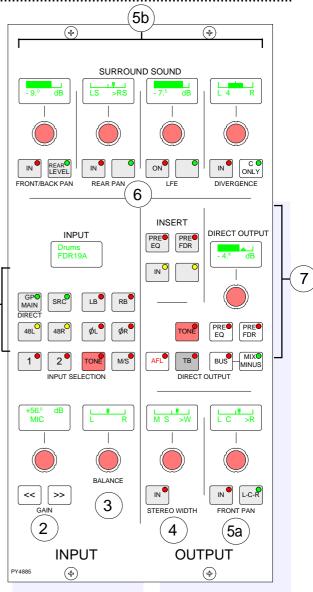
Operates on stereo channels only. When LB & RB are selected, the balance control acts as an input pan control.

### (4) Width Control

Operates pre fader on stereo channels and groups. The rotary control adjusts the width from mono, through stereo, to wide. The control is switched in and out of the path using the IN button.

### (5a & 5b) Stereo and Surround Panning

Stereo and surround panning is provided for channels and groups. Signals can be panned to both stereo groups and 5.1 outputs simultaneously. AFL can be heard in surround, post the pan controls, provided that the monitoring is surround. Stereo and surround panning controls are strengthened with the inclusion of the optional motorised joystick panel.







The Front Pan allows the front signal to be panned from left, through center, to right. On stereo channels and groups, the L-R PAN acts as a balance control.

The Front/Back pan control pans the signal between Front and Back. When Rear Level is switched IN, the levels to the rear and front are controlled seperately. This allows signal to be fed to the rear without affecting the balance of the mix in the front speakers. Also, the front signal can be turned off and a level set to the rear which is different to that being sent to any stereo groups or mains which the path is feeding.

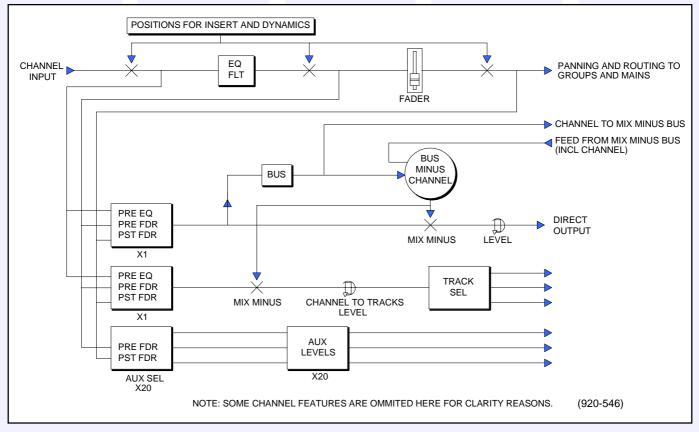
The divergence controls set an amount of the centre signal to also feed L & R. Divergence does not operate on stereo channels and groups. The C ONLY button connects the channel output to the centre BUS only. All other panning controls are disabled. The channel is fed to both L & R of stereo busses. On stereo channels and groups, C ONLY feeds a mono reduction of the stereo signal to the centre bus only.

### (6) Inserts

Assignable inserts can be patched in and out of the channel path, using the IN button. The buttons allow the insert to be patched post fader, pre fader or pre EQ. Assignable inserts must first be set up using the I/O Matrix or I/O screens.

### (7) Direct Output and Mix Minus

In the direct output section, the BUS button feeds the direct output signal to the mix minus bus. The output of the mix minus bus feeds back into the channel (or group) where the channel's signal is subtracted. The MIX MINUS button then feeds the resulting signal to the direct output. Therefore, every channel can produce a mix minus output which is a mix of all the channels routed to the bus apart from itself. MIX MINUS & BUS are independent buttons, so the track routing selector and the direct output can be fed with the mix minus bus, even if the channel is not feeding the bus.







### **FUNCTIONS CONTROLS**

### (1) Assigning Wild Controls

The Wild controls are assigned either from this panel, or from the USER CHAN screen. All the Assign panel rotary controls incorporate a switch which is operated by pushing the control. These switches are used to assign the control to a Wild control as follows:

- Select the required fader by pressing it's Assign Button (A or B).
- Select WILD ASSIGN 1, 2, 3 or 4.
- Push one Assign panel rotary control. For example, Aux 1 Send.

CLR will clear the selected Wild control from it's assignment.

It is possible to assign controls to more than one fader path at a time, either by selecting individual fader assign buttons (A or B), or by defining a "block" or "Region" of faders.

After selecting Wild 1, 2, 3 or 4, press HOLD, then a number of fader paths can be selected individually by pressing their fader assign buttons (A or B). Pushing an Assign panel rotary control will assign that control to all selected faders.

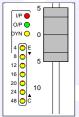
A block or region of faders can be defined by holding down HOLD and then pressing the fader assign buttons of the first and last fader path in the required region. Pushing an Assign panel rotary control will assign that control to all fader paths in the selected region.

It is possible to assign the same control to Wilds 1, 2, 3 or 4 for all fader paths by selecting ALL before pushing the required Assign panel rotary control.

The gains of the two inputs 1 and 2 can be assigned seperately to Wild controls, by holding down the required input button on the Input/Output panel before pushing the gain adjustment rotary control.

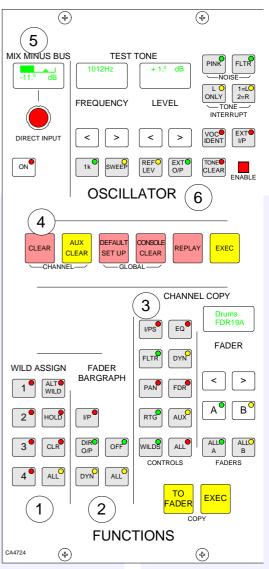
The ALT WILD button allows switching between two complete sets of Wild settings. This would then allow up to 8 available Wild controls per fader.

Aux output controls cannot be assigned to Wild controls. If the fader is touched instead of pushing an Assign panel rotary control, then the fader for the alternate layer will be assigned to the Wild control. If the track output level control is assigned to a block of Wild controls, each fader's Wild control will have a different numbered track output level control, beginning with the track currently selected on the first fader in the block.



#### (2) Fader Bargraph Assignment

Buttons I/P, DIR O/P, DYN and OFF on this panel will set the function of the fader bargraph for the currently assigned fader, to either the channel input (post the input gain & the tone switching), the channel direct output, or the gain reduction of the dynamics. If ALL is pressed first, all fader bargraphs will be set to the selected function. Fader Bargraph assignment is also definable on the USER-CHAN screen.







# (3) Channel Copy

Nine sections of a channel or ALL together can be copied to another channel or channels using this panel. The nudge buttons (< and >), plus A & B, can select the channel to be copied by calling it to the Assign panels.

TO FADER (flashes) allows the destination/s to be chosen. Multiple destinations can be selected on the Assign Buttons, or by using the ALL A or ALL B buttons.

The nudge buttons (and the keypad on the I/O Matrix panel) can select an individual destination, which can be in addition to any multiple destinations set. Once all the destinations have been chosen, the EXEC button executes the Copy.

If a stereo channel's settings are copied to mono channels, only the relevant settings will be copied. Other settings on the mono channels will be reset to the cleared down state. If groups or main outputs are included in the selected destinations, they will simply be ignored.

- I/Ps copies the LB, RB, ØL, ØR, M/S & balance settings (only Ø for a mono channel) for inputs 1 & 2, and also the input gains, SRC or phantom power when the inputs are of the same type.
- EQ and FLTR copy the EQ and filter settings including IN/OUT, Alternate and assignment (CH or Dyn) settings.
- DYN copies the dynamics settings but not whether the EQ or filters are switched in the dynamics.
- PAN copies pan and width settings as appropriate.
- FDR copies the fader and CUT switch settings but not PFL or AFL selections. It does not copy VCA group assignments.
- RTG copies the routing to main outputs and groups but not the routing to tracks.
- AUX copies the routing and levels to the auxiliaries.
- WILDS copies the Wild assignments but not their settings.
- ALL copies all of the above.

Copy functions can also be executed using the Copy screen.

### (4) Console Functions

Clear, Aux Clear, Default Set-up and Console Clear flash when pressed and require the EXEC button to be pressed before the operation is carried out. It is recommended that settings are saved to memory before these functions are used. Channel Clear clears the currently assigned channel from all settings apart from the port assignment.

The Default set-up will usually be created upon installation of the Alpha 100 using the TECH-INFO screen. This is a default memory, which could contain the fixed port set-ups which match the studio wiring, and any other settings which hardly ever change. It could have all channel settings OFF or flat, with no routes made, and would be available as a start up memory, from which more specific memories could be created.

### (5) Mix Minus Bus & Direct Input

The mix minus bus and the direct input are switched ON or OFF using the button, and a rotary control is provided for level adjustment. The port for this is patched on the I/O Input screen.

(6) The Oscillator - The Oscillator controls are described on page 50.





## **ROUTING PANEL**

Routes for the selected channel can be made or removed by pressing the numbered buttons on the routing panel.

To route several adjacent channels to one bus, the nudge buttons (on the Functions or I/O Matrix panel) can be used to quickly select the channels.

### (2) Interrogate Mode

It is possible to discover which fader paths are feeding each of the routing busses by putting the panel into "Interrogate" mode. This is done by pressing the INTER button. If any of the routing buttons (groups, mains, tracks) are held down, the fader assign buttons of all the paths feeding that bus will light. This button can also be used to interrogate mix minus feeds using the BUS button on the Input/Output panel.

### **Reverse Routing**

Paths can be added or removed from the bus under interrogation, by selecting or de-selecting their fader assign buttons.

### (3) Tracks (General Purpose Bus Outputs)

The Channel/Group to Tracks section controls the signal, from the channel or group, feeding the track routing selector.

PAN makes the control into a Pan control (Balance on stereo paths). Routing is Left to Odd tracks, Right to Even tracks.

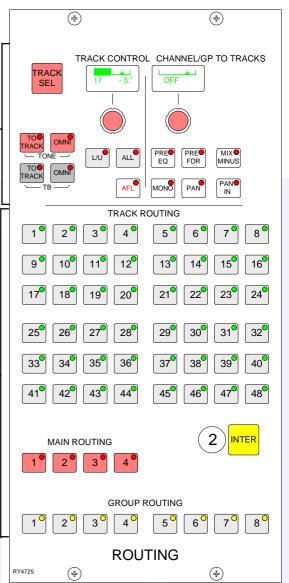
On stereo paths, the mono button monos the signal after the balance control. The mono signal can then be routed to any track. This allows the mix of L & R to mono to be adjusted for when the signal is dual mono.

The Mix Minus, Pre-EQ and Pre-Fader buttons act as a cancelling set. When none are selected the signal is sent to the track routing selector Post-Fader. Mix Minus feeds the Mix Minus signal of the channel or group, as set up on the direct output section of the Input/Output panel, to the Track Routing selector.

The Track Control section of the Routing panel, controls the output to the multi-track, after the track mix. These outputs can also be used as IFB or general purpose bus outputs. 48 optional bargraphs can be fitted to monitor the output level.

The track output being controlled is selected by the Track Sel button plus the track routing buttons 1-48. ALL makes the control a master, controlling all the tracks at once.

Tone or Talkback can be fed to the selected track output. The OMNI buttons feed tone or talkback to all the track outputs.







# TALKBACK

Talkback is available to all groups, mains, auxes and 8 external sources (via relay switching) using the buttons on this panel. Talkback is also available to direct outputs and individual tracks using the buttons on the channel control panels, Input/Output panel and Routing panel.

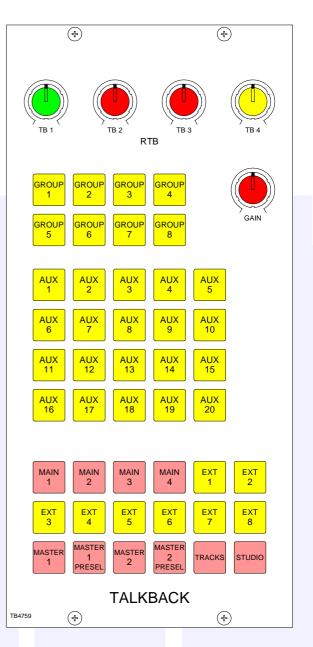
MASTER 1 and 2 operate all the TB buttons which have been preselected by the respective PRESEL button.

All Talkback buttons are subject to On-Air inhibits, set up on the TX/REH screen.

The GAIN control sets the level of the talkback microphone.

TB1 to TB4 set the level of 4 RTB (Reverse Talkback) signals.

There can be a mix of all four signals to feed a single loudspeaker. This can mix with the PFL feed to the PFL loudspeaker.

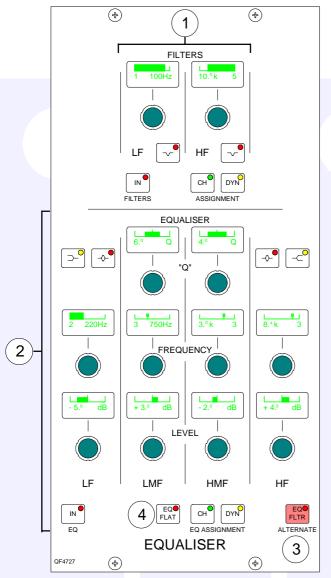






# **EQ & FILTERS PANEL**

The Equaliser panel controls EQ & Filters on the channel paths only. Excessive control ranges are deliberately avoided to simplify operation. Once a channel has been selected by pressing it's Assign button (A or B), it's frequencies can be adjusted using the following controls.



# (1) Filters

LF 12dB/octave plus notch, 20Hz to 330Hz HF 12dB/octave plus notch, 3.3kHz to 20kHz

### (2) Equaliser

LF	30Hz to 470Hz, shelf or bell (Q of 1)
LMF	160Hz to 2.4kHz, Q from 0.3 to 10
HMF	500Hz to 7.5kHz, Q from 0.3 to 10
HF	1kHz to 16kHz, shelf or bell (Q of 1)

EQ level controls are adjustable by ±15dB

EQ and Filters are switched in and out of the signal path using the IN buttons in each section.

The EQ and Filter sections each have two assignment buttons. The CH buttons ensure that the EQ and Filters are switched into the assigned channel's path, and the DYN buttons allow the EQ and Filters to be switched in and out of the dynamics of the assigned channel. These buttons are not mutually exclusive, EQ and Filters can either be in the channel path or the dynamics, but not both at the same time. Selecting DYN will de-select CH and viceversa.

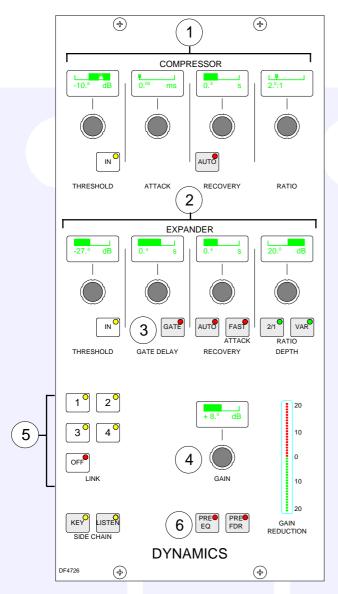
- (3)The ALTERNATE EQ FLTR button allows switching between two complete sets of EQ and Filter controls.
- (4) EQ FLAT will clear any EQ settings to flat.





# DYNAMICS PANEL

The Dynamics panel controls Compressor and Expander or Gate, on Channels and groups, and Compressor on main outputs. Once a channel has been selected by pressing it's Assign button (A or B), it's dynamics can be adjusted using the following controls.



# (1) Compressor:

Threshold +20dB to -20dB

Recovery 75ms to 4 sec + AUTO Ratio 1 to 50 Attack 50µs to 5ms

# (2) Expander:

Threshold 0dB to -40dB Recovery 75ms to 4 sec + AUTO Depth 0dB to 40dB Fast attack 300µs (normal 16ms) Ratio 2/1 and VAR (variable - according to level)

### (3) Gate:

Threshold 0dB to -40dB Recovery 75ms to 4 sec + AUTO Depth 0dB to 40dB Fast attack 300µs (normal 16ms) Gate delay 0 to 1 sec in addition to 6dB hysteresis

(4) Make up gain is adjustable from 0dB to +20dB.

- (5) It is possible to have the dynamics of many channels linked by assigning them to one of four available link busses. This is useful for when the same dynamics settings need to be applied to more than one channel, for example, when 4 channels represent a 5.1 signal. With the channel selected, press 1, 2, 3 or 4 to assign the channel to the bus.
- (6) The dynamics can be applied Pre EQ or Pre Fader. The Pre EQ button will not function on group & main paths.

A 0dB setting on the dynamics equates to the chosen reference level for the console.





### AUXILIARY CONTROL PANEL

The Auxiliaries panel controls the feeds from the channels or groups to the auxiliary output busses.

The ON button switches the feed from the currently assigned channel or group to that auxiliary output buss. Each feed is post the channel or group fader, but can be pre fader, selected using the PRE button.

There are 20 mono auxiliary output busses, which can be paired up to be used as stereo auxiliary output busses. The busses are pre-set to be mono or stereo on the USER-BUSSES screen. If, for example, aux 9 is set to be stereo, then aux 19 will not be available (and Aux 19 will not work on the Monitor Selector panel). When a pair of auxes are changed in this way, all settings of the pair are cleared.

On mono auxiliaries, buttons 11 to 20 switch the control to that numbered aux send.

On stereo auxiliaries a dual level display will be shown. For example, aux 9 & 10. Here buttons 19 & 20 will be inoperative.

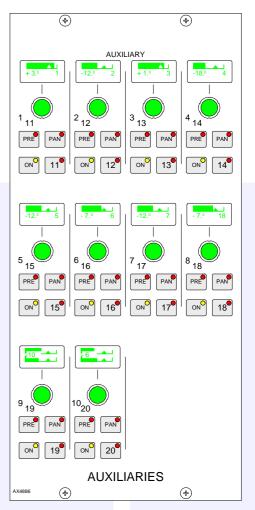
PAN makes the control into a Pan control (balance on stereo channels). Any pan offset will be shown as an offset between the two bars of the display.

The auxiliary output levels and direct input switching and levels are controlled on the Auxiliary Output panels.

Options are available for the pre-send to be cut :

- when the channel or group is cut
- when the channel or group fader is closed
- when the fader is open and not cut (Bird Beater). This option cancels the other two pre-send cut options.

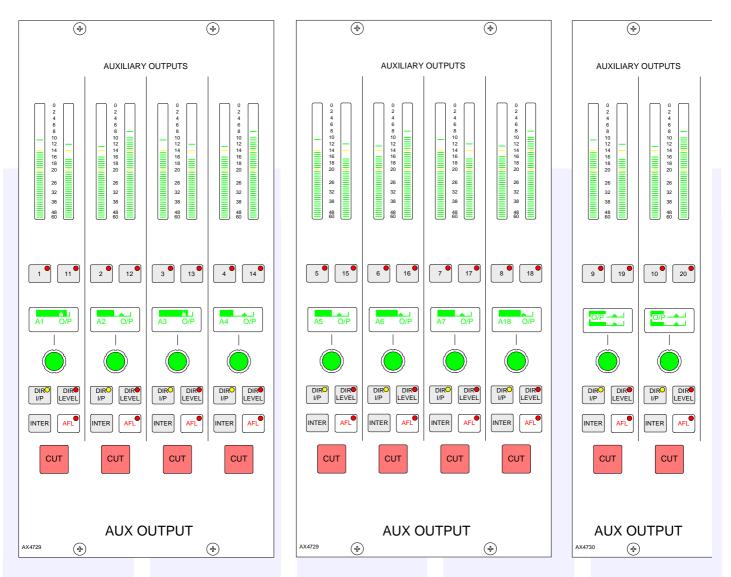
Stereo	0	Π			0				×.	×
	1	2	з	4	5	6	7	8	9	10
Pre-send cut when chan/gp cut	G			0	0.			0	Ο	
Pre-send cut when chan/gp fader closed										
Bird Beater: Pre-send cut if fader open and not cut										
	11	12	13	14	15	16	17	18	19	20
Pre-send cut when chan/gp cut										
Pre-send cut when chanigp fader closed										
Bird Beater. Pre-send cut if fader open and not cut										







## AUXILIARY OUTPUT PANELS



These panels control the auxiliary outputs.

The displays above each rotary control show what is being controlled (e.g O/P or DIR) until they are adjusted, when the level is then displayed. A short time after the adjustment has been made, the display will show the label again.

On stereo auxiliaries a dual level display will be shown, for example, aux 9 & 10. Here buttons 19 & 20 will be inoperative. There cannot be a level offset on the output display.

It is possible to discover which fader paths are feeding each of the aux output busses by holding down the Interrogate button (momentary). The fader assign buttons of all the paths feeding that bus will light. Paths can be added or removed from the bus under interrogation, by selecting or de-selecting their fader assign buttons.

DIR I/P switches on the direct input to the auxiliary bus.

DIR LEVEL makes the Aux output control into the direct input level control.





### MAIN OUTPUTS PANEL

The Assign Button on each main fader calls the main output to the Assign panels to allow; routing (of one main to another - indicated on the routing leds above the faders), insert on/off, and control of the Compressor and direct input.

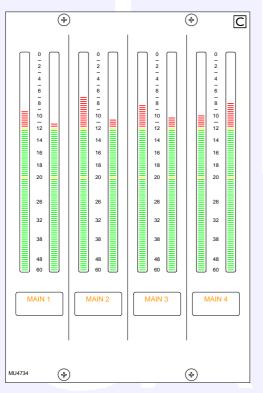
Each main output can be pre-set to be either surround or stereo. Surround mains are 5.1 plus a rear downmix to allow a simultaneous LCRS. There is also a stereo downmix and a mono downmix (potentially 10 outputs for each surround main).

The insert and direct input are also surround.

If a surround main is routed to a stereo main, the stereo downmix will be routed.

(4)	)		<b>(</b>
MAIN 1 5.1 TONE	MAIN 2 5.1 TONE M2	MAIN 3 5.1 TONE M3	MAIN 4 STEREO
STEREO SURROUND	STEREO	MAIN 2 STEREO SURROUND	MAIN 2 STEREO SURROUND
PRAK 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0	PREAC PREAC DVN0 10 4 0 E 8 0 C 20 48 0 C 30 40	PRE 0 DVN 10 4 0 0 0 0 0 0 0 0 0 0 0 0 0 0	PEAK 0 0 M4 PEAK 0 0 M4 PRE0 10 4 0 E VVN0 10 4 0 E 10 0 20 20 24 0 20 24 0 20 20 24 0 20 20 24 0 20 20 24 0 20 20 24 0 20 20 24 0 20 20 20 20 20 20 20 20 20 20 20 20 2
DYN 🔿	DYN •	DYN O	DYN O
60 70 ° °	60 70 ° °	60 70 ° °	60 70 ° °
PFL (+)	PFL +	PFL +	PFL +
	MAIN O	UTPUTS	
IM5415	)	1	4

#### **Main Meters**



The main output meters display the stereo downmix if the output is surround.

If the main line monitor is set to be fed back from the studio distribution via external inputs to the desk, then the meters will display this instead.



### **BROADCAST FACILITIES**

#### **Condition Switching**

There are three modes which the system can be in: Transmit (TX or On Air), Rehearse, or neither. These are controlled from the ON AIR and REH buttons or from external inputs set up on the OPTO screen.

The OPTIONS-TX REH screen allows the condition switching for the system to be set up, whereby many different functions can be set to be active, or not, in any of the three states. This can significantly reduce the risk of human error when in the various modes, making the whole system a more robust, less stressful, user friendly environment for operators to work in.

#### **Power Supply Monitoring**

The rack mounted PSU monitor module monitors the power supplies for failures, and ensures "hot" changeover to the spare should there develop a fault. The PSU FAIL Indicator/Cancel button on this panel will flash if any one PSU fails (the hot spare PSU would prevent the desk from being affected). Pressing this button will change the flashing to a steady lit condition. In this mode, in the unlikely event of a second PSU failing, the light will begin to flash again.





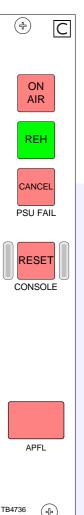
If a problem does develop, the PC will report this on the Automatic Warning and Correction System (AWACS) screen. The AWACS icon will flash to draw attention to the report.

Because the system has many back-up features, such as automatic change over to hot spares for power supplies, control cards & DSP cards, it is possible to continue operating after errors are reported. Message history is saved to the PC's hard disk for future analysis.

#### **Console Reset**

Pressing the CONSOLE RESET button resets the control system only. Independent DSP operation ensures audio continuity during console reset. The most recent console settings will be fully restored in less than 15 seconds.

As the console operates independantly of the PC, rebooting or failure of the PC will affect neither the audio nor the operation of the console.









### MONITORING, METER SELECT AND LS CONTROL

The Monitor and Meter Selectors are used to select the source to monitor, and what to display on the meters. Selectors 1 & 2 are sub-selectors which feed the other selectors. All Selector external inputs can be mono, stereo, or 5.1. Mono inputs are fed to L + R.

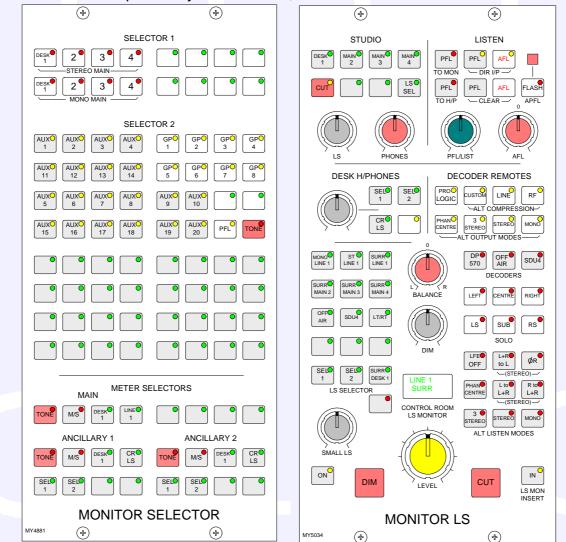
The SMALL LS level control is in series with the Main LS level control. The ON button diverts the monitor output to the small LS for near field, or domestic check, monitoring. Both main and small LS can be stereo, 3 stereo, or 5.1 independently.

DIM, CUT & SOLO operate on both sets of loudspeakers. DIM & CUT can be externally operated. DIM can be controlled from the TB (See Condition Switching (TX/REH) screen).

If a surround signal is monitored on a stereo loudspeaker or meter, a stereo downmix will be created and monitored. If the loudspeaker system is surround, stereo and mono sources will be heard in stereo and mono, with no signals on the other speakers.

If a main output is surround, the stereo monitor buttons for that main output will monitor the stereo (downmix) output of that main output. If a main output is stereo, the surround monitor buttons for that main output will be disabled.

For Studio LS, two parallel LS outputs are provided, post the level control, with separate Mic Open cuts. These can be independently either stereo, 3 stereo or 5.1.







### **Meter Selectors**

The main meter is in addition to the four stereo main output meters, which display the stereo downmix of the main output, if the main is surround.

The Ancillary 1 meter would normally be used to meter the Control Room LS monitor selection, but can also be selected to Main 1 Desk, Sel 1 or 2, or 2 external inputs.

Both the main and Ancillary 1 meters have a Tone switch to send Tone directly to the meter. They can both be stereo only, surround only, or surround plus stereo, with an optional separate M/S (L-R sum/difference) meter.

The Ancillary 2 meter can be selected to meter the Control Room LS monitor selection, Main 1 Desk, Sel 1 or 2, or 2 external inputs. It is stereo only with an optional separate M/S (L-R sum/ difference) meter. When metering surround signals, it displays the stereo downmix.

### **Alternative Listening Modes**

All off indicates NORMAL (mono, stereo or surround depending on the source selected and the LS arrangement).

3 STEREO with Phan Centre ON is the same as STEREO except the LFE is optional.

ØR, L+R to L, L to L+R, and R to L+R will work in any mode, but are really designed for use in STEREO mode or when monitoring stereo sources.

MONO feeds L, C, R, LS & RS to L + R.

## AFL & PFL

AFL feeds the Control Room LS outputs (post the surround panning controls), overriding the LS SEL. PFL can also do this if PFL TO MON is selected (overrides AFL). If PFL to MON is not selected, PFL can override the small LS (if it has been set to do this in the setup application). Alternatively, there can be a separate stereo PFL LS output. An external RTB input can mix with PFL to the PFL LS output.

PFL to H/P feeds the PFL signal to the headphones.

PFL clear & AFL clear, clear any latched buttons.

PFL from surround mains is a stereo downmix of the surround signal.

### **Deocoder Remotes**

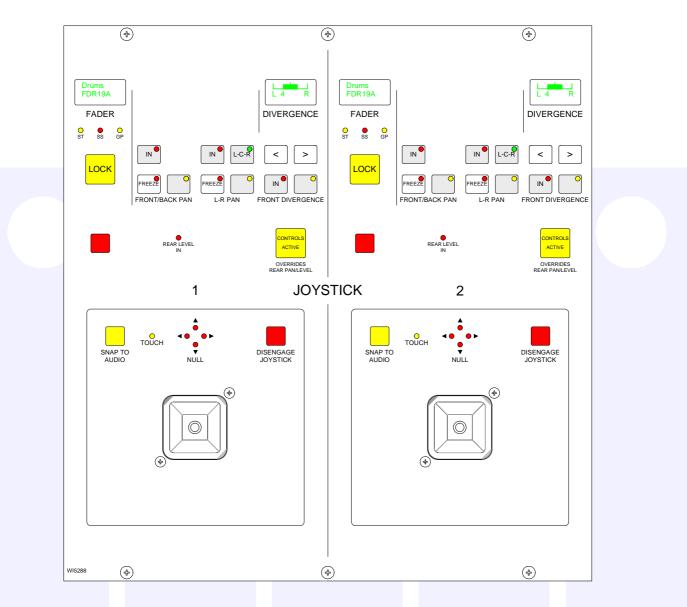
The DECODER REMOTE buttons control whichever Decoder is currently selected. (Other Decoders remain in their previously set state). The buttons are shown engraved for a Dolby DP570.

- 4 buttons for Alternate Output Modes (all off indicates Full surround).
- 3 buttons for Alternate Compression Modes (all off indicates no compression and no dialogue normalisation).
- 1 button for Pro Logic mode. It is assumed that the DP570 will be set on the unit, to Dolby Digital mode either in manual or auto detect mode.
- When controlling a Dolby SDU4, LT/RT decoder, only the stereo and mono output mode buttons will function.





# MOTORISED JOYSTICK PANEL (OPTIONAL)



The joystick panel is available as an option, and can be either a single joystick, or twin joysticks as shown above. The joysticks allow accurate stereo and surround panning of the channel.

The joysticks are touch-sensitive, and the TOUCH LED will light when the joystick is touched. In normal operation, the joystick controls the currently selected fader path (Chosen by pressing A or B on the fader panel). LOCK allows the joystick to be fixed to a specific path. Pressing LOCK again will unlock the panel. The fader display shows the path currently assigned to the joystick panel.

LEDs show the type of path being controlled:

- SS Indicates a surround sound main.
- ST Indicates a stereo source.
- GP Indicates a group.





Front/Back pan, L/R Pan and Front Divergence each have a set of dedicated controls. Each has an IN button to enable the function. The IN buttons and L-C-R button work in parallel with the buttons on the Input/Output panel. The blank buttons are there so that additional features can be implemented in the future.

CONTROLS ACTIVE must be selected for the joystick controls to take effect. When Controls Active is selected, the joystick moves to the position set by the Front Pan and F-B controls (including IN/OUT status). If the joystick is being touched when Controls Active is selected, then the audio will move to the position of the joystick. Any Rear Pan and Rear Level settings are disabled, and the Rear Pan and Rear Level displays on the Input/Output panel and any Wild Controls show "JOYSTK". De-selecting Controls Active does not restore any previous Rear Level or Rear Pan controls, but leaves the Rear Level switched out, and the Rear Pan at the same setting and IN/OUT status as the Front Pan.

If a blank fader or a main path is assigned, Controls Active is disabled. The divergence display will be blank and the buttons should be all off. If the joystick is engaged, it will default to the central position, unless it is being touched, in which case it will stay where it is. Similarly, if a path is assigned where Controls Active is off, the joystick (if engaged) will default to the central position, unless it is being touched, in which case it will stay where it is.

FREEZE	:	that axis. Free controls, they	s pressed on eith eze does not affe can still alter the n the joystick mu	ect the Input/0 frozen axis.	Output par The null I	nel or Wild _EDs show
DISENGAGE	:	does not con against accide direction in wh When Disenga position of the	ne joystick only. atrol or move to for ental changes. The nich the joystick r age is de-selecte audio, unless it re to the position	ollow the audi ne null LEDs nust be move d, the joystic is being toucl	io. This is will still inc ed to matc k will mov hed, in wh	to protect dicate the th the audio. ye to the
SNAP TO AUDIO	:	Pressing this the audio.	outton will cause	the joystick to	o snap to	the position of







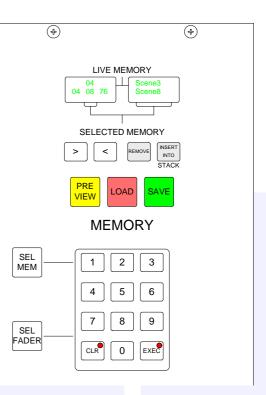
### MEMORY CONTROLS

99 memories can be held in the Flash ROM for different console arrangements. In addition to this, the PC back-up can allow an unlimited number of memories, which can be called into the Flash ROM quickly and easily. Memories can be stored to removable media. This can be useful for when many different operators use the same console (for example an Outside Broadcast vehicle), or when the console is used to broadcast many different weekly productions.

#### Live and Selected Memories

The display at the top of the panel shows the "Live Memory" on the top half, and the "Selected Memory" on the bottom half.

The Live Memory shows the last memory loaded onto the console. Changes made since this memory was loaded will not be stored in this memory number unless it is re-saved. They will be stored in the "Hidden" memory so that they are restored after a power down.



The buttons on this panel will affect the Selected Memory. The Selected Memory can be thought of as the "Ready" position, where the operator can place the next required memory until it is needed. Pressing LOAD will launch the Selected Memory into the Live Memory position, overriding the previous console settings.

#### **Choosing the Selected Memory**

With SEL MEM lit, enter the two digit memory number followed by EXEC on the keypad to call that memory number into the Selected Memory position. The Selected Memory can also be chosen using the memories screen which accompanies this panel. Selecting the required memory in the Flash ROM list on the left of the memories screen will call it into the Selected Memory position.

#### **Clearing the Selected Memory**

The contents of the Selected Memory can be cleared by pressing SEL MEM + CLR + EXEC on the keypad or selecting CLR MEM on the screen.

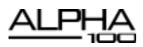
#### **Saving Memories**

The Save button will save console settings to the Selected Memory. Therefore, the memory to which you want to save must be in the Selected Memory position when Save is pressed. Alternatively, SAVE+Memory Number + EXEC will save into that memory number.

To create a new memory, choose an empty memory by pressing SEL MEM and typing it's number on the keypad, or by selecting it from the list on the left of the Memory screen. If however, you wish to simply update changes you have made to the Live Memory, it must be showing as both the Live Memory and the Selected Memory in the display. The PC can be used to change the title of the memory being saved.

#### **Preview Memory**

When the Preview button is pressed, the Selected Memory settings will be displayed on the control surface. The Assign panel displays will be blanked out. Upon release of the Preview button, the control surface will display the live settings again.





### **Stacked Memories**

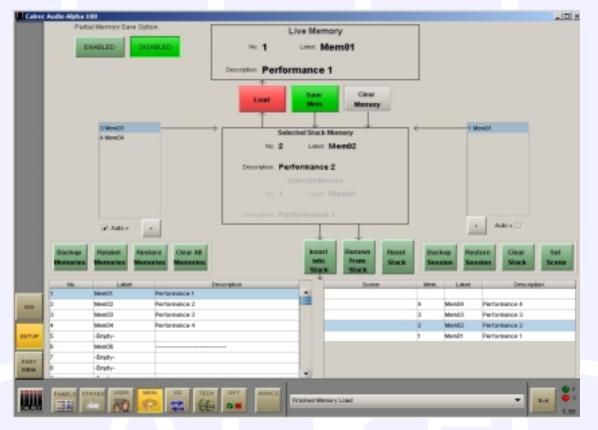
The memories can be arranged into a Pre-set list, known as a stack. This can be useful for setting up an easy-to-access shortlist of specific memories for use during a show. Stacks can be saved to the hard disk or removable media as sessions.

To allow the stack to use the Selected Memory position, any memory which has been selected manually, and is not part of the stack (shown in inverse text), must first be removed from the Selected Memory position, by pressing REMOVE. If REMOVE is pressed while a stack memory occupies the Selected Memory position, it will be removed from the stack. A second press will remove it from the Selected Memory position.

The > and < buttons scroll through the stack. Pressing both > and < together, will reset the position so that the last number loaded is back in the central position. To add a memory to the stack, ensure it is in the Selected Memory position, and press INSERT INTO STACK. Pressing REMOVE will remove a stacked memory from the stack, or will remove a non-stack memory from the Selected Memory position. Inverse text in the display indicates that the memory is not part of the stack.

#### **Memory Screen**

The Memory panel is accompanied by a screen which duplicates the memory functions available on this panel. It also allows management of stored memories and stacks (sessions). Memories and sessions can be backed up and recalled using the Memory screen. When a stored memory is recalled onto the console from disk, the system checks that the current desk configuration matches that of the stored memory. If there are discrepancies, a warning that the memory may not work correctly will be given.



#### **Memory Isolation**

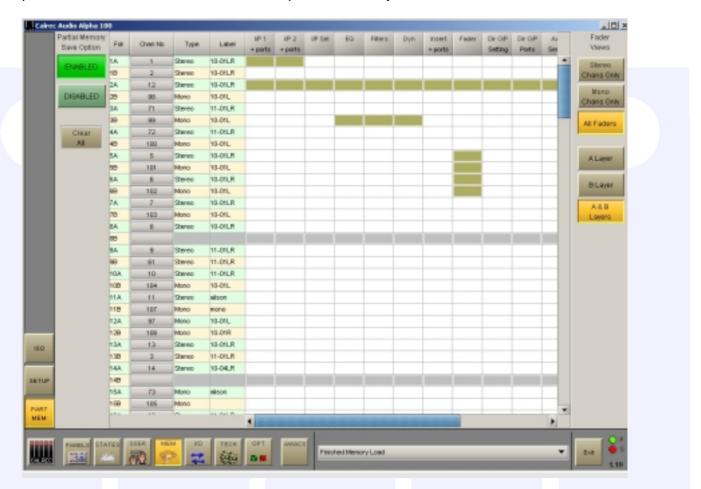
Some console settings can be isolated from memory recall, so that they will not be over-written when a memory is loaded onto the console. This is done using the Isolate screen.





### PARTIAL MEMORIES

The Partial Memories function allows components of console settings to be saved in the same way as full console memories. When a partial memory is recalled, only the settings saved will be updated. The partial memories screen provides a mechanism for selecting channels or sub-components of channels to be saved in a partial memory.



Partial Memories mode is enabled and disabled using a button on the partial memories screen. When enabled, all memory saves are partial memory saves. When disabled, all saves are full console snapshot saves. Once partial memory mode is active, the save buttons on the screen and control surface are used to save partial memories in the same way as full console snapshot memories.

The partial memory screen contains a table with rows of channel numbers and columns of partial memory components. The sequence of the channels is in fader number order. Partial memory component selections are made by selecting the intersection of a channel and a console setting. The Channel Number field in the fader table is a selectable button, which selects or deselects ALL partial memory components for the channel occupying that fader.

Partial memory selections are stored and are recalled if a console reset occurs.

A partial memory only loads those channels or components on the control surface that were selected using the partial memory selections. A partial memory load does not affect the currently assigned fader and the A/B layer assignments.





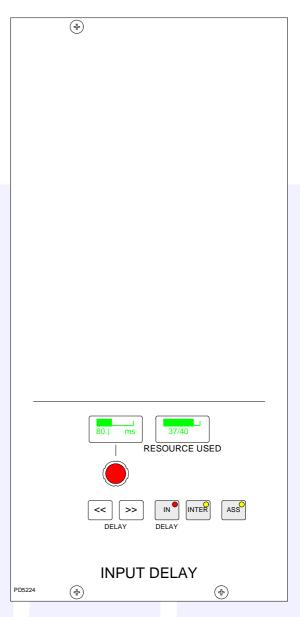
## **INPUT DELAY PANEL (OPTIONAL)**

The Input Delay panel allows the user to apply specific amounts of delay to each channel path.

There are 42 legs of delay available for channel assignment. Stereo channels use two legs. Each leg provides up to 250 ms of delay, adjustable in 0.1ms steps using the rotary control, and 10ms steps using the nudge buttons.

The RESOURCE USED display shows how many legs are already assigned.

- ASS : To assign delay to a channel, select the fader path by pressing it's assign button (A or B), and then press ASS on the Delay panel. Once delay is assigned to a channel, a delay value can be set using the rotary control or nudge buttons.
- IN : Switches the set value of delay in and out of the channel's path.
- INTER : Holding down the interrogate button will indicate the channels which have delay assigned by lighting their fader assign buttons.





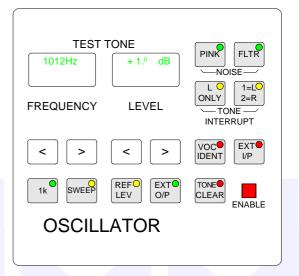


### **OSCILLATOR CONTROLS**

The Oscillator controls are located above the Fucnctions controls, and are used to generate test tones for alignment and testing.

The frequency of the tone can be adjusted from 20Hz to 20KHz in in incremental steps using the nudge buttons, or set to 1KHz using the 1K button. Alternatively, the sweep button will set the oscillator to sweep through all frequencies.

The level of the test tone can be adjusted from -60dBFS to 0dBFS using the nudge buttons, or set to the reference level using the REF LEV button.



The Tone Interrupt buttons are useful for testing stereo paths. They allow the tone to be interrupted on the left side only, or on the left and right sides in an alternating pattern.

EXT I/P when pressed replaces the tone with a mono or stereo external source of your choice. This allows for external oscillators to be used if preferred. The ports for this are set up on the OPT-MON I/P & TB screens.

TONE CLEAR clears all oscillator routes made, providing an easy way of removing test tones from signal paths.

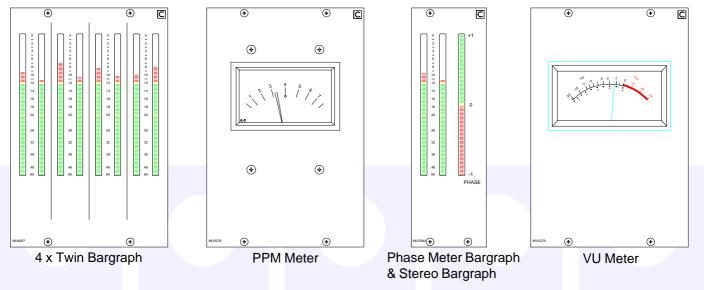
The Enable LED is lit to show that the Oscillator controls are enabled.

Oscillator controls are accompanied by a supporting screen.





#### **METERING OPTIONS**



The MAIN and ANCILLARY 1 meters can be stereo only, surround only, or surround plus stereo (displaying a downmix of the surround signal). There can be a separate M/S meter (fed from the same downmix). They can be PPM's, VU's, bargraphs, phase display incorporating bargraphs, or a mixture of these.

The Main meters are fed from the Main meter selector which is on the Monitor Selector panel. It can select either Main 1 or 2 Desk (pre Tone & TB), Main 1 or 2 Line (which can be an external input), or Tone.

In addition to the Main and Ancillary 1 meters, a comprehensive set of optional meters are available:

- Track Bargraphs displaying the track output levels, post Tone &TB.
- ANCILLARY 2 Meter: This is stereo only. It can be PPM's, VU's or bargraphs.
- Stereo APFL or surround AFL Bargraph. AFL is monitored post the channel/group panning and is in surround. The APFL meter will display the stereo downmix of these signals.
- MIX MINUS: Displays signal on the mix minus bus (mono).
- GROUPS: 8 stereo bargraphs for the groups. For mono groups, the meter will display the left bar only.

Calrec can supply either bargraphs, Moving Coil VU or PPM meters. (except for the Aux output meters, which are always bargraphs). All Calrec meters including moving coil types, are fed directly from the meter processor.

Calrec bargraphs provide a bar which can be either VU or PPM. In addition, there can be a True Peak spot (which incorporates a long release time). Together, these allow the operator to see the level of the signal using a familiar meter and at the same time to see how close the peaks of the signal are to the digital maximum.

The bargraphs can have yellow markers at specified points (to mark the "nominal" and "peak" levels). The top of the bargraph always equals full scale digital level. The scale on the bargraph is normally 0 (at the top) to -60 in dB. Other scales can be provided to special order.

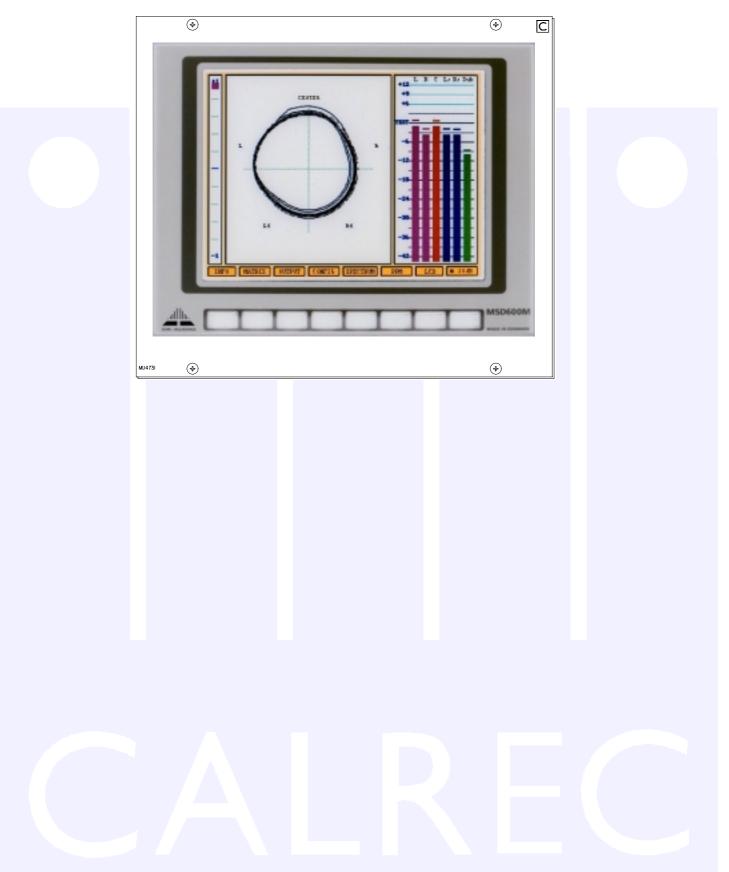
The meter bridge is continental height allowing alternative European bargraph meters to be fitted. These would require audio outputs from the I/O Rack.





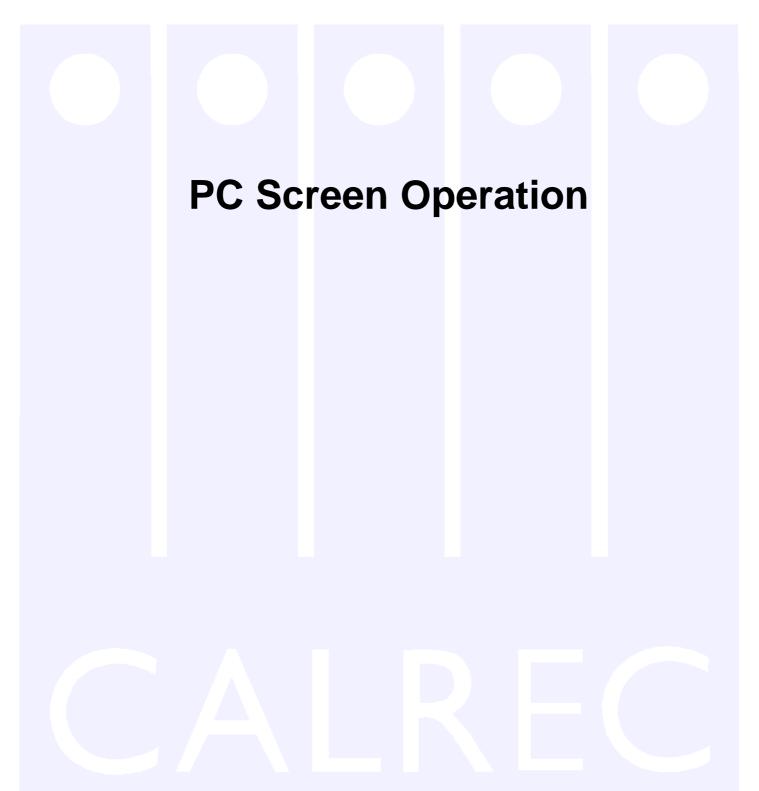
## **OPTIONAL THIRD PARTY METERING**

It is possible to incorporate third party metering options into the Alpha 100 design, such as the DK Audio MSD600M shown here. This would require audio outputs from the I/O Rack.













## PC SCREEN USAGE & LAYOUT

The Alpha 100 is designed to minimise the need for the operator to use the screen once the console has been preset. A logical user interface provides easy and quick access to the functions and information on the touch screen. Failure of the screen's computer has no effect on the operation of the control surface or the audio.

The Alpha 100 screens are divided into groups which are accessed using the buttons along the bottom of the display. There are groups for:

PANELS	PANELS	Operational reproductions of the console panels for off-line work or in case of a panel failure.
STATES	STATES	Sets the current state of various functions (these are not stored with the memories or options - only in the live (hidden) memory.)
USER	USER	Operational screens which enhance the controls on the console and for setting options which are stored with the memories.
MEM	MEM	Memory control screens to supplement the panel controls.
1/0	I/O	Set up and display of all the I/O connections stored with the memories.
TECH	TECH	Entry to and control of password-protected operational modes, trouble- shooting screens.
OPT	OPT	Options screens for pre-set items which are not stored with the memories.
	NET	Screens for setup and control of an audio network system (Only visible if the Hydra audio networking system is installed).

Within each group there are a number of screens accessed by buttons up the left side of the display. On some screens, there are drop boxes or additional buttons to access sub-sets of the screen's function.



The "EXIT" button at the bottom corner of the screen will exit the application. Next to this button are two indicators which show the status of the primary and secondary control processors. During normal operation, the primary processor will be in use, and its indicator will be green. When busy, the processor's indicator

will be amber, during which time, no changes can be made to the control screens, (Although changes to the console's control surface can be made, and will take immediate effect).





•	Audie Alpha 100	i Pendir	0				كلم
ERROR	WARN	NFO Only Roled	Count	- Grant		6 matrix	
Read	CH. 19.A	40-0413:58:57	Gewed	Source Control Surface	UN4805 Started	Sumery	
12	00 104	ag of the best		Presenter derrichde	President of Ballon		
				Message D	escription		
inestage de	scription						
message de	scription						

If a problem does develop, messages will be delivered on the Automatic Warning and Correction System (AWACS) screen. The AWACS button at the bottom of the screen will flash to alert the user that a message has been reported. Selection of this button will open the AWACS page, where messages can be viewed. Selecting a message will reveal a more detailed description. Message history is saved to the PC's hard disk for future analysis.

Three types of messages are reported:

Information messages, eg "Control Surface UN4806 processor started successfully"

- Warning messages, where the system back-up has taken over
- Fatal Error messages, where the system cannot recover by itself (perhaps because the back-up is already in use)

Because the system has many back-up features, it is possible to continue operating after errors are reported. If un-cleared errors are still present, an icon will flash in the AWACS button. Selecting this button at any time will switch back to the AWACS screen. Information messages can be cleared by selecting them and then leaving the AWACS screen. Warning and Fatal Error messages can only be cleared by clearing the error and restoring the system to its normal operational state.





### **INPUT/OUTPUT PORTS SCREENS**



This screen allows:

- Patching" of input sources to channel inputs, insert returns, direct inputs or to output ports.
- "Patching" of console output signals to main, auxiliary and track output ports, insert sends and direct outputs.
- "Patching" of insert sends and returns to channels and groups.



On the control surface, ports are assigned using the I/O Matrix panel, but these screens provide an alternative set of controls. The screens automatically scroll to follow the Assign button (A and B) presses on the faders.

#### (1) Input, Output or Insert

These buttons select between three screens for input patching, output patching or insert assignment.

### (2) Source Lists

All of the available input ports can be grouped into suitable lists at the time of installation. These lists can then be displayed on the left of this screen, ready to be patched to channels on the right. Different lists are accessed using the selection buttons.

### (2) Viewing Options

The sources can be viewed as pairs (best for patching to stereo or surround paths), individual (best for patching to mono paths), or individual with the actual rack number, card slot and input shown (for diagnostic purposes).





### (3) Input Views

These buttons select the different console path types which can have input ports attached (channel inputs, insert returns, direct inputs or outputs). They will then be displayed in the main section of this screen. Selecting a source from the source list and a channel, insert return or output, then selecting PATCH will assign that source to the channel.

#### (4) Fader Views

It is possible to choose which set of faders are to be available on and altered by this screen.

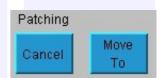
### (5) Patching

Assignment is made by selecting a source, *stand* an input or output, *stand* and selecting Patch.

The input source label will appear in the channel input label field and on the fader on the console (if that input, 1 or 2, is selected on the Input/Output panel). By selecting the label cell on the screen, the input name can be edited using the keyboard. The new name is stored with the channel input and replaces the source label on the fader display.

Once patches are made, they can be removed when selected by clicking REMOVE.

Connections can be moved between channel inputs when selected using the MOVE FROM button. The Input 1 or 2 field will be highlighted and the PATCH, REMOVE and MOVE FROM buttons will be replaced with MOVE TO, and CAN-CEL. Upon selection of a new patch point, pressing MOVE TO will move the connection. CANCEL will cancel the operation.



Multiple Patching - It is possible to patch regions of sources to a region of inputs.

				_					1.00.00	
Түрө	s an	s and the second se	MO		Mio	Pdr	Туре	Label	O-	Right
NL.	10-01 L	1 R							0	
ML.	10-02 L	10 R			Use .	140	588780			
ML.	10-00 L	10 R			Dia	15A				
ML.	1 10-04 L	10 R			Dig	158	Stereo			
ML	1 10-05 L	10 R				16A				
ML	1 10-06 L	10 R				158				
ML	1 10-07 L	10 R				17A	Stereo	1D-D1LR	🐴 10-01 L	∐rn –
ML	1 10-08 L	10 R				17B	Stereo	1D-02LR	10-02 L	∐-R
ML	1 10-00 L	ÚR.				10A	Stereo	10-00LR	10-03 L	₫-R
ML	1 10-10 L	() R				108	Stereo	10-04LR	10.04 L	∐-R
ML	1 10-11 L	d.e				19A	Stereo	10-05LR	10-05 L	(TR
ML	10-10-11 L					198	Stereo	10-06LR	10-09 L	(TR
		<u>0</u> π				20A	Stereo	10-07LR	110-07 L	(TR
ML.	10-10-13 L	<u>0</u> π				208	Stereo	10-06LR	110-08 L	(1 R
ML.	10-14 L	<u>O</u> R	**			21A	Stereo			<u> </u>
ML.	10-15 L	1 C R				- 115			-	_

- Select a list of input ports using the trackball by dragging down the column
  - Select the fader to start patching to
- Select Patch

### (6) Port Isolation

The ISOLATE button allows the selected port connection to be isolated from memory recall, so that its current settings will not be over-written by what is in the memory. Clicking the button a second time will de-isolate the connection. A brown cell in the Label column indicates that a port has been isolated. Other console settings can be isolated using the ISOLATE screen.

### (7) Locking (only available on output ports)

Some output ports may need to be 'locked' once they have been set up to avoid accidental removal. For this reason, Calrec provides a system of software locks to protect critical parts of each configuration. The Alpha 100 can be in one of three modes, "User", "Technician" and "Supervisor". Operation of the locking system is only available in "Technician" or "Supervisor" mode, which are password protected to add an extra layer of security. Modes are selected using the TECH screen. If a lock is active, the port name will be highlighted in bright green text.





## **CHANNEL FUNCTIONS SCREEN**



USER

This screen provides alternative controls for channel functions already available on the control surface.

- (1) The right side of the screen shows the fader paths A and B. To make changes, click on the required fader path and use the controls on the left side of the screen.
- (2) The path type can be selected as mono or stereo on this screen in addition to the I/O Matrix panel controls.
- The path can be selected as a group using the buttons 1-8 (3)
- (4) Paths can be moved and cleared using this screen, these controls are in addition to the I/O Matrix panels controls.



## (5) Assigning Wild Controls from the USER-CHAN Screen

The Wild controls are assigned from either the Functions panel, or from the USER-CHAN screen. All the Assign panel rotary controls incorporate a switch which is operated by pushing the control. These switches are used to assign the control to a Wild control as follows:

- Select a Fader Path by pressing its Assign Button (A or B).
- Select WILD ASSIGN 1, 2, 3 or 4 on the USER-CHAN screen.
- Push one Assign panel rotary control. For example, Aux 1 Send.

The control is now assigned and changes will show in the display. The colour of the Wild control display will show which fader the control is related to: Green for A, Amber for B.

It is possible to assign controls to more than one fader path at a time, either by selecting individual fader assign buttons (A or B), or by defining a "block" or "Region" of faders. Clicking on the button above HOLD will toggle between SELECT mode and REGIONS mode.

In SELECT mode, click HOLD, then a number of fader paths can be selected individually by pressing their fader assign buttons (A or B). Pushing an Assign panel rotary control will assign that control to all selected faders.

In REGIONS mode, a block or region of faders can be defined by clicking HOLD and then pressing the fader assign buttons of the first and last fader path in the required region. Pushing an Assign panel rotary control will assign that control to all fader paths in the selected region.

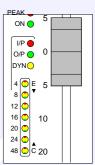
It is possible to assign the same control to Wilds 1, 2, 3 or 4 for all fader paths by selecting ALL before pushing the required Assign panel rotary control.

The gains of the two inputs 1 and 2 can be assigned seperately to Wild controls, by holding down the required input button on the Input/Output panel before pushing the gain adjustment rotary control.

CLR will clear the selected Wild control from it's assignment.

#### (6) Fader Bargraph Assignment

The fader bargraph can indicate the level at the channel input (post the input gain & switching and the tone switching), the channel direct output, or the gain reduction of the dynamics. Buttons I/P, DIR O/P, DYN and OFF on the USER-CHAN screen will set the function of the fader bargraph on the currently assigned fader. If ALL is pressed first (flashes) all fader bargraphs will be set to the selected functions.



Fader Bargraph assignment can also be altered using the Functions panel





Celrec Audio Alpha 100						- 19
Available Sourceo		Synchronise	on Sources			
Source	Rack - Slot - Input	Carrent	Order	Connection	R-S-I Sar	quie Rate
Video NTSC	Rear at Dig. NO	333	1#	Internal	-43	
TTL Workblock	Rew of Dig. NO	_	2nd	Internal	-43	
Internel		_	3rd	Internal	-45	
11-01 LR	1 - 11 - 1 LR	_	-Rh	Internal	-43	
11-02 LR	1 - 11 - 2 LR		5th	Internal	-43	
11-03 LR	1 - 11 - 3 LR		@h	Internal	45	
11-04 LR	1 - 11 - 4 LR					
11-05 LR	1 - 11 - 5 LR					
11-06 LR	1 - 11 - 6 LR					
esc						
			Patr	h To Reset to 1 s	z	
5196 CR IP + 79			Patr	h To Reset to 1s SAVE options to Disk and Flash	st LOAD options from Disk to Flash	
MISC SINC ACHIP + TB SAMEN GPI OPO PAL			Patr	SAVE options	LOAD options	

OPT

The system can be pre-set with up to five external sync sources, plus internal, such that if the 1st source fails, it will automatically switch to the 2nd, and so on. One of the external sources can be Video (PAL or NTSC), and TTL wordclock is another possible external source.

Digital inputs on the console can also be used as an external source. Please note that the facility for locking to external AES sources is restricted to the first six inputs of each AES card in the console. These can be patched to the five selections using the Patch button. When using a digital input or wordclock as a source, the system will tolerate a variation of up to +/- 100 Hz in the frequency of the source.

If the system is running on any of the selections 2 to 6, because the lower numbered ones have failed, and the 1st source is repaired, the system can be RESET TO 1ST during any convenient off-air period.

The Options screens are used to pre-set the system to the studio's required settings. These settings are not stored in the individual console memories but are saved separately using the LOAD and SAVE buttons. This allows options to be changed without invalidating any saved memories.





MONIP

TalkBack

### MONITOR I/P & TB SCREEN - TALKBACK INPUT PORTS VIEW

		HL D			TB RTD1 RTD2 RTD3 RTD4 Ed Tons IP L Ed Tons IP R Ed Tons IP M	ON OTT OTT OTT OTT OTT OTT	OFF OFF OFF OFF	120d5 00d5 00d5 00d5 00d5 00d5 00d5 00d5	Selecto (Mon Se Inputs Mon Se (Ed MP)
		HL D			RTB2 RTB3 RTB4 Ext Tone IP L Ext Tone IP R	017 017 017 017	OFT OFT OFT OFT	0.045 0.045 0.045 0.045 0.045	(Mon Se TalkBar Inputs Mon Se
					RTBS RTB4 Ext Tone IP L Ext Tone IP R	an an an	097 097 097 097	0.0 dB 0.0 dB 0.0 dB 0.0 dB	TakBa Inputs Non Se
					RTEH Ext Tone IP L Ext Tone IP R	orr orr	OFF OFF OFF	0.045 0.045 0.045	Inputs Mon Se
					Ect Tone IP L Ect Tone IP R	017 017	OFF OFF	0.045 0.045	Mon Se
					Ext Tone IP R	orr	OFF	CED GED	
						-			
5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5					Ecd Tone IP M	017	OFF	0.048	(Est NP
5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5									
5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5 5									
R R R R R R R									
5 5 5 7 7 7 7 7									
5 5 5 5 5									
R R R									
R R									
n n									
<b>R</b>									
<b>R</b>									
<b>R</b>									
<b>R</b>									
<b>R</b>									
<b>R</b>				Mic	UP PH D)	UP SRC /	Inalogue Gain		
<b>R</b>									
<b>R</b>				_			_		
8	-			Detables		SAVE and		OAD antimax	
Views			-	- atomy					
Mono	Diagno	ostes	Patch	Remove	Fram	_			
	R R R R Viewo	N N N Viewo	R R R Views	R R R Viewo	R R Views Views	R R R Viewo Viewo Disgnostics	R R R Viewo Viewo Nono Diagnostics Patch Ramova Move Fram	R R R Viewo Viewo Diagnostics Viewo I I I I I I I I I I I I I I I I I I I	N     Mr. UP PH     Dig UP SRC     Analogue Gain       N     Nono     Patching     SAVE options     LOAD options       Nono     Diagnostics     Patch     Remove     Move

The input sources for Talkback and Reverse Talkback can be patched here in the same way that channel inputs are patched. Talkback input ports can be any kind of port.

The parameter buttons provide controls for analogue gain control (coarse), Phantom Power (if mic/line) and SRC switching for the input (if digital). When selecting analogue gain, a box will appear where the gain can be selected. Selecting Mic i/p PH will turn phantom power on for the selected input. Selecting Dig i/p SRC will switch SRC on for the selected input.

Select gain	<u>×</u>
1 49	6.09
12.69	18.08
24.69	30.08
36.69	42.08
48.69	54 ctb
68.00	96 c/D
72.60	70 dD
Can	Let







	udio Alpha 100									
	ype Function	Ca	nd Relay	List	-	and Circuit	Тури	Function Applied APPL YOY	Latch	1
	On eir LED	1	-	-	_	5	Relay	APTL OF	Lation	
	Reh. LED	1	2		-	6	Ralay			-1
	PSU tel LED	1	3			7	Relay			-17
	APFL 'Orf	1	4				Relay			-17
	On air (Tx)		_			9	Relay			-17
	Reh.		_			10	Relay			
	Red light				1	11	Relay			
	Fire olorm mute				1	12	Relay			
	PSU teil elevre				1	13	Relay			
	AFL YON'				1	14	Relay			-17
	PPL YOM				1	15	Relay			
	Mic. open 1 YOM				1	16	Relay			
	Mic. open 2 'ON'				2	1	Relay	DP570 surround	P On	
	DPS70 surround	2			2	2	Relay	DPS70 steres	P On	
	DPS70 stereo	2	2		2	3	Relay	DPS70 mano	P On	
	DF570 mona	2	3		2	4	Relay	DPS70 phantom centre	P On	
MISC	DPS70 phenton centre	2	4		2	6	Relay	DP570 3-stereo	P On	
_	DPS70 3-stereo	2	5		2	6	Relay	DPST0 Prologic	P On	
	DP570 Prologic	2	8		2	7	Relay	DPSTD line	P On	
SYNC	DP570 oustom	2	8		2	a	Relay	DPST0 custom	P On	
	DP570 line	2	7	-	2	9	Relay	DPSTO RF	P On	
NON IP	DES20.8F			-	2	10	Relay			
+ TD		Views			2	11	Relay			
	Misc.		Chi	an féir	1	15	Market 1		_	1
NUMBER OF	funce		0.94			Latch		Pulse Pulse	Pulse	
		(1)						on of .	Both	
		U								
0.91						F	atching		AD options	
_					(2)	) Patch F	emove	Move He Disk and Flash Bro	m Disk to Resh	
070					2		Certifie .	From		

OPT

- (1) The Relay-isolated outputs can have various console functions assigned (with "Misc Functions" selected), or they can be set to operate when particular faders are opened (with "Channel Fader Open" selected). Console Functions can be assigned to more than one relay.
- (2) To make an assignment, select a function (left side of screen), and a Relay-isolated output (right side of screen), and select Patch. Assignment can also be moved and removed, in a similar way to port connections.
- (3) The relay can be set to latch or pulse for 100 ms, when the Console Function is activated. When setting the relay to pulse, there are three different options.
  - Pulse On
    Pulse Off
    Pulse Both
    The relay is set to pulse when the function is activated.
    The relay is set to pulse when the function is de-activated.
    The relay is set to pulse once when the function is activated, and again when the function is de-activated.





_	Card C	Dato Sig. Req.	Console Function Connected						- 10
,	1	dan ndir unit		Type Puncti	an Cand	Opto	Sig. Req.		Lists
,	2			SLS 1 OU			Letch		1000
1	3			SLS 2 cut			Leth		-
1	4			Channel cut buts			Latch		
1	5			Channel cut buss	2		Latch		
,	6			DP570 powered DP570 Protogic	2	6	Latch		
,	7			DPS70 surround	2	-	Latch		
1	8						Letch		
2	1	Letch	DP570 stereo	DPS70 stores DPS70 mono	2	-	Latch		
2	2	Letch	DP570 mano			-	Latch		
2	3	Letch	DP570 phantom centre	DPS70 phentam c DPS70 3 stereo	2	4	Latch		
2	4	Letch	DP578 3-stereo	DPS70 custors	2	-	Latch		
2	6	Letch	CP570 Protogia	DPS70 line	2	· · · ·	Letch		
2	6	Letch	CP570 Inc	D#570 PF	2	-	Letch		
2	7	Letch	DP570 custom	Dec.rem.2 powe		-	Latch		
2	в	Ladoh	DP570 HF	SDU-4 powered			Latch		
				Main 1 TB			Latch		
sc				Main 2 TB			Latch		
				Main 3 TB			Leich		
				Main 4 TB			Letch		
NC				Ed. TB 1			Latch		
				Ed. TB 2			Latch		
1IP				Ext. TB 3			Latch		
					1 Views			٣	
REFH				Miss.	Channe	el	1	AU	
				tuncs	cut			fad	te
			2 Patching	SAVE O			ptions sk to Rasl		
0			Patch Remove From						

- (1) The Opto-isolated inputs can be assigned to various console functions (with "Misc Functions" selected), or they can be set to cut channels (with 'Channel Cut' selected). With "Auto Fade" selected, the opto-isolated inputs can be assigned to auto-faders to allow automatic cross-fading (see page 62).
- (2) To make an assignment, select an opto-isolated input (left side of screen), and a function or channel (right side of screen), and select Patch. Assignment can also be moved and removed, in a similar way to port connections.

If optos are patched to input ports, when fired externally, they will cut any channel to which that input port is connected.





### **AUTOMATIC CROSS-FADING**

This feature allows the user to automatically fade channel and group faders in or out under the control of an external signal. Cross-fades may be achieved by driving two or more optos with one fader being faded out while a second fader is being faded in.

#### **Assignable Auto-Faders**

32 assignable "auto-faders" are provided. Each auto-fader provides the ability for one path to be faded up to and down from the current fader level.

Each auto-fader can be assigned to any one opto input. An auto-fader without an assigned opto will remain inactive, it's operation will have no effect on the audio but it may still be assigned to a path. The association of the auto-faders to opto inputs is stored in the Options file.

Any single channel/group path may be assigned to an auto-fader. The association of channel/ group paths to auto-faders will be stored in the console memories.

#### Operation

Once an autofader has been assigned to an opto input, and has a channel or group path assigned, it is possible to automatically fade in or out the assigned channel or group fader under the control of the assigned opto input. When the opto input is fired, the path connected to the opto will be automatically faded in to the current fader level (after taking into account any VCA fader adjustment). When the opto is not fired, the fader connected to the opto will be automatically faded out.

#### **Options Screen**

Optos are assigned to auto-faders using the Options-GPI screen below. To assign an opto to an auto-fader, select an Opto from the available list on the left, and an auto-fader from the auto-fader list on the right, and then select "Patch".

Auto

fade

Each autofader has a 6 character user editable label.

rex Audio Alpha	199			9 C 3	1005.	20 <b>2</b> 2	<u> 1</u> 2		1 00 <u>- 0</u> ×
Card	Casto	Sig. Res.	Canada Function Connected	D	1 1	shol	Card	Ceto	Lists
<u> </u>	1								- Inc.
<u></u>	2			2					1.00
2	3			0					
1	4	Latch	ALITO FADER 15	4					
1	6			5					
	6	Latch	AUTO FADER 6	8	PI		1	6	
1	7			7					
<u> </u>	0			0					
2	1	Latch	ALEO FADER 9		P2		2	1	
2	2			10					
2	3			11					
2	4			12					
Ě				12					
	8			14					
2	0			15	P3		1	4	
2	0			16	_			-	_
				17					
				18			-		
				19					
				20			_		
				21					
P				22	-		_		-
				23		Views			-
-									
					Misc.	Channel	1		Auto
					tunca	put .			føde
-									
			Patching		SAVE option	a 16	ALD opt	ion a	
_			The second second second		to Disk and	Resh In	om Dink	te Rash	
			Path Remove From						
_				_	1		_	_	_
	TATES	HEM UD	TECH OPT ANALS						
	100							•	5.0 0 S
a 114	0.7		🦚 ni 🛛 🚽 🚽						1.12





-Fade	Sc		USEF		AUT	-								
Cole	ec Aude	Alpha 199									12 21.	90 EI 49 (	- I I I I I I I I I I I I I I I I I I I	3 <b>8 8 9 .</b>
	D	Auto Fade Lab	el Carr	Opte	P#	TT.	For	Type	Label	Auto Fede	Pade in (m5)	Pade Out (m5)	Fede	Fador
	1					-	1.4.	Sieres					<ul> <li>MOvi</li> </ul>	Views
	2			_			10	Mono						29100
	2			-			1A.	Stereo					· · · ·	Chans Only
	4					1	20	Mono						-
	1						3.8.	Steres	0491		5000	900		Mono Chens Only
	s -	PI	1	6	34		30	Mono						COMPACE AND
	7						4A,	Stereo						All Faders
	0						-0	Mono						
	۰.	P2	2	1	64		6A.	Stereo	can2		5000	9000		
	10						55	Mono			1.000			A Layer
	91						SA.	Stereo						
	12						80	Mono						G Layer
	13						2A.	Stereo						
	14						78	Mono			_			84A
	15	R0	1	4			RA.	Stereo		1				Layers
	16						10	Mono	cand					
	17						54.	Stereo	_					
	18		_	-	-		10	Mono						
	12		_	_	_		AOF	580.00						
CHIN				-			100	Mono						
_	21		_	-	-		11A	Stereo						
	22		-	-	-	-	118	Mono			-			
60.9561			-	-	-		427	Sleves			-			
_	- 24		-	-	-		120	Mono				_		
090	25		-	-	-	-	104	Stereo				_		
	77		-	-	-	-	144	Steres						
	1		-	-				Sarres .	-		-	-	-	
COPY														
	-									Patching	_			
AUTO									Patch	Renow	Maye			
FADE											Fram			

The User-Auto Fade screen is used to allow assignment of each auto-fader to a channel/group fader. A list of auto-faders is on the left hand side of the screen, and shows auto-fader number, auto-fader label, assigned Opto card and circuit, and assigned fader number.

Only faders with valid channel or group paths will be available for assignment although other faders may be displayed.

### Fade In/Out Times

The nudge buttons allow fade in/out time adjustment for each auto-fader assignment. The fade in and out times of each auto-fader are individually selectable. The range for both parameters are 10 ms to 5secs, as follows:

- 10ms to 100ms in 10ms steps
- 100ms to 1sec in 100ms steps
- 1sec to 5s in 500ms steps

The fade in and fade out times are stored in the console memories.

Auto-faders are assigned to channel and group paths by selecting an auto-fader and a channel or group from the available lists, and selecting "Patch".

Indication of an auto-fade is provided by illuminating the down NULL LED on the fader strip when the fader is or is currently being faded out.





. . . . . . . . . . .

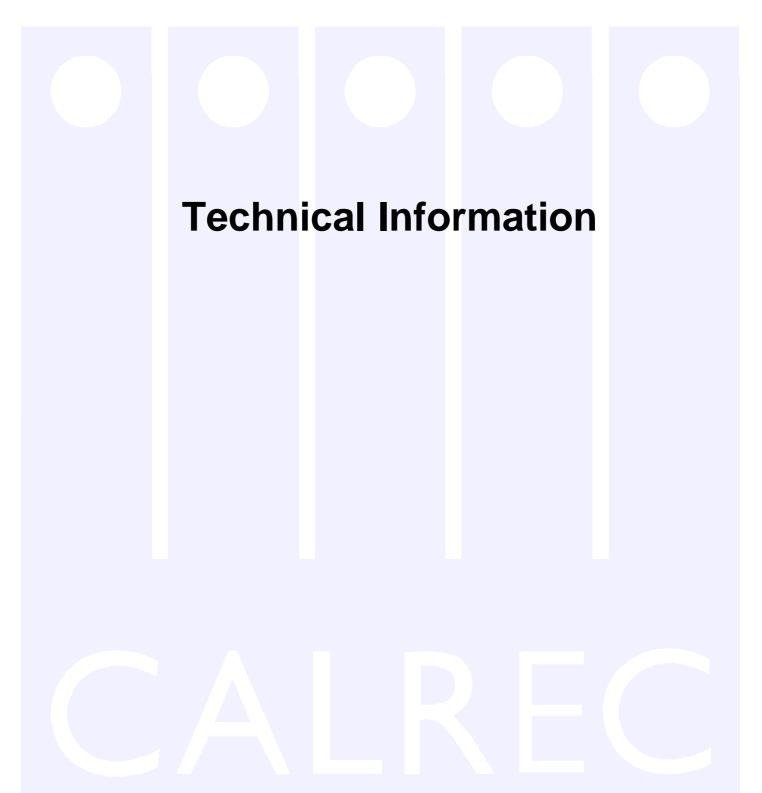
.....









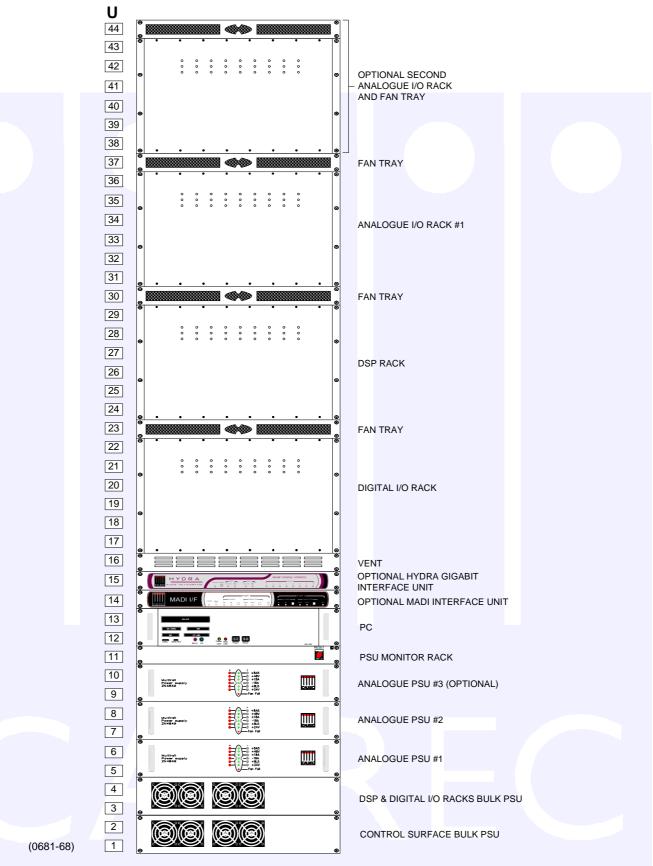






## **TYPICAL RACK LAYOUT**

Equipment can be mounted in separate enclosures. Please refer to the cable lengths table before planning this. The PSU monitor rack can be mounted on the rear of the equipment bay if pre-ferred.







## **RACK SPECIFICATIONS**

It is recommended that all equipment over 8Kg (17.5 lbs) in weight, or over 150mm (6 inches) deep is mounted into equipment bays which offer mechanical supports under each of the units. This will allow units to be supported as they slide forward during removal for maintenance purposes.



Equipment can be mounted in separate enclosures. Please refer to the cable lengths table before planning this. The PSU monitor rack can be mounted on the rear of the equipment bay if desired.

Each audio rack (DSP, Digital I/O, and Analogue) is supplied with a 1U low noise fan tray which should be positioned immediately above the rack. The fan tray incorporates a baffle such that warm air is sucked up out of the rack and out through the rear of the fan tray. A vent in the front of the fan tray allows ambient air to enter. The baffle deflects this air up into the rack above. The bottom rack should have a 1U vent beneath it to allow ambient air to enter. It should also not be positioned above any equipment producing significant heat.

Items	Height	Approx depth (incl. mating cons)		Approx weight		Approx Power Output (W)	Approx AC Power (VA) (full load)
		inches	mm	lbs	kgs	(full load)	()
DSP Rack (with cards for an average size system)	6U	18.9	480	26.0	11.8	-	-
DSP Rack (with cards for the largest system)	"	"	II	38.4	17.4	-	-
Digital I/O Rack (with cards for an average system)	6U	18.1	460	29.8	13.5	-	-
Digital I/O Rack (with cards for the largest system)	"	"	II	42.6	19.3	-	-
Analogue I/O (1/2 full)	6U	18.1	460	26.0	11.8	-	-
Analogue I/O (full)	"	"	u	35.7	16.2	-	-
Bulk PSU rack with one PSU*	2U	18.5	470	17.4	7.9	1000	1250
Extra PSU for Bulk rack	-	-	-	7.3	3.3	1000	1250
Analogue racks PSU*	2U	18.1	460	22.1	10.0	440	660
Power for Hot spare (any type)	-	-	-	-	-	No extra	Less than 5% extra
Fan Tray	1U	19.7	500	6.6	3.0	-	-
PSU Monitor box	1U	6.7	170	4.4	2.0	-	-
PC*	2U	23.7	600	27	12.2	-	360
MADI Unit	1U	11.9	300	6.6	3	-	-
Hydra Gigabit Interface Unit	1U	10.4	265	5.5	2.6	-	-

\* Note: Units have handles protruding approx 1.3" (32mm) from the surface of the front panel.

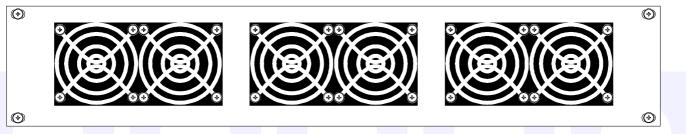




#### POWER SUPPLIES

All power supply units (PSUs) are rack mounting and are separate from the units they power, except for the PC which has its power supply built in. Diode feeding allows supplies of the same type to be parallelled together.

### Bulk Power Supply Rack (Powers Control Surface & DSP/Digital I/O Rack)



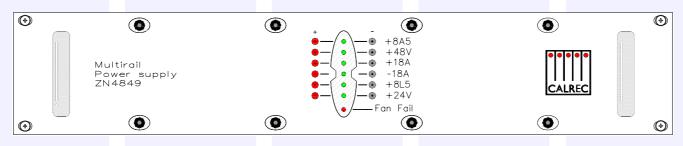
The Bulk PSU Rack is a 2U rack which can hold up to three identical 24V 1kW plug-in PSU's. The rack has separate AC power inputs and DC outputs for each of the three PSU's. Any one PSU can be removed from the rack without disturbing the operation of the others in the rack. The number of PSU's required in the rack is dependant upon the size of the system, the distance between console and rack, and the "hot spare" requirement. The DSP Rack and Digital I/O Rack are powered as one unit from one of these 2U racks. The Control Surface is separately powered from another of these 2U racks. The rack is fan cooled with fans mounted in the front of each PSU. The warm air is directed out of the rear of the rack.

### Fan Noise (dB SPL A-Weighted)

The following measurements were taken on axis at 1 metre from the dominant noise source:

Bulk PSU Rack				
1 x 24V 1kW PSU	49dBA			
2 x 24V 1kW PSU	52dBA			
3 x 24V 1kW PSU	54dBA			

## Multi-Rail PSU (Powers Analogue I/O Racks)



Analogue I/O Racks use a 2U Multi-Rail PSU. The number required will depend on the type of installation. Generally, one Multi-Rail PSU is required for one Analogue I/O Rack, and two for two or three fully populated Analogue I/O Racks. An additional Multi-Rail PSU can serve as the hot spare for several analogue I/O racks, provided they are housed together. If racks are housed in different locations, each may require a hot spare. This is dependent upon the cable lengths involved. All hot spares are optional. The Multi-Rail PSU is also fan cooled but uses a very low noise fan, drawing air from side to side through the PSU instead of in from the front, to further minimise noise. A noise measurement of 29dBA was taken on axis at 1 metre from the dominant noise source.

### **PSU Monitoring and Distribution Unit**

The Power Monitoring and Distribution unit monitors the power supplies for failures, and ensures "hot" changeover to the spare should a fault develop. The Reset button reboots the racks only, the control surface is unaffected.

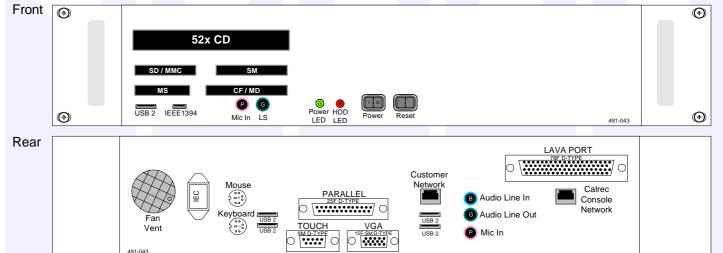




## **PC INFORMATION**

Failure of the console's PC does not prevent continued operation of the control surface or the audio.

Operating System	Windows 2000
CPU	Intel Celeron Processor (2GHz)
RAM	256 MB DDR RAM
HDD	40GB
CD ROM	52x
Network Ports	2 x 10/100
Card Slots	Compact Flash/Microdrive, SmartMedia, Memory Stick, Secure Digital/Multimedia Card
USB 2 Ports	4 (Rear of Unit), 1 (Front of Unit)
IEEE1394 Port	1 (Front of Unit)
Additional Hardware	Lava Octopus 8 Port Serial Card
Additional Software	PC Antywhere



#### **Remote Access**

USB connectors are provided on both the front and rear of the PC for the option to add an external modem of your choice. If a modem is added, and a suitable telephone line installed, the console can be remotely accessed by Calrec Support Engineers to aid software upgrades and diagnostic work. This can greatly enhance the level of service and support we can provide. A dial-up facility must first be activated at the PC before this is possible, to ensure that connections are not made at inappropriate times or without the user's knowledge and consent.

#### **Network Ports**

A network port is provided to enable the user to connect to their own LAN. Calrec will not be responsible for the configuration of this port or for any performance issues arising from its use. A second Ethernet port is provided to enable the PC to be connected to a Calrec Hydra Audio Network, which is an option which can either be purchased with the console or in the future.

#### **Software Supplied**

An OEM PC Operating System license is supplied with each console, and the operating system software is pre-installed. The console software is also pre-installed, and supplied on a CD-ROM.

#### 3<sup>rd</sup> Party Software

Calrec recommends that the PC is regarded as an integral control device for the console, and not as a general purpose PC. If 3<sup>rd</sup> party software is installed on the PC, care must always be taken to ensure that it does not interfere with the normal performance of the PC. The installation of inappropriate software on the PC may invalidate the console warranty.





### INTERFACE CONNECTOR PANELS

### **AES Interface Connector Panels**

AES inputs and outputs may be connected directly to the Alpha 100's Digital I/O rack using 36 way SCSI mating connectors. Optionally, break out connector panels and cabling can be provided. Ideally, interface panels should be fitted within 3m (9.8ft) of the backplane they connect to.

For digital inputs and outputs, interface panels can be either XLR (16 male or female, on a 1U panel) or BNC (32 on a 1U panel).

#### **Analogue Interface Connector Panels**

Analogue inputs and outputs may be connected directly to the Alpha 100's Analogue I/O rack using 36 way SCSI mating connectors. Optionally, break out connector panels and cabling can be provided. Ideally, interface panels should be fitted within 3m (9.8ft) of the backplane they connect to.

For analogue I/O, 8 x 38 way or 12 x 38 way EDAC connector 2U panels are available in the following styles:

- Mic/Line Inputs 4 pairs per EDAC
- Line Only Inputs
- 8 pairs per EDAC
- Line Outputs
- 8 pairs per EDAC





## MAXIMUM CABLE LENGTHS

Cables		Maximum Length	
From	То	Feet	Metres
Control surface	Control Surface Bulk PSUs	100.0	30.0
Control surface	PC	500.0	150.0
Control surface *	DSP & Digital I/O Racks *	100.0	30.0
PC	DSP & Digital I/O Racks	100.0	30.0
DSP Rack	Digital I/O Rack	1.3	0.4
DSP & Digital I/O Racks	Racks Bulk PSUs	100.0	30.0
Digital I/O Rack	Analogue I/O Racks	33.0	10.0
Digital I/O Rack	Digital I/O Interface Panels (BNC/XLR)	9.8	3
Analogue I/O Rack	Analogue I/O Interface Panels (EDAC)	9.8	3
Analogue I/O Rack	Multi-Rail PSU	33.0	10.0
Multi-Rail PSU	Other Multi-Rail PSU's	1.3	0.4
MADI Unit	Digital I/O rack	16.5	5
Hydra Unit	Digital I/O Rack	16.5	5

Power monitor rack cables are limited by other cable lengths.

\* Optionally, extenders can be supplied to provide console data connections up to 150 metres (500 feet).

### ENVIRONMENTAL CONSIDERATIONS

Temperature range: Operating  $0^{\circ}$ C to +30°C, in the immediate environment Non-operating -20°C to +60°C.

Relative humidity: Operating 25% to 80% non condensing. Non-operating 0% to 90% non condensing.

Altitude:

Operating: Up to 2,000 metres (6500ft). (This is the limit to which the safety tests are valid). Non-operating: Up to 15,000 metres (49,000ft).





#### SYSTEM SPECIFICATION

DIGITAL INPUTS			
Formats Supported	AES/EBU (AES3) 24-bit		
	Also suitable for use with SPDIF (IEC958 Type 2) signals		
Interface	110 Ohm transformer balanced, 5V Pk-Pk		
	75 Ohm unbalanced (BNC), 1V Pk-Pk		
Sample Rate Conversion	24-Bit switchable on all digital inputs		
SRC THD+N	-117dB @ 1kHz, 0.00014%		
DIGITAL OUTPUTS			
Formats Supported	AES/EBU (AES3) 24-bit		
Interface	110 Ohm transformer balanced 4V Pk-Pk (nominal) into 110 Ohm load 75 Ohm unbalanced 1V Pk-Pk (nominal) into 75 Ohm load (BNC)		

ANALOGUE INPUTS		
Analogue - Digital Conversion	24-Bit	
Input	Electronically Balanced	
Input Impedance	>1k Ohms for Mic gains 10k Ohms for Line gains	
Sensitivity	+18 / -78dB on Mic/Line Input Card +18/-24dB on Line Only Input Card.	
Equivalent Input Noise	-126dB (150 Ohm source)	
Distortion	-1dBFS @ 1kHz - Better than 0.003% -20dBFS @ 1kHz - Better than 0.006% -60dBFS @ 1kHz - Better than 0.3%	
Frequency Response	20Hz to 20kHz +/- 0.5dB on Mic/Line Input Card 20Hz to 20kHz +/- 0.25dB on Line Only Input Card	

ANALOGUE OUTPUTS	
Digital - Analogue Conversion	24-Bit
Output Balance	Electronically Balanced, 20Hz to 20kHz, Better than -35dB, typically -45dB
Output Impedance	<40 Ohms
Distortion	-1dBFS @ 1kHz - Better than 0.003%
	-20dBFS @ 1kHz - Better than 0.006%
	-60dBFS @ 1kHz - Better than 0.3%
Frequency Response	20Hz to 20kHz +/- 0.25dB

- Analogue input for 0dBFS can be pre-set globally to +28, +24, +22, +20, +18 or +15 dBu
- Pre-fader headroom on analogue inputs is adjustable globally from +24 to +36dB in 2dB steps
- Analogue output for 0dBFS Matches input setting into >1kOhms (+24dBu max into 600 Ohms)

PERFORMANCE			
Digital to Digital (AES/EBU) Distortion	-1dBFS, 20Hz to 10kHz - Better than 0.002%		
Digital to Digital (with SRC) Distortion	-1dBFS, 20Hz to 10kHz - Better than 0.005%		
Frequency Response (Analogue Input to Output)	20Hz to 20kHz +/- 0.5dB		
SYNCHRONISATION			
48kHz synchronisation	NTSC/PAL Video		
	Internal Crystal Reference		
	TTL Wordclock		
	AES/EBU Digital Input		

The system can be pre-set with up to five external sync sources, plus internal, such that if the 1st source fails, it will automatically switch to the 2nd, and so on.









.....

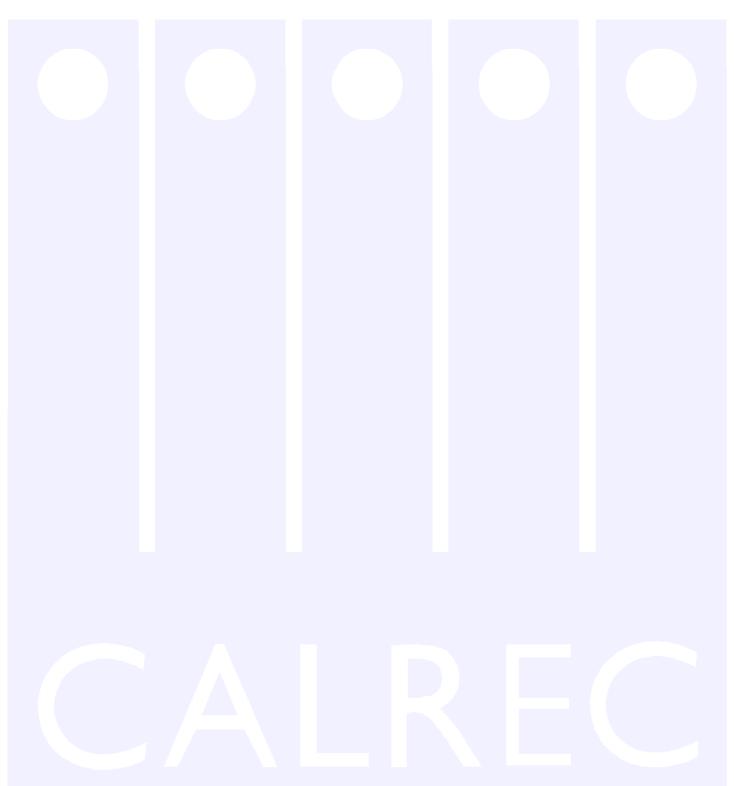
<u>NOTES</u>













Calrec Audio Ltd reserve the right to change specifications without notice. E & O.E. Alpha 100 is a trademark of Calrec Audio Ltd. All other trade marks acknowledged.