



USER GUIDE



by HARMAN



User Guide



IMPORTANT

Please read this manual carefully before using your mixer for the first time.



This equipment complies with the EMC directive 2004/108/EC and LVD 2006/95/EC

This product is approved to safety standards:

EFX Models

IEC 60065: 2001 + AMD1:2005
EN60065: 2002 + AMD1:2006+A11:2008
UL6500 7th Edition: 2003
CAN/CSA-E60065-03

and EMC standards:

EN55103-1: 1996 (E2)
EN55103-2: 1996 (E2)

EPM Models

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Part No. BD10.530000 Issue 1210KA

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For your own safety and to avoid invalidation of the warranty please read this section carefully.

SAFETY SYMBOL GUIDE

For your own safety and to avoid invalidation of the warranty all text marked with these symbols should be read carefully.

WARNINGS



The lightning flash with arrowhead symbol, is intended to alert the user to the presence of un-insulated “dangerous voltage” within the product’s enclosure that may be of sufficient magnitude to constitute a risk of electric shock to persons.

CAUTIONS



The exclamation point within an equilateral triangle is intended to alert the user to the presence of important operating and maintenance (servicing) instructions in the literature accompanying the appliance.

NOTES



Contain important information and useful tips on the operation of your equipment.



HEADPHONES SAFETY WARNING

Contain important information and useful tips on headphone outputs and monitoring levels.

Recommended Headphone Impedance \geq 150 Ohms.

IMPORTANT SAFETY INSTRUCTIONS

Read these instructions.

Keep these instructions.

Heed all warnings.

Follow all instructions.

Do not use this apparatus near water.

Clean only with a dry cloth.

Do not block any ventilation openings. Install in accordance with the manufacturer's instructions.

Do not install near any heat sources such as radiators, heat registers, stoves, or other apparatus (including amplifiers) that produce heat.

Do not defeat the safety purpose of a polarised or grounding type plug. A polarised plug has two blades with one wider than the other. A grounding type plug has two blades and a third grounding prong. The wide blade or the third prong are provided for your safety. If the provided plug does not fit into your outlet, consult an electrician for replacement of the obsolete outlet

Protect the power cord from being walked on or pinched particularly at plugs, convenience receptacles and the point where they exit from the apparatus.

Only use attachments/accessories specified by the manufacturer.



Use only with the cart, stand, tripod, bracket or table specified by the manufacturer, or sold with the apparatus. When a cart is used, use caution when moving the cart/apparatus combination to avoid injury from tip-over.

Unplug this apparatus during lightning storms or when unused for long periods of time.

Refer all servicing to qualified service personnel. Servicing is required when the apparatus has been damaged in any way, such as power-supply cord or plug is damaged, liquid has been spilled or objects fallen into the apparatus, the apparatus has been exposed to rain or moisture, does not operate normally, or has been dropped.

Note: It is recommended that all maintenance and service on the product should be carried out by Soundcraft or its authorised agents. Soundcraft cannot accept any liability whatsoever for any loss or damage caused by service, maintenance or repair by unauthorised personnel.

WARNING: To reduce the risk of fire or electric shock, do not expose this apparatus to rain or moisture.

Do not expose the apparatus to dripping or splashing and do not place objects filled with liquids, such as vases, on the apparatus.

No naked flame sources, such as lighted candles, should be placed on the apparatus.

Ventilation should not be impeded by covering the ventilation openings with items such as newspapers, table cloths, curtains etc.

THIS APPARATUS MUST BE EARTHED. Under no circumstances should the safety earth be disconnected from the mains lead.

The mains supply disconnect device is the mains plug. It must remain accessible so as to be readily operable when the apparatus is in use.

If any part of the mains cord set is damaged, the complete cord set should be replaced. The following information is for reference only.

The wires in the mains lead are coloured in accordance with the following code:

Earth (Ground): Green and Yellow (US - Green/Yellow)

Neutral: Blue (US - White)

Live (Hot): Brown (US - Black)

As the colours of the wires in the mains lead may not correspond with the coloured markings identifying the terminals in your plug, proceed as follows:

The wire which is coloured Green and Yellow must be connected to the terminal in the plug which is marked with the letter E or by the earth symbol. 

The wire which is coloured Blue must be connected to the terminal in the plug which is marked with the letter N

The wire which is coloured Brown must be connected to the terminal in the plug which is marked with the letter L

Ensure that these colour codes are followed carefully in the event of the plug being changed

This unit is capable of operating over a range of mains voltages as marked on the rear panel.

NOTE: This equipment has been tested and found to comply with the limits for a Class A digital device, pursuant to Part 15 of the FCC Rules. These limits are designed to provide reasonable protection against harmful interference when the equipment is operated in a commercial environment. This equipment generates, uses and can radiate radio frequency energy and, if not installed and used in accordance with the instruction manual, may cause harmful interference to radio communications. Operation of this equipment in a residential area is likely to cause harmful interference in which case the user will be required to correct the interference at his own expense.

This Class A digital apparatus meets the requirements of the Canadian Interference-Causing Equipment Regulations.

Cet appareil numérique de la Classe A respecte toutes les exigences du Règlement sur le matériel brouilleur du Canada.

INTRODUCTION

Thank you for purchasing a Soundcraft EFX/EPM mixer. The EFX/EPM range is a most cost-effective mixing solution, bringing you all the features and performance that you expect from a Soundcraft product, at an extraordinarily low price. The packaging, in which your mixer arrived, forms part of the product and must be retained for future use.

Owning a Soundcraft console brings you the expertise and support of one of the industry's leading manufacturers, and the results of nearly 3 decades of supporting some of the biggest names in the business. Our knowledge has been attained through working in close contact with leading professionals and institutes to bring you products designed to get the best possible results from your mixing.

Built to the highest standards using quality components and surface mount technology, the mixer is designed to be as easy to use as possible. We have spent years researching the most efficient methods of control for two key reasons: 1) Engineers, musicians, writers and programmers all need to have very few interruptions to the creative process; our products have been designed to be almost transparent, allowing this process to breathe.

2) Whether performing or recording, time is a very expensive and rare commodity. Our products have a user interface which is recognised by millions to be the industry standard because of its efficiency.

The sonic qualities of our products are exemplary - some of the same circuits which are used on our most expensive consoles are employed in the EFX and EPM, bringing you the great Soundcraft quality in a small format console without compromise.

You will also be glad to know you have a one year warranty with your product from the date of purchase. The mixer has been designed using the latest high-end software based engineering packages. Every console from Soundcraft has been proven to stand up to all the stress and rigours of modern day mixing environments.

The entire mixer is manufactured using some of the most advanced techniques in the world, from high density surface mount PCB technology, to computer aided test equipment able to measure signals well outside the range of normal hearing. As each console passes through to be quality checked before packing, there is also a human listening station. Something we have learnt over the years is that the human touch counts - and only by using people can you ensure the product meets the high demands of the user.



ADVICE FOR THOSE WHO PUSH THE BOUNDARIES

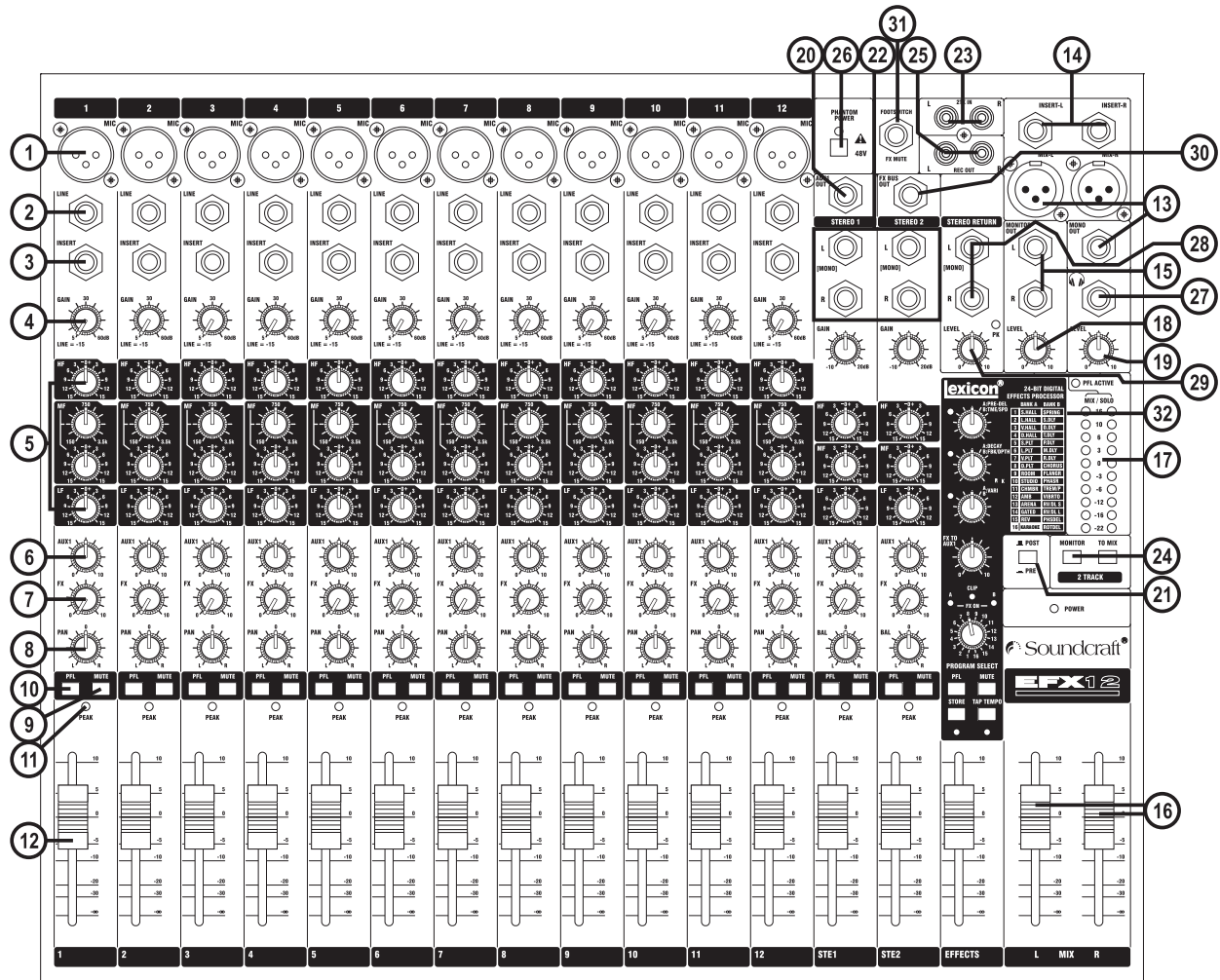
Although your new console will not output any sound until you feed it signals, it has the capability to produce sounds which when monitored through an amplifier or headphones can damage hearing over time.

Please take care when working with your audio - if you are manipulating controls which you don't understand (which we all do when we are learning), make sure your monitors are turned down. Remember that your ears are the most important tool of your trade, look after them, and they will look after you.

Most importantly - don't be afraid to experiment to find out how each parameter affects the sound - this will extend your creativity and help you to get the best from your mixer and the most respect from your artists and audience.

OVERVIEW

To get you working as fast as possible, this manual begins with a 60-second guide. Here you can find quick information on any feature of the EFX console.



THE 60-SECOND GUIDE - EFX

1 MIC INPUT (XLR)

Connect Microphones here. If you are using a condenser mic, ensure phantom power is supplied by pressing the switch at the top of the master section.

WARNING: Do Not apply Phantom Power before connecting a microphone.

2 LINE INPUT (1/4" Jack)

Connect Line level sources here, e.g. Synth, Drum Machine, DI etc.

3 INSERT POINT (1/4" Jack)

Connect Signal processors here, e.g. Compressor, Gate etc.

4 GAIN CONTROL

Adjust this to increase or decrease the level of the incoming signal.

5 EQ STAGE

Adjust these controls to change the signal tone (the character of the signal).

6 AUX 1 SEND

Adjust this control to change the level of the signal to an artist's monitors (headphones/in-ear/stage monitors). Aux 1 is globally switchable pre/post fade.

7 FX SEND

This control sets the level of the post-fade signal being sent to the FX bus; from there it is routed to the FX processor.

8 PAN CONTROL


Use this control to position the signal within the stereo field.

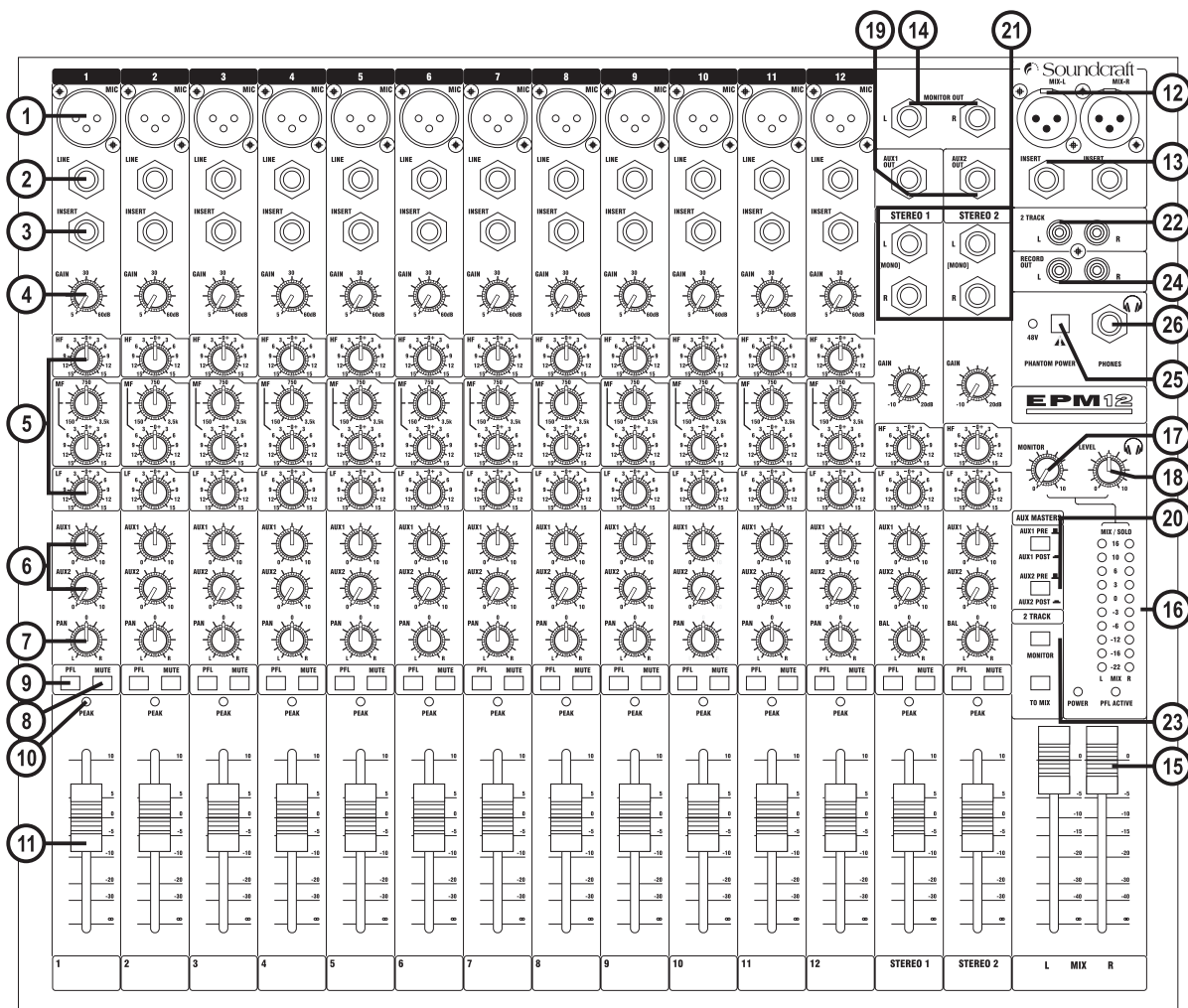
9 MUTE SWITCH

When this is pressed you will hear no signal from the channel (post-mute signals).

10 PFL

When pressed the signal will appear on the monitor and headphone outputs - use this to monitor the post-EQ signal from the channel.

- 11 PEAK LED** This is used to indicate that the signal is close to distorting (clipping) on a specific channel.
- 12 INPUT CHANNEL FADER** This is used to control the level fed to the Mix Bus and post-fade sends.
- 13 MIX OUTPUTS (XLR) & MONO OUT (1/4" Jack)** Connect these to your analogue recording device, or to your amplification system.
- 14 MIX INSERTS (1/4" Jack)** This is a pre-fade break in the signal path which can be used to feed a dynamics or mastering device. The signal is sent from the tip of the jack plug and the return path comes back in on the ring of the jack plug.
- 15 MONITOR O/Ps (1/4" Jack)** These are used to feed your monitoring system. This can be directly connected to powered monitors, or indirectly via an amplifier to standard monitors.
- 16 MASTER FADERS** These faders control the overall level of the mix outputs.
- 17 MAIN METERS** These show the level of the mix outputs. When the PFL ACTIVE LED is lit, the meters show the level of the selected PFL signal.
- 18 MONITOR CONTROL** This controls the level of the signal sent to your monitoring system.
- 19 PHONES CONTROL** This controls the level of the signal sent to the headphones jack socket.
- 20 AUX 1 OUTPUT (1/4" Jack)** This output can be used to send the channel signal to an artist's monitors (headphones/in-ear/stage monitors). It is switchable pre/post fade.
- 21 AUX CONTROL** This switch globally switches the AUX1 feed on all the input modules to be either pre or post-fade.
- 22 STEREO INPUTS (1/4" Jack)** These two inputs can be used to connect line level stereo inputs from keyboards, sound modules, samplers, computer based audio cards etc. These inputs pass through a normal channel strip, with EQ, Auxes and a Balance control.
- 23 2-TRACK INPUTS (RCA Phono)** Here you can connect the playback from your recording device.
- 24 2-TRACK CONTROLS** Use these to control the 2 Track signal. The MONITOR switch sends the signal to the monitor outputs and phones, whilst the TO MIX switch sends it to the main mix.
- 25 RECORD OUTPUTS (RCA Phono)** You can connect these to the inputs of your recording device.
- 26 PHANTOM POWER** Press this to switch the phantom power (48V) on for condenser microphones.
-  **WARNING:** Do Not apply Phantom Power before connecting a microphone.
- 27 HEADPHONES (1/4" Jack)** Plug your headphones into this socket. Recommended headphones impedance is 150 ohms or greater.
- 28 STEREO RETURN INPUTS** This pair of inputs accept 3-pole 'A' gauge (TRS) jacks. Use these inputs for sources such as keyboards, drum machines, synths or CDs. The inputs are BALANCED. Mono sources may be used by plugging into the left jack only.
- 29 STEREO RETURN CONTROL** This control sets the level of signal routed to the main mix busses. There is an associated PK LED to warn of signals which are too high.
- 30 FX BUS OUTPUT** This output carries the signal from the FX bus. It could be used as a second Aux Output if desired, if the FX Processor is not needed at the time. The FX sends on the inputs channels to the FX bus are always post-fade.
- 31 FOOTSWITCH CONNECTOR** This is used by the FX Processor, see page 26.
- 32 LEXICON® FX PROCESSOR** See the information starting on page 27



THE 60-SECOND GUIDE - EPM

1 MIC INPUT (XLR)

Connect Microphones here. If you are using a condenser mic, ensure phantom power is supplied by pressing the switch at the top of the master section.

⚠ WARNING: Do Not apply Phantom Power before connecting a microphone.

2 LINE INPUT (1/4" Jack)

Connect Line level sources here, e.g. Synth, Drum Machine, DI etc.

3 INSERT POINT (1/4" Jack)

Connect Signal processors here, e.g. Compressor, Gate etc.

4 GAIN CONTROL

Adjust this to increase or decrease the level of the incoming signal.

5 EQ STAGE

Adjust these controls to change the signal tone (the character of the signal).

6 AUX 1/2 SENDS

Adjust these controls to change the level of the signal to an artist's monitors (headphones/in-ear/stage monitors). Each Aux is globally switchable pre/post fade.

7 PAN CONTROL


Use this control to position the signal within the stereo field.

8 MUTE SWITCH

When this is pressed you will hear no signal from the channel (post-mute signals).

9 PFL

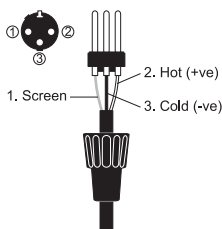
When pressed the signal will appear on the monitor and headphone outputs - use this to monitor the post-EQ signal from the channel.

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11 INPUT CHANNEL FADER	This is used to control the level fed to the Mix Bus and post-fade sends.
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13 MIX INSERTS (1/4" Jack)	This is a pre-fade break in the signal path which can be used to feed a dynamics or mastering device. The signal is sent from the tip of the jack plug and the return path comes back in on the ring of the jack plug.
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17 MONITOR CONTROL	This controls the level of the signal sent to your monitoring system.
18 PHONES CONTROL	This controls the level of the signal sent to the headphones jack socket.
19 AUX OUTPUTS (1/4" Jack)	These outputs can be used to send the channel signal to an artist's monitors (headphones/in-ear/stage monitors). They are switchable pre/post fade.
20 AUX CONTROL	These switches globally switch the AUX1 and AUX 2 feeds on all the input modules to be either pre or post-fade.
21 STEREO INPUTS (1/4" Jack)	These two inputs can be used to connect line level stereo inputs from keyboards, sound modules, samplers, computer based audio cards etc. These inputs pass through a normal channel strip, with EQ, Auxes and a Balance control.
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25 PHANTOM POWER	Press this to switch the phantom power (48V) on for condenser microphones.  WARNING: Do Not apply Phantom Power before connecting a microphone.
26 HEADPHONES (1/4" Jack)	Plug your headphones into this socket. Recommended headphones impedance is 150 ohms or greater.

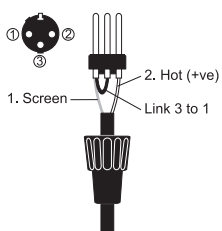
WIRING UP



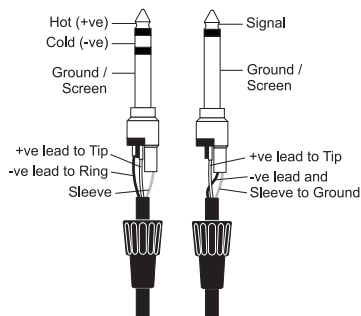
Balanced Mic XLR



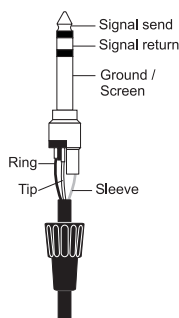
Unbalanced Mic XLR



3 pole jack 2 pole jack
Balanced Unbalanced



Inserts



Mic Input

The MIC input accepts XLR-type connectors and is designed to suit a wide range of BALANCED or UNBALANCED low-level signals, whether from delicate vocals requiring the best low-noise performance, or drum kits needing maximum headroom. Professional dynamic, condenser or ribbon mics are best because these will be LOW IMPEDANCE. While you can use low-cost HIGH IMPEDANCE mics, you do not get the same degree of immunity to interference on the microphone cable and as a result the level of background noise may be higher. If you turn the PHANTOM POWER on, the socket provides a suitable powering voltage for professional condenser mics.



DO NOT use UNBALANCED sources with the phantom power switched on. The voltage on pins 2 & 3 of the XLR connector may cause serious damage. BALANCED dynamic mics may normally be used with phantom power switched on (contact your microphone manufacturer for guidance)

The input level is set using the input GAIN knob.

The LINE input offers the same gain range as the MIC input, but at a higher input impedance, and is 20dB less sensitive. This is suitable for most line level sources.



WARNING !

Start with the input GAIN knob turned fully anticlockwise when plugging high level sources into the LINE input to avoid overloading the input channel or giving you a very loud surprise!

Line Input

Accepts 3-pole 'A' gauge jacks, or 2-pole mono jacks which will automatically ground the 'cold' input. Use this input for sources other than mics, such as keyboards, drum machines, synths, tape machines or DI boxes. The input is BALANCED for low noise and immunity from interference, but you can use UNBALANCED sources by wiring up the jacks as shown, although you should then keep cable lengths as short as possible to minimise interference pick-up on the cable. Note that the ring must be grounded if the source is unbalanced. Set the input level using the GAIN knob, starting with the knob turned fully anticlockwise. Unplug any MIC connection when using the LINE input.

Insert Point

The unbalanced, pre-EQ insert point is a break in the channel signal path, allowing limiters, compressors, special EQ or other signal processing units to be added in the signal path. The Insert is a 3-pole 'A' gauge jack socket which is normally bypassed. When a jack is inserted, the signal path is broken, just before the EQ section.

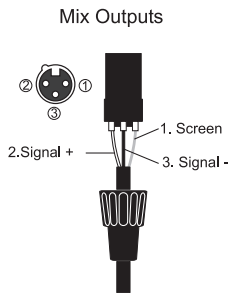
The signal from the channel appears on the TIP of the plug and is returned on the RING, with the sleeve as a common ground.

The Send may be tapped off as an alternative pre-fade, pre-EQ direct output if required, using a lead with tip and ring shorted together so that the signal path is not interrupted.

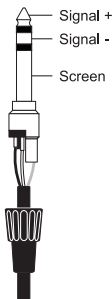
Stereo Inputs STEREO 1/2 and Stereo

These accept 3-pole 'A' gauge jacks, or 2-pole mono jacks which will automatically ground the 'cold' input. Use these inputs for sources such as keyboards, drum machines, synths, tape machines or as returns from processing units. The input is BALANCED for low noise and immunity from interference, but you can use UNBALANCED sources by wiring up the jacks as shown, although you should then keep cable lengths as short as possible to minimise interference pick-up on the cable. Note that the ring must be grounded if the source is unbalanced. Mono sources can be fed to both paths by plugging into the Left jack only.

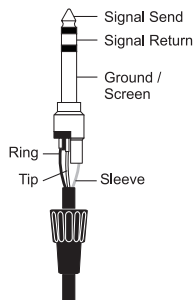
OUTPUTS



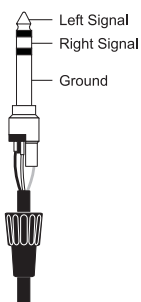
Aux Outputs Monitor Outputs



Mix Inserts

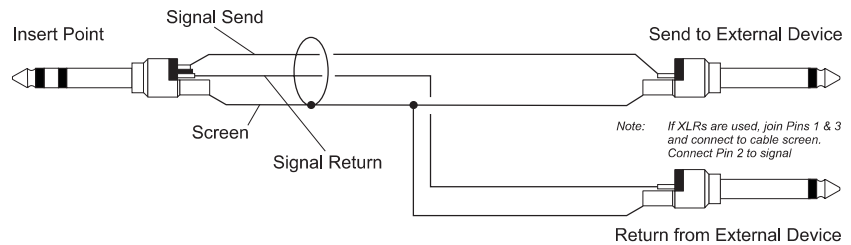


Headphones



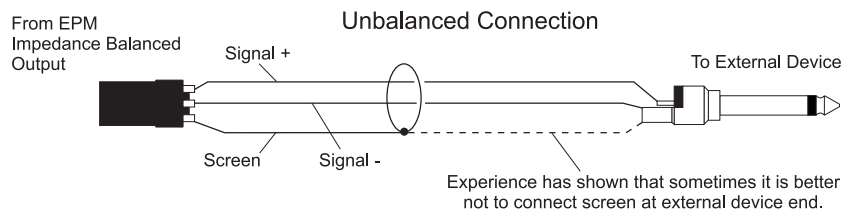
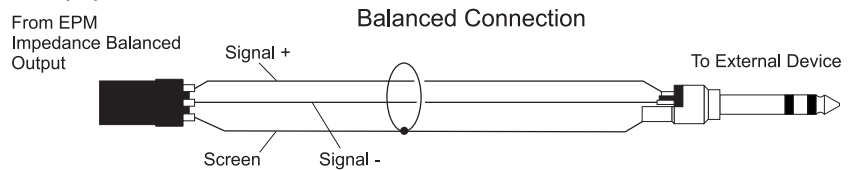
Mix Inserts

The unbalanced, pre-fade Mix insert point is a break in the output signal path to allow the connection of, for example, a compressor/limiter or graphic equaliser. The Insert is a 3-pole 'A' gauge jack socket which is normally bypassed. When a jack is inserted, the signal path is broken, just before the mix fader. The mix signal appears on the TIP of the plug and is returned on the RING. A 'Y' lead may be required to connect to equipment with separate send and return jacks as shown below:



Mix Outputs

The MIX outputs are on XLR's, wired as shown, and incorporate impedance balancing, allowing long cable runs to balanced amplifiers and other equipment.



Aux & FX Bus Outputs

The Aux & FX Bus outputs are on 3-pole 'A' gauge jack sockets, wired as shown on the left, and are impedance-balanced, allowing long cable runs to balanced amplifiers and other equipment. **Note EPM has 2 x AUX, EFX has 1 x AUX, 1 x FX.**

Headphones

The PHONES output is a 3-pole 'A' gauge jack, wired as a stereo output as shown, ideally for headphones of 150Ω or greater. 8Ω headphones are not recommended.

Polarity (Phase)

You will probably be familiar with the concept of polarity in electrical signals and this is of particular importance to balanced audio signals. Just as a balanced signal is highly effective at cancelling out unwanted interference, so two microphones picking up the same signal can cancel out, or cause serious degradation of the signal if one of the cables has the +ve and -ve wires reversed. This phase reversal can be a real problem when microphones are close together and you should therefore always take care to connect pins correctly when wiring audio cables.

Grounding and Shielding

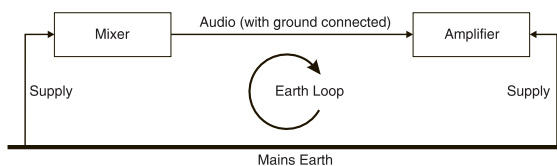
For optimum performance use balanced connections where possible and ensure that all signals are referenced to a solid, noise-free earthing point and that all signal cables have their screens connected to ground. In some unusual circumstances, to avoid earth or ground 'loops' ensure that all cable screens and other signal earths are connected to ground only at their source and not at both ends.

If the use of unbalanced connections is unavoidable, you can minimise noise by following these wiring guidelines:

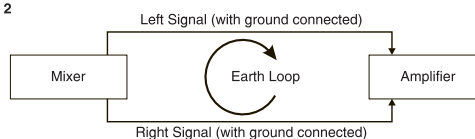
- On INPUTS, unbalance at the source and use a twin screened cable as though it were balanced.
- On OUTPUTS, connect the signal to the +ve output pin, and the ground of the output device to -ve. If a twin screened cable is used, connect the screen only at the mixer end.
- Avoid running audio cables or placing audio equipment close to thyristor dimmer units or power cables.
- Noise immunity is improved significantly by the use of low impedance sources, such as good quality professional microphones or the outputs from most modern audio equipment. Avoid cheaper high impedance microphones, which may suffer from interference over long cable runs, even with well-made cables.

Grounding and shielding is still seen as a black art, and the suggestions above are only guidelines. If your system still hums, an earth/ground loop is the most likely cause. Two examples of how an earth loop can occur are shown below.

Example 1



Example 2



Warning!

Under NO circumstances must the AC power mains earth be disconnected from the mains lead.



PROBLEM SOLVING

Basic problem solving is within the scope of any user if a few basic rules are followed.

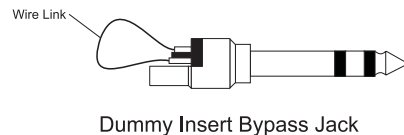
- Get to know the Block Diagram of your console (see page 16/17).
- Get to know what all controls and/or connections in the system are supposed to do.
- Learn where to look for common trouble spots.

The Block Diagram is a representative sketch of all the components of the console, showing how they connect together and how the signal flows through the system. Once you have become familiar with the various component blocks you will find the Block Diagram is quite easy to follow and you will have gained a valuable understanding of the internal structure of the console.

Each component has a specific function and only by getting to know what each part is supposed to do will you be able to tell if there is a genuine fault! Many “faults” are the result of incorrect connection or control settings which may have been overlooked.

Basic Troubleshooting is a process of applying logical thought to the signal path through the console and tracking down the problem by elimination.

- Swap input connections to check that the source is really present. Check both Mic and Line inputs.
- Eliminate sections of the channel by using the insert point to re-route the signal to other inputs that are known to be working.
- Route channels to different outputs or to auxiliary sends to identify problems on the Master section.
- Compare a suspect channel with an adjacent channel which has been set up identically. Use PFL to monitor the signal in each section.
- Insert-point contact problems may be checked by using an insert bypass jack with tip and ring shorted together as shown below. If the signal appears when the jack is inserted it shows that there is a problem with the normalling contacts on the jack socket, caused by wear or damage, or often just dirt or dust. Keep a few in your gig tool box.



If in doubt please contact Soundcraft customer support.

PRODUCTS UNDER WARRANTY

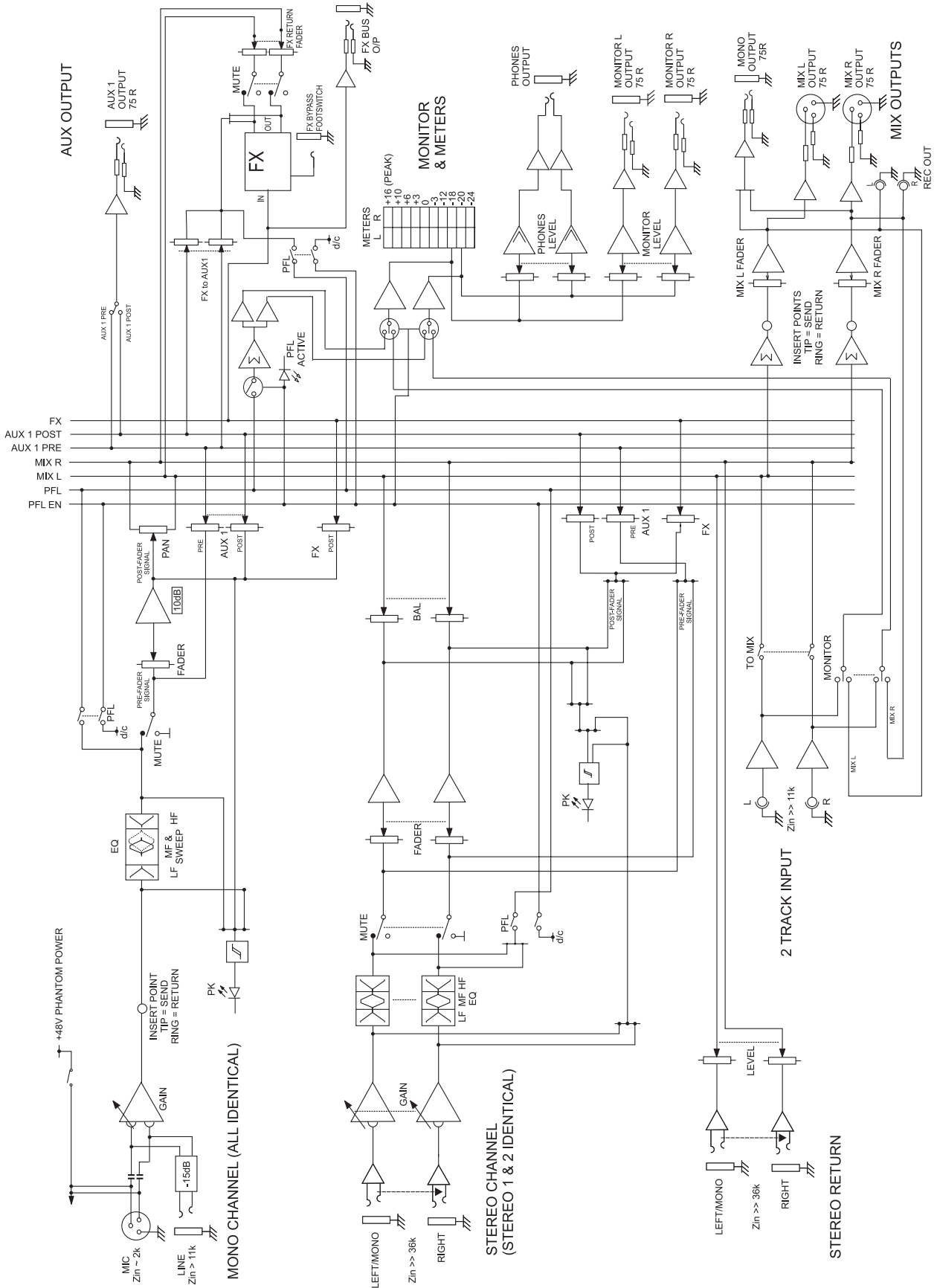
UK customers should contact their local dealer. Customers outside the UK are requested to contact their territorial distributor who is able to offer support in the local time zone and language. Please see the distributor listings on our website (www.soundcraft.com) to locate your local distributor.

OUT-OF-WARRANTY PRODUCTS

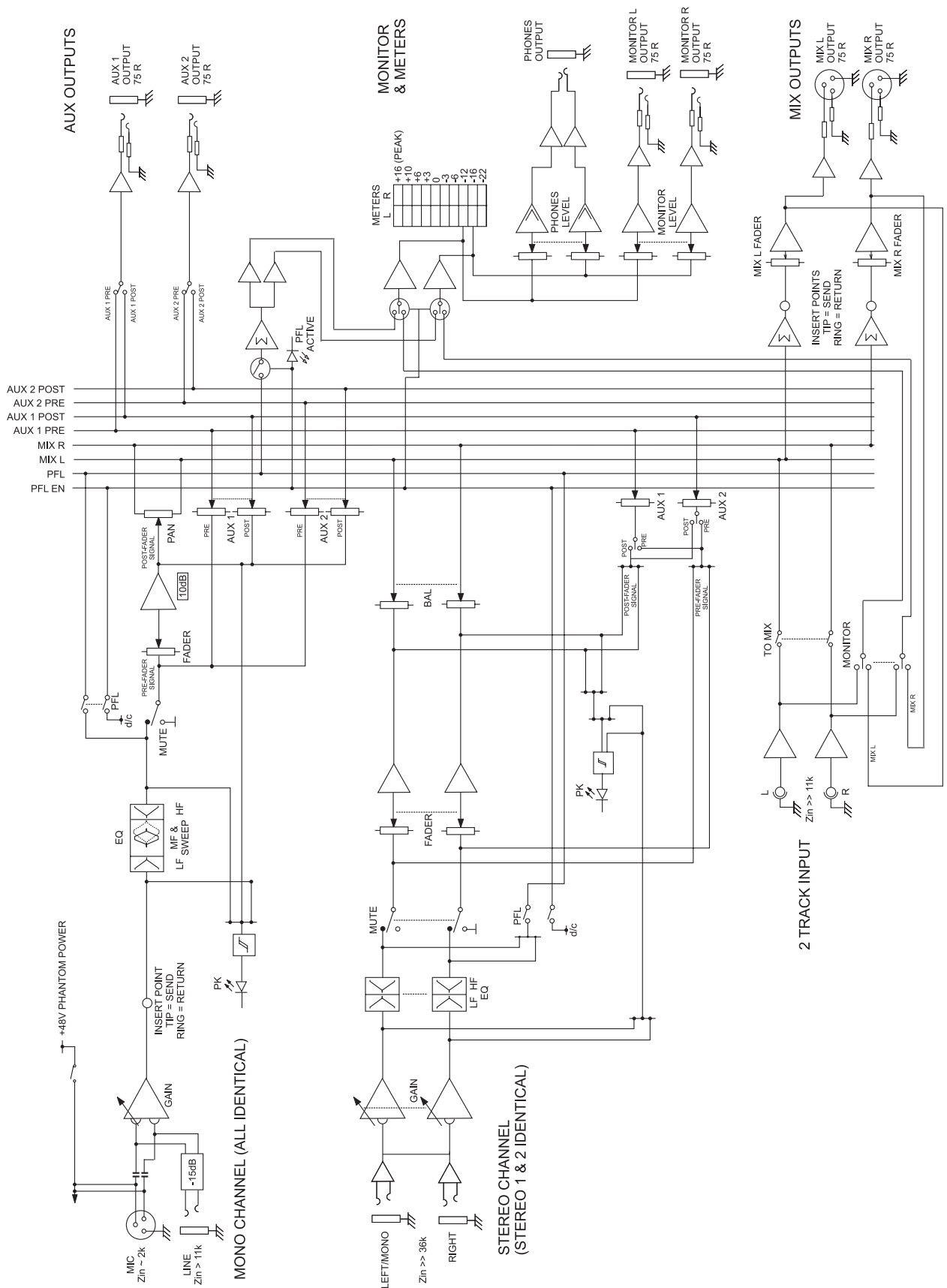
For out-of-warranty consoles purchased in the United Kingdom, please contact the Customer Services Department (e-mail: soundcraft.csd@harman.com) at the factory in Potters Bar, Hertfordshire: Telephone +44 (0)1707 665000. For all other out-of-warranty consoles, please contact the appropriate territorial distributor.

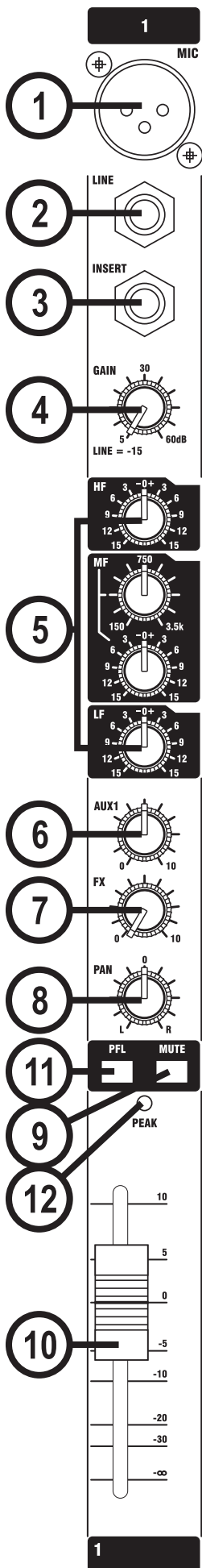
When mailing or faxing please remember to give as much information as possible. This should include your name, address and a daytime telephone number. Should you experience any difficulty please contact Customer Services Department (e-mail: soundcraft.csd@harman.com)

BLOCK DIAGRAM - Soundcraft EFX



BLOCK DIAGRAM - Soundcraft EPM





MONO INPUT CHANNEL - EFX shown

1 Mic Input

The MIC input accepts XLR-type connectors and is designed to suit a wide range of BALANCED or UNBALANCED signals. Professional dynamic, condenser or ribbon mics are best because these will be LOW IMPEDANCE. You can use low-cost HIGH IMPEDANCE mics, but the level of background noise will be higher. If you turn the PHANTOM POWER on (top right-hand side of the mixer) the socket provides a suitable powering voltage for professional condenser mics.



ONLY connect condenser microphones with the +48V powering OFF, and ONLY turn the +48V powering on or off with all output faders DOWN, to prevent damage to the mixer or external devices.

TAKE CARE when using unbalanced sources, which may be damaged by the phantom power voltage on pins 2 & 3 of the XLR connector.

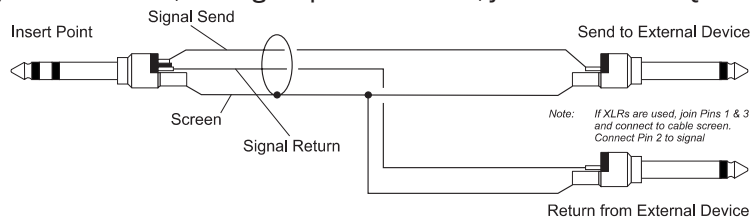
Unplug any mics if you want to use the LINE Input. The input level is set using the GAIN knob.

2 Line Input

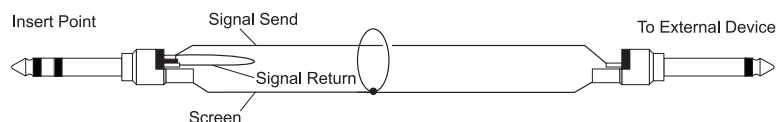
Accepts 3-pole 'A' gauge (TRS) jacks. Use this input for sources other than mics, such as keyboards, drum machines, synths, tape machines or DI'd guitars. The input is BALANCED for low noise and top quality from professional equipment, but you can use UNBALANCED sources by wiring up the jacks as shown below, although you should then keep cable lengths as short as possible. Unplug anything in the MIC input if you want to use this socket. Set the input level using the GAIN knob.

3 Insert Point

The unbalanced, pre-EQ insert point is a break in the channel signal path, allowing limiters, compressors, special EQ or other signal processing units to be added in the signal path. The Insert is a 3-pole 'A' gauge jack socket which is normally bypassed. When a jack is inserted, the signal path is broken, just before the EQ section.

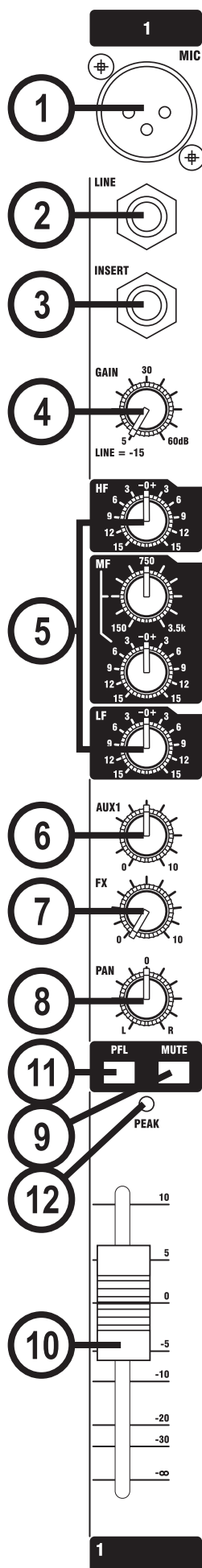


The Send may also be tapped off as an alternative pre-fade, pre-EQ direct output if required, using a lead with tip and ring shorted together so that the signal path is not interrupted (see below).



4 Gain

This knob sets how much of the source signal is sent to the rest of the mixer. Too high, and the signal will distort as it overloads the channel. Too low, and the level of any background hiss will be more noticeable and you may not be able to get enough signal level to the output of the mixer.



Note that some sound equipment, particularly that intended for domestic use, operates at a lower level (-10dBV) than professional equipment and will therefore need a higher gain setting to give the same output level. See “Initial Setup” on page 31 to learn how to set GAIN correctly.

5 Equaliser

The Equaliser (EQ) allows fine manipulation of the sound, particularly to improve the sound in live PA applications where the original signal is often far from ideal and where slight boosting or cutting of particular voice frequencies can really make a difference to clarity. There are three sections giving the sort of control usually only found on much larger mixers. The EQ knobs can have a dramatic effect, so use them sparingly and listen carefully as you change any settings so that you get to know how they affect the sound.

HF EQ

Turn to the right to boost high (treble) frequencies above 12kHz by up to 15dB, adding crispness to cymbals, vocals and electronic instruments. Turn to the left to cut by up to 15dB, reducing hiss or excessive sibilance which can occur with certain types of microphone. Set the knob in the centre-detented position when not required.

MID EQ

There are two knobs which work together to form a SWEPT MID EQ. The lower knob provides 15dB of boost and cut, just like the HF EQ knob, but the frequency at which this occurs can be set by the upper knob over a range of 150Hz to 3.5kHz. This allows some truly creative improvement of the signal in live situations, because this mid band covers the range of most vocals. Listen carefully as you use these controls together to find how particular characteristics of a vocal signal can be enhanced or reduced. Set the lower knob to the centre-detented position when not required.

LF EQ

Turn to the right to boost low (bass) frequencies below 80Hz by up to 15dB, adding warmth to vocals or extra punch to synths, guitars and drums. Turn to the left to cut low frequencies by up to 15dB for reducing hum, stage rumble or to improve a mushy sound. Set the knob to the centre-detented position when not required.

6 Aux 1 Send

This is used to set up a separate mix for FOLDBACK or recording, and the combination of each Aux Send is mixed to the Aux1 Output. For recording it is useful for the signal to fade up and down with the fader (this is called POST-FADE), but for Foldback or Monitor feeds it is important for the send to be independent of the fader (this is called PRE-FADE).

AUX 1 SEND is globally switchable between pre and post-fade (see master section on page 24/25).

7 FX Send (EFX only)

This control sets the level of the post-fade signal being sent to the FX bus; from there it is routed to the FX processor.

7 Aux 2 Send (EPM only)

This is used to set up a separate mix for FOLDBACK or recording, and the combination of each Aux Send is mixed to the Aux 2 Output. For recording it is useful for the signal to fade up and down with the fader (this is called POST-FADE), but for Foldback or Monitor feeds it is important for the send to be independent of the fader (this is called PRE-FADE).

AUX 2 SEND is globally switchable between pre and post-fade (see master section on page 24/25).

8 PAN

This control sets the amount of the channel signal feeding the Left and Right MIX buses, allowing you to move the source smoothly across the stereo image. When the control is turned fully left or right you are able to route the signal at unity gain to either left or right outputs individually.

9 MUTE

All outputs from the channel except inserts are on when the MUTE switch is released and muted when the switch is down, allowing levels to be pre-set before the signal is required.

10 INPUT CHANNEL FADER

The 60mm FADER, with a custom-designed law to give even smoother control of the overall signal level in the channel strip, allows precise balancing of the various source signals being mixed to the Master Section. You get most control when the input GAIN is set up correctly, giving full travel on the fader. See the “Initial Setup” section on page 36 for help in setting a suitable signal level.

11 PFL

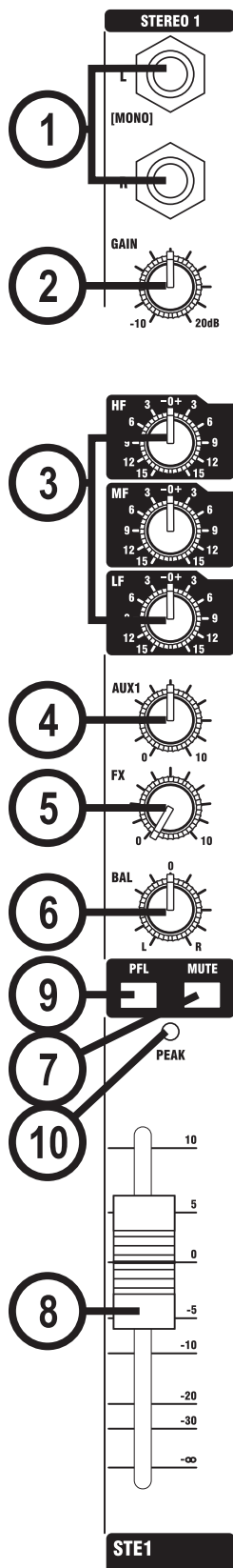
When the latching PFL switch is pressed, the pre-fade pre-mute signal is fed to the headphones, control room output and meters, where it replaces the MIX. The PFL ACTIVE LED on the Master section illuminates to warn that a PFL is active. This is a useful way of listening to any required input signal without interrupting the main mix, for making adjustments or tracing problems. When PFL is pressed anywhere on the console, the Control Room outputs automatically switch from monitoring the Mix Outputs.

12 PEAK LED

This LED will light when the signal level approaches clipping at any of the three monitored points:

- a) PRE-EQ
- b) POST-EQ
- c) POST-FADE

STEREO INPUT CHANNELS (EFX shown)



1 INPUTS STEREO 1/2

These inputs accept 3-pole 'A' gauge (TRS) jacks. Use these inputs for sources such as keyboards, drum machines, synths, tape machines or processing units. The inputs are BALANCED for low noise and top quality from professional equipment, but you can use UNBALANCED sources by wiring up the jacks as shown in the "Wiring it Up" section earlier in this manual, although you should then keep cable lengths as short as possible. Mono sources may be used by plugging into the left jack only.

2 GAIN

The GAIN control sets the level of the channel signal.

3 EQUALISER

HF EQ

Turn to the right to boost high (treble) frequencies, adding crispness to percussion from drum machines, synths and electronic instruments. Turn to the left to cut these frequencies, reducing hiss or excessive brilliance. Set the knob in the centre-detented position when not required. The control has a shelving response giving 15dB of boost or cut at 12kHz.

MF EQ

Turn to the right to boost mid frequencies, turn to the left to cut these frequencies. The centre frequency of the MID EQ is 720Hz

LF EQ

Turn to the right to boost low (bass) frequencies, adding extra punch to synths, guitars and drums. Turn to the left to reduce hum, boominess or improve a mushy sound. Set the knob to the centre-detented position when not required. The control has a shelving response giving 15dB of boost or cut at 80Hz.

4 AUX 1 SEND

This is used to set up a separate mix for FOLDBACK or recording, and the combination of each Aux Send is mixed to the Aux1 Output. For recording it is useful for the signal to fade up and down with the fader (this is called POST-FADE), but for Foldback or Monitor feeds it is important for the send to be independent of the fader (this is called PRE-FADE).

AUX 1 SEND is globally switchable between pre and post-fade (see master section on page 24/25). The send pot is fed with a mono sum of the L & R signals.

5 FX SEND (EFX only)

This control sets the level of the post-fade signal being sent to the FX bus; from there it is routed to the FX processor.

5 AUX 2 SEND (EPM only)

This is used to set up a separate mix for FOLDBACK or recording, and the combination of each Aux Send is mixed to the Aux 2 Output. For recording it is useful for the signal to fade up and down with the fader (this is called POST-FADE), but for Foldback or Monitor feeds it is important for the send to be independent of the fader (this is called PRE-FADE).

AUX 2 SEND is globally switchable between pre and post-fade (see master section on page 24/25). The send pot is fed with a mono sum of the L & R signals.

6 BALANCE

This control sets the amount of the channel signal feeding the Left and Right MIX buses, allowing you to balance the source in the stereo image. When the control is turned fully right or left you feed only that side of the signal to the mix. Unity gain is provided by the control in the centre-detented position.

7 MUTE

All outputs from the channel are enabled when the MUTE switch is released and muted when the switch is down.

8 FADER

The 60mm FADER gives you smooth control of the overall signal level in the channel strip, allowing precise balancing of the various source signals being mixed to the Master Section. It is important that the input level is set correctly to give maximum travel on the fader which should normally be used at around the “0” mark. See the “Initial Setup” section on page 36 for help in setting the right level.

9 PFL

When the latching PFL switch is pressed, the pre-fade pre-mute signal is fed in mono to the headphones, control room output and meters, where it replaces the MIX. The PFL ACTIVE LED on the Master section illuminates to warn that a PFL is active. The Left and Right meters display the PFL signal in mono. This is a useful way of listening to any required input signal without interrupting the main mix, for making adjustments or tracing problems.

10 CHANNEL PEAK LED

This LED will light when the signal level approaches clipping at any of the three monitored points:

- a) PRE-EQ
- b) POST-EQ
- c) POST-FADE

MASTER SECTION (EFX)

1 PHANTOM POWER

Many professional condenser mics need PHANTOM POWER, which is a method of sending a powering voltage down the same wires as the mic signal. Press the switch to enable the +48V power to all of the MIC inputs. The adjacent LED illuminates when the power is active.



WARNING: TAKE CARE when using unbalanced mics which may be damaged by the phantom power voltage. Balanced dynamic mics can normally be used with phantom power switched on (contact your microphone manufacturer for guidance).

phantom power switched on (contact your microphone manufacturer for guidance).

Mics should always be plugged in, and all output faders set to minimum before switching the Phantom Power ON to avoid damage to external equipment

2 POWER INDICATOR

This LED lights to show when power is connected to the console.

3 MASTER FADERS

The MASTER FADERS set the final level of the MIX outputs, and separate faders are provided for each output. These should normally be set close to the '0' mark if the input GAIN settings have been correctly set, to give maximum travel on the faders for smoothest control.

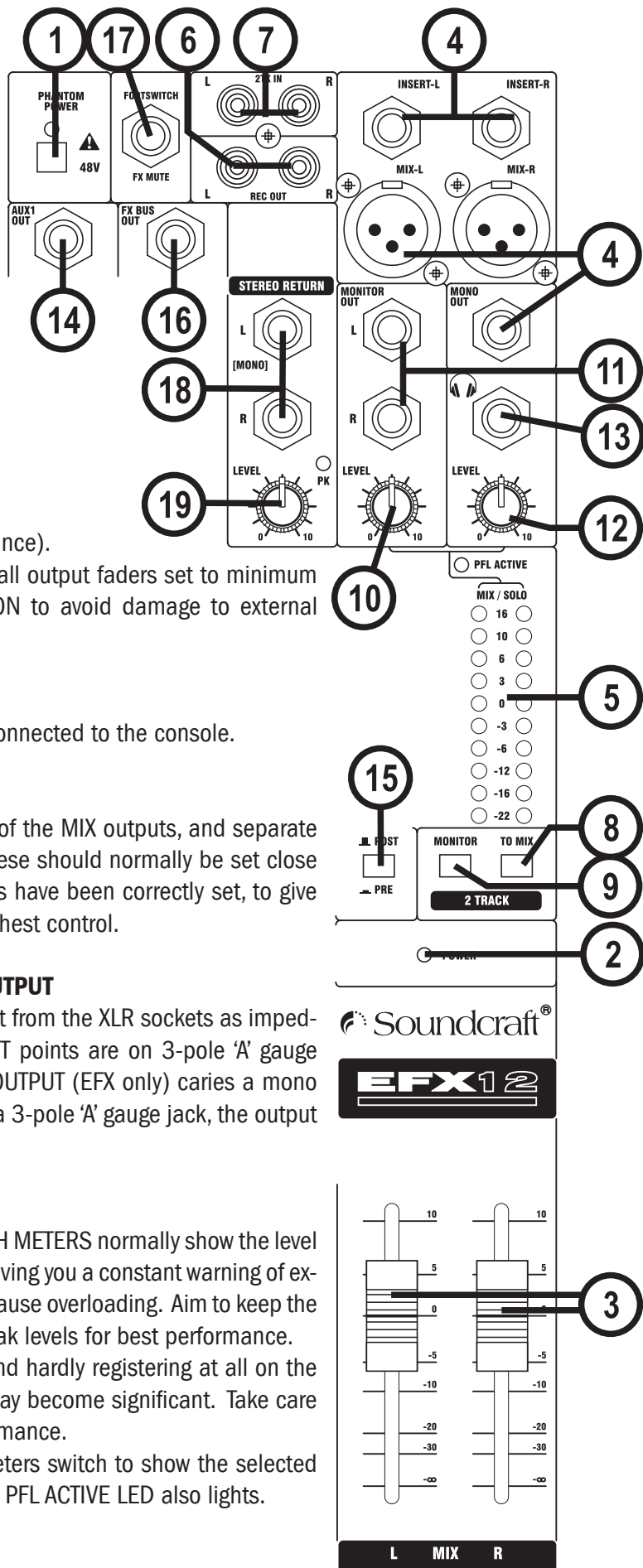
4 MIX OUTPUTS, INSERTS & MONO OUTPUT

The Mix LEFT and RIGHT outputs are sent from the XLR sockets as impedance-balanced signals. The Mix INSERT points are on 3-pole 'A' gauge jacks and are unbalanced. The MONO OUTPUT (EFX only) carries a mono sum of the left and right mix signals on a 3-pole 'A' gauge jack, the output is impedance-balanced.

5 BARGRAPH METERS

The three-colour peak reading BARGRAPH METERS normally show the level of the MIX RIGHT and MIX LEFT outputs, giving you a constant warning of excessive peaks in the signal which might cause overloading. Aim to keep the signal within the amber segments at peak levels for best performance. Similarly, if the output level is too low and hardly registering at all on the meters, the level of background noise may become significant. Take care to set up the input levels for best performance.

When any PFL switch is pressed, the meters switch to show the selected PFL signal on both meters, in mono; the PFL ACTIVE LED also lights.



6 RECORD OUTPUTS

These two RCA outputs carry a copy of the MIX L and MIX R signals. They allow the use of a recording device, e.g. DAT player, Minidisc recorder etc.

7 2 TRACK IN

These two RCA phono sockets are unbalanced Left and Right line-level inputs, used for connecting a playback device.

8 2 TRACK TO MIX

Press this switch to route the 2 Track in signals, connected to the Left and Right RCA sockets 7, to the MIX.

9 MONITOR 2 TRACK

Press this switch to route the 2 Track signal to the monitor and phones, over-riding the default Monitor/Phones signal.

10 MONITOR LEVEL

This control sets the level to the MONITOR LEFT & RIGHT outputs. If headphones are plugged into the PHONES jack, the headphone level will track the Monitor Level.

11 MONITOR OUTPUTS

The Monitor Outputs are on 3-pole 'A' gauge jacks and are balanced connections.

12 PHONES LEVEL

This control sets the output level to the Headphone outputs. If headphones are plugged into the PHONES jack, then the knob sets a comfortable headphone listening level without affecting the Monitor output levels.

13 HEADPHONES JACK

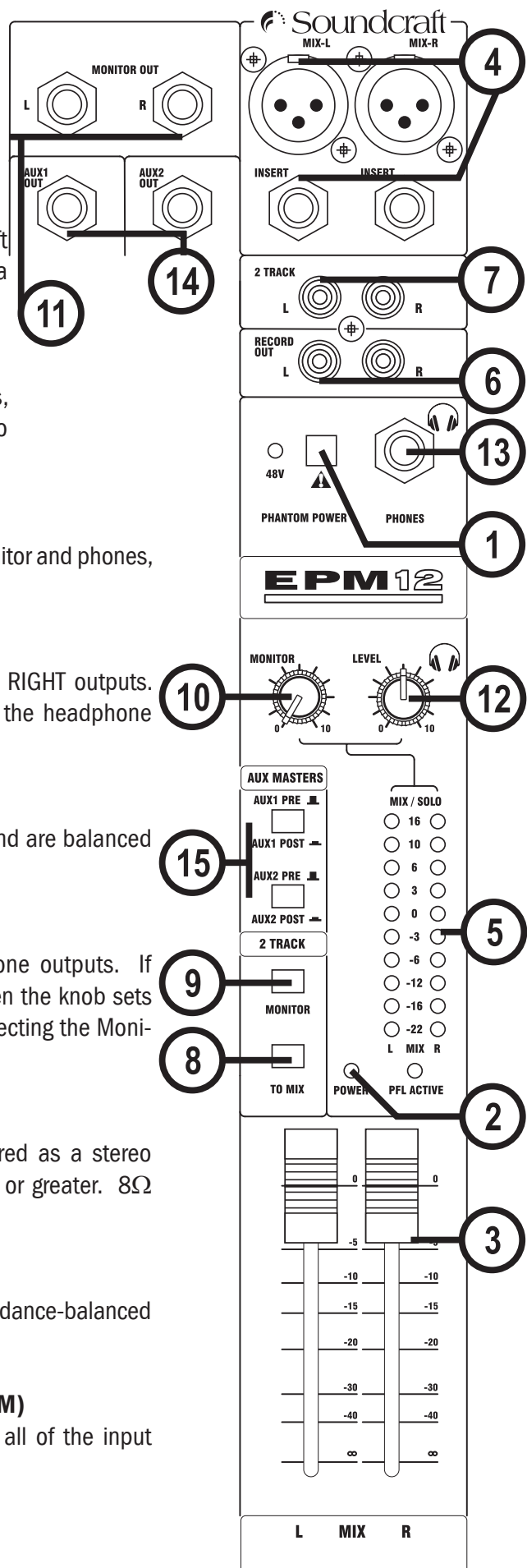
The PHONES output is a 3-pole 'A' gauge jack, wired as a stereo output as shown, ideally for headphones of 150Ω or greater. 8Ω headphones are not recommended.

14 AUX 1 OUTPUT (AND AUX 2 ON EPM)

This output is on a 3-pole 'A' gauge jack and is an impedance-balanced output.

15 AUX 1 PRE/POST SWITCH (AND AUX 2 ON EPM)

This switch globally switches the AUX 1 sends, on all of the input modules, to be either pre-fade or post-fade.



16 FX BUS OUTPUT (EFX only) (Aux 2 on EPM)

This impedance-balanced output carries the signal from the FX bus. It could be used as a second Aux Output if desired, if the FX Processor is not needed at the time. The FX sends on the inputs channels to the FX bus are always post-fade.

17 FOOTSWITCH (EFX only)

Footswitch Input (see diagram on page 20) - Using a single-pole, momentary footswitch inserted into the FOOTSWITCH input the effects processor can be muted/un-muted.

18 STEREO RETURN INPUTS (EFX only)

This pair of inputs accept 3-pole 'A' gauge (TRS) jacks. Use these inputs for sources such as keyboards, drum machines, synths or CDs. The inputs are BALANCED. Mono sources may be used by plugging into the left jack only.

19 STEREO RETURN LEVEL (EFX only)

This control sets the level of signal routed to the main mix busses. There is an associated PK LED to warn of signals which are too high.

Lexicon® FX PROCESSOR OVERVIEW (EFX only)

The effects within the console have been designed with both live sound reinforcement and home recording in mind. Featuring the deep, rich reverb algorithms that Lexicon® are renowned for the effects processor offers increased versatility and high quality effects, all instantly accessible via the extremely intuitive front panel controls. The effects processor has 32 programs which are held in two banks of 16 programs which can be stored to allow you to create your own custom effect settings.

Front panel controls include a **Program Select** knob, **Tempo** and **Store** buttons, and three independent **Parameter** knobs that provide instant access and control over the most critical parameters for the selected effect. The table on page 35 lists the functions of the parameter knobs for each fx program.

Note: When the console is powered up the program recalled will always be the selected program in BANK A.

FX OPERATION

Select and Load a Program

Turn the Program Select knob to choose a program. Note that the console has 32 programs which are held in two banks of 16 programs. There are individual BANK A and BANK B LED's to indicate which bank is currently active. When turning the rotary Program Select knob through 360 degrees (a full rotation) the selected bank will alternate between BANK A and BANK B.

Set Audio Levels

1. Set the gain on the input channel appropriate to the source (vocal microphone, guitar, keyboard, etc.).
2. Set the FX send on the input channel to the 12 o'clock position.
3. Set the EFFECTS Fader on the FX section to the fully down position.
4. Provide source signal (by speaking or singing into the microphone, playing guitar, keyboard, etc.) on the selected channel.
5. Turn up the FX Send level on the channel until the Red CLIP LED in the FX Panel lights only occasionally. If the red Input LED stays lit, too much signal is being sent to the effect processor: reduce FX Send on the input channel.
6. Raise the EFFECTS Fader towards the 0dB position to feed the required level of FX Processor output signal to the mix.
7. To increase or decrease the amount of effect on the signal, adjust the FX Send level on the channel that you want affected.

FX PROCESSOR CONTROLS (EFX only)

1. Tempo Button - Tapping this button twice sets the Delay Time of the selected program. The LED flashes to indicate current tempo. Can be tapped in time with music source to synchronise the delay.

2. Store Button - Stores program modifications to one of the program locations. Press and hold for three seconds will store the preset in the current location. The LED will flash rapidly during the store operation and then stay illuminated for 1 second to show the operation is complete.

3. Pre Delay / Time/ Speed Knob - Controls Pre Delay of the reverbs or the first parameter (time or speed related) of the selected effect. The LED illuminates when the parameter matches the stored setting.

4. Decay / Feedback/Depth Knob - Controls Decay of the reverbs or the second parameter (feedback or depth related) of the selected effect. The LED illuminates when the parameter matches the stored setting.

5. Variation - Controls Liveliness or Diffusion (depending on the reverb selected) or the third parameter of the selected effect. The LED illuminates when the parameter matches the stored setting.

6. Program Select Knob - Navigates through programs, turning to the required program will initiate the loading of the program which take approximately 1 second. The knob can be rotated clockwise or anticlockwise and will alternate between BANK A and BANK B every full rotation. The current bank is shown by its illuminated LED, which flashes if the FX processor is muted. There is a handy aide memoir of the programs printed on the front panel.

7. Clip LED - This LED illuminates when either the incoming audio or the processed audio (within the effect processor) overloads, and causes distortion of the signal.

8. PFL - This switch routes a post effects processor, pre-fade (EFFECTS FADER) signal to the monitor system.

9. MUTE - This switch mutes the output of the FX processor. It doesn't mute the PFL signal or the FX TO AUX 1 pre-fade signal.

10. FX TO AUX 1 - This pot routes a pre-fade and a post-fade signal to the Aux 1-pre and Aux 1-post busses respectively.

11. EFFECTS FADER - This fader controls the level of the signal, from the FX processor, routed to the main mix busses.

FX BUS OUT (see diagram on page 24) - This output carries the signal from the FX bus. It could be used as a second Aux Output if desired, if the FX Processor is not needed at the time.

