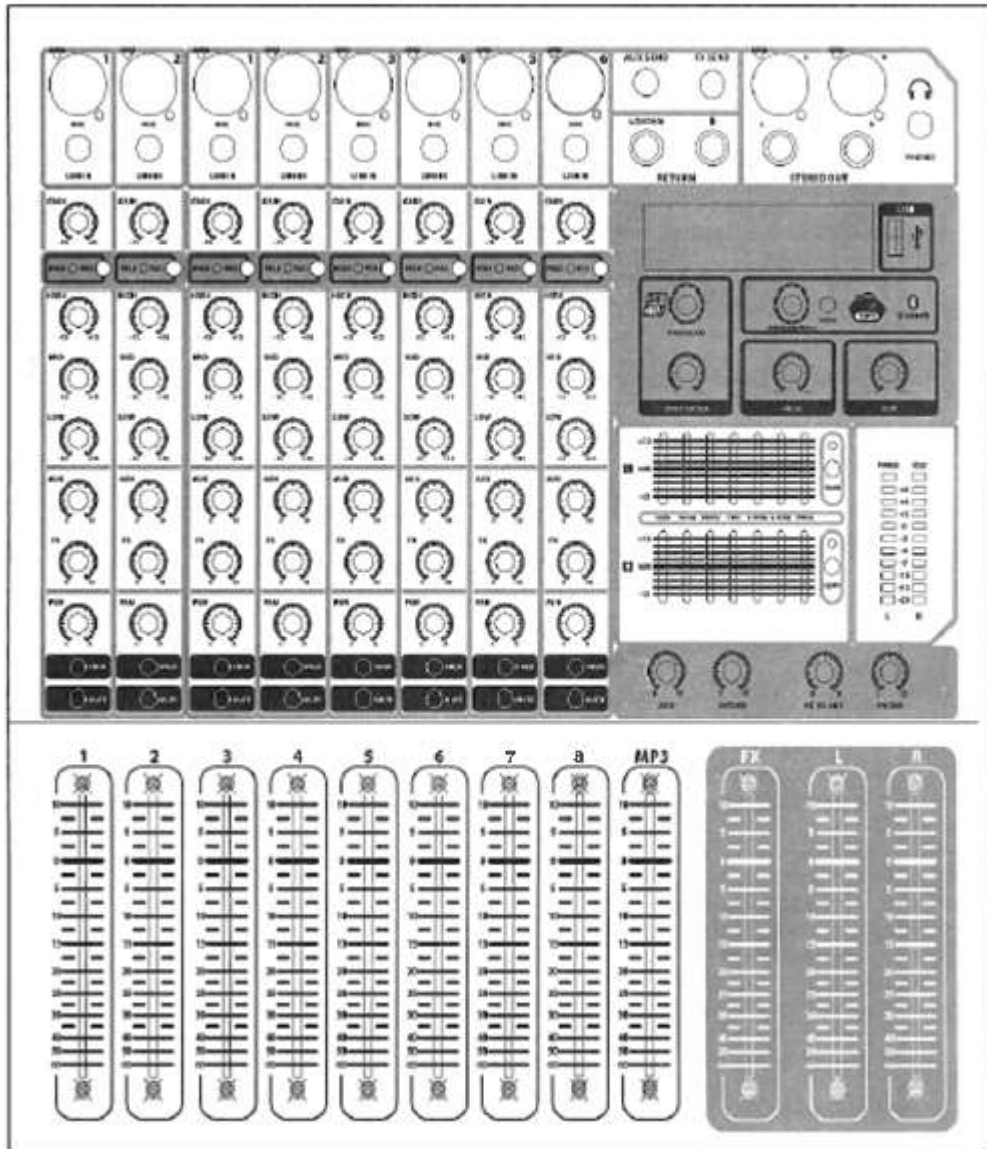


ARISTON AMP25

PROFESSIONAL POWER MIXER

Instruction Manual



INSTRUCTIONS FOR USE

Important safety instructions



Caution !

To reduce the risk of electric shock, do not remove the top cover (or the rear section). No user serviceable parts inside. Refer servicing to qualified personnel

Caution !

To reduce the risk of fire or electric shock, do not expose this appliance to rain and moisture. The apparatus shall not be exposed to dripping or splashing liquids and no objects filled with liquids, such as vases, shall be placed on the apparatus.



This symbol, wherever it appears, alerts you to important operating and maintenance instructions in the accompanying literature. Please read the manual.



This symbol, wherever it appears, alerts you to the presence of uninsulated dangerous voltage inside the enclosure - voltage that may be sufficient to constitute a risk of shock.

Caution !

- [1]. Keep these instructions.
- [2]. Heed all warnings.
- [3]. Follow all instructions.
- [4]. Follow all instructions.
- [5]. Do not use this apparatus near water
- [6]. Clean only with dry cloth.
- [7]. Do not block any ventilation openings. Install in accordance with the manufacturer's instructions.
- [8]. Do not install near any heat sources such as radiators, heat registers, stoves, or other apparatus (including amplifiers) that produce heat.
- [9]. Do not defeat the safety purpose of the polarized or grounding-type plug. A polarized 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.
- [10]. Place the power cord so that it is protected from being walked on and sharp edges. Be sure that the power cord is protected particularly at plugs, convenience receptacles and the point where it exits from the apparatus
- [11]. The apparatus shall be connected to a MAINS socket outlet with a protective earthing connection.

- [12]. Where the MAINS plug or an appliance coupler is used as the disconnect device, the disconnect device shall remain readily operable.



- [13]. Only use attachments/accessories specified by the manufacturer.
- [14]. 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.
- [15]. Unplug this apparatus during lightning storms or when unused for long periods of time.
- [16]. 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 have fallen into the apparatus, the apparatus has been exposed to rain or moisture, does not operate normally, or has been dropped.

Before you get started

Shipment

Your mixing console was carefully packed in the factory to guarantee safe transport. Nevertheless, we recommend that you carefully examine the packaging and its contents for any signs of physical damage, which may have occurred during transit.

- ☞ If the unit is damaged, please do NOT return it to us, but notify your dealer and the shipping company immediately, otherwise claims for damage or replacement may not be granted.

Initial operation

Be sure that there is enough space around the unit for cooling purposes and to avoid over-heating please do not place your mixing console on high-temperature devices such as radiators or power amps. The console is connected to the mains via the supplied cable. The console meets the required safety standards. Blown fuses must only be replaced by fuses of the same type and rating.

- ☞ Please note that all units must be properly grounded. For your own safety, you should never remove any ground connectors from electrical devices or power cables, or render them inoperative.
- ☞ Please ensure that only qualified people install and operate the mixing console. During installation and operation, the user must have sufficient electrical contact to earth, otherwise electrostatic discharges might affect the operation of the unit.

Introduction

Please read through this manual carefully before beginning use, so that you will be able to take full advantage of this mixer's superlative features.

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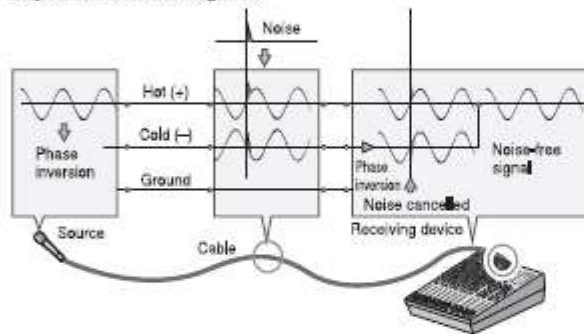
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You've got yourself a mixer and now you're ready to use it. Just plug everything in, twiddle the controls, and away you go ... right? Well, if you've done this before you won't have any problems, but if this is the first time you've ever used a mixer you might want to read through this little tutorial and pick up a few basics that will help you get better performance and make better mixes.

Balanced, Unbalanced—What's the Difference?

In a word: "noise." The whole point of balanced lines is noise rejection, and it's something they're very good at. Any length of wire will act as an antenna to pick up the random electromagnetic radiation we're constantly surrounded by: radio and TV signals as well as spurious electromagnetic noise generated by power lines, motors, electric appliances, computer monitors, and a variety of other sources. The longer the wire, the more noise it is likely to pick up. That's why balanced lines are the best choice for long cable runs. If your "studio" is basically confined to your desktop and all connections are no more than a meter or two in length, then unbalanced lines are fine—unless you're surrounded by extremely high levels of electromagnetic noise. Another place balanced lines are almost always used is in microphone cables. The reason for this is that the output signal from most microphones is very small, so even a tiny amount of noise will be relatively large, and will be amplified to an alarming degree in the mixer's high-gain head amplifier.

Balanced noise cancellation



To summarize

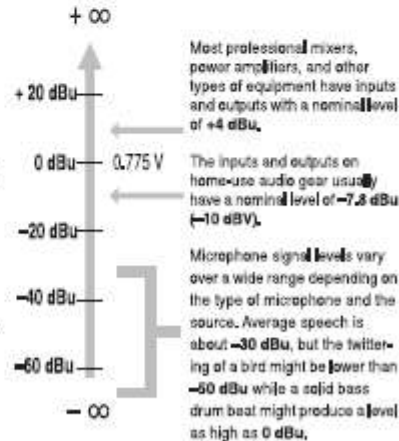
Microphones	Use balanced lines.
Short line-level runs	Unbalanced lines are fine if you're in a relatively noise-free environment.
Long line-level runs	The ambient electromagnetic noise level will be the ultimate deciding factor, but balanced is best.

Signal Levels and the Decibel

Let's take a look at one of the most commonly used units in audio: the decibel (dB). If the smallest sound that can be heard by the human ear is given an arbitrary value of 1, then the loudest sound that can be heard is approximately 1,000,000 (one million) times louder. That's too many digits to deal with for practical calculations, and so the more appropriate "decibel" (dB) unit was created for sound-related measurements. In this system the difference between the softest and loudest sounds that can be heard is 120 dB. This is a non-linear scale, and a difference of 3 dB actually results in a doubling or halving of the loudness.

You might encounter a number of different varieties of the dB: dBu, dBV, dBm and others, but the dBu is the basic decibel unit. In the case of dBu, "0 dBu" is specified as a signal level of 0.775 volts. For example, if a microphone's output level is -40 dBu (0.00775 V), then to raise that level to 0 dBu (0.775 V) in the mixer's preamp stage requires that the signal be amplified by 100 times.

A mixer may be required to handle signals at a wide range of levels, and it is necessary match input and output levels as closely as possible. In most cases the "nominal" level for a mixer's input and outputs is marked on the panel or listed in the owner's manual.



To EQ or Not to EQ

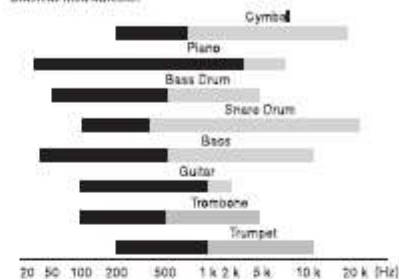
In general, less is better. There are many situations in which you'll need to cut certain frequency ranges, but use boost sparingly, and with caution. Proper use of EQ can eliminate interference between instruments in a mix and give the overall sound better definition. Bad EQ—and most commonly bad boost—just sounds terrible.

Cut for a Cleaner Mix

For example: cymbals have a lot of energy in the mid and low frequency ranges that you don't really perceive as musical sound, but which can interfere with the clarity of other instruments in these ranges. You can basically turn the low EQ on cymbal channels all the way down without changing the way they sound in the mix. You'll hear the difference, however, in the way the mix sounds more "spacious," and instruments in the lower ranges will have better definition. Surprisingly enough, piano also has an incredibly powerful low end that can benefit from a bit of low-frequency roll-off to let other instruments—notably drums and bass—do their jobs more effectively. Naturally you won't want to do this if the piano is playing solo.

The reverse applies to kick drums and bass guitars: you can often roll off the high end to create more space in the mix without compromising the character of the instruments. You'll have to use your ears, though, because each instrument is different and sometimes you'll want the "snap" of a bass guitar, for example, to come through.

The fundamental and harmonic frequency ranges of some musical instruments.



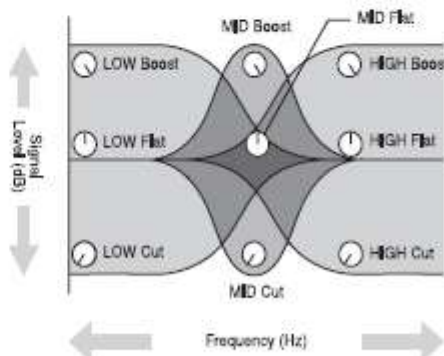
- **Fundamental:** The frequency that determines the basic musical pitch.
- **Harmonics:** Multiples of the fundamental frequency that play a role in determining the timbre of the instrument.

Some Frequency Facts

The lowest and highest frequencies that can be heard by the human ear are generally considered to be around 20 Hz and 20,000 Hz, respectively. Average conversation occurs in the range from about 300 Hz to about 3,000 Hz. The frequency of a standard pitchfork used to tune guitars and other instruments is 440 Hz (this corresponds to the "A3" key on a piano tuned to concert pitch). Double this frequency to 880 Hz and you have a pitch one octave higher (i.e. "A4" on the piano keyboard). In the same way you can halve the frequency to 220 Hz to produce "A2" an octave lower.

Boost with Caution

If you're trying to create special or unusual effects, go ahead and boost away as much as you like. But if you're just trying to achieve a good-sounding mix, boost only in very small increments. A tiny boost in the midrange can give vocals more presence, or a touch of high boost can give certain instruments more "air." Listen, and if things don't sound clear and clean try using cut to remove frequencies that are cluttering up the mix rather than trying to boost the mix into clarity. One of the biggest problems with too much boost is that it adds gain to the signal, increasing noise and potentially overloading the subsequent circuitry.



Ambience

Your mixes can be further refined by adding ambience effects such as reverb or delay. The GS-12FX's internal effects can be used to add reverb or delay to individual channels in the same way as external effects processors. (Refer to page 15).

Reverb and Delay Time

The optimum reverb time for a piece of music will depend on the music's tempo and density, but as a general rule longer reverb times are good for ballads, while shorter reverb times are more suited to up-tempo tunes. Delay times can be adjusted to create a wide variety of "grooves". When adding delay to a vocal, for example, try setting the delay time to dotted eighth notes corresponding to the tune's tempo.

Reverb Tone

Different reverb programs will have different "reverb tone" due to differences in the reverb time of the high or low frequencies. Too much reverb, particularly in the high frequencies, can result in unnatural sound and interfere with the high frequencies in other parts of the mix. It's always a good idea to choose a reverb program that gives you the depth you want without detracting from the clarity of the mix.

Reverb Level

It's amazing how quickly your ears can lose perspective and fool you into believing that a totally washed-out mix sounds perfectly fine. To avoid falling into this trap start with reverb level all the way down, then gradually bring the reverb into the mix until you can just hear the difference. Any more than this normally becomes a "special effect."

The Modulation Effects:

Phasing, Chorus, and Flanging

All of these effects work on basically the same principle: a portion of the audio signal is "time-shifted" and then mixed back with the direct signal. The amount of time shift is controlled, or "modulated", by an LFO (Low-frequency Oscillator).

For phasing effects the shift is very small. The phase difference between the modulated and direct signals causes cancellation at some frequencies and reinforces the signal at others and this causes the shimmering sound we hear.

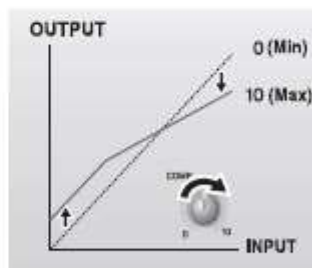
For chorus and flanging the signal is delayed by several milliseconds, with the delay time modulated by an LFO, and recombined with the direct signal. In addition to the phasing effect described above, the delay modulation causes a perceived pitch shift which, when mixed with the direct signal, results in a harmonically rich swirling or swishing sound.

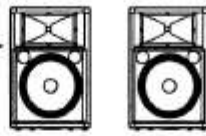
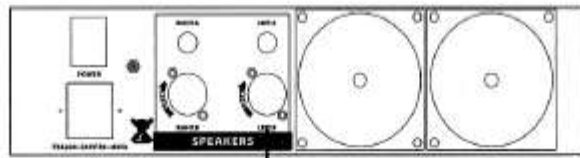
The difference between chorus and flanging effects is primarily in the amount of delay time and feedback used—flanging uses longer delay times than chorus, whereas chorus generally uses a more complex delay structure. Chorus is most often used to thicken the sound of an instrument, while flanging is usually used as an outright "special effect" to produce otherworldly sonic swoops.

Compression

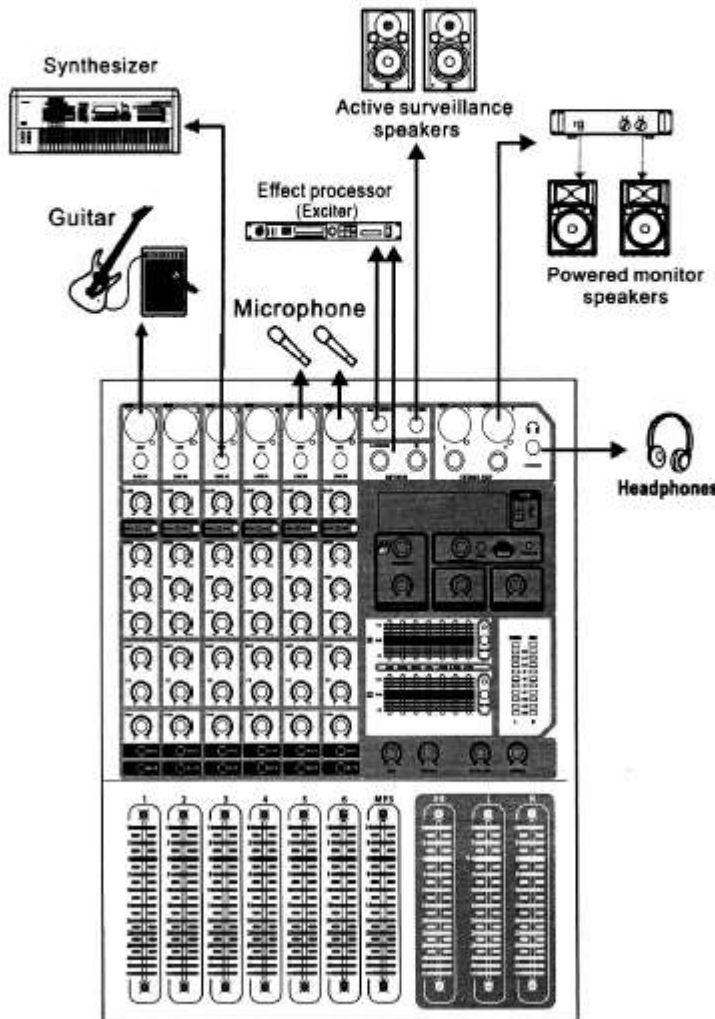
One form of compression known as "limiting" can, when properly used, produce a smooth, unified sound with no excessive peaks or distortion. A common example of the use of compression is to "tame" a vocal that has a wide dynamic range in order to tighten up the mix. With the right amount of compression you'll be able to clearly hear whispered passages while passionate shouts are still well balanced in the mix. Compression can also be valuable on bass guitar. Too much compression can be a cause of feedback, however, so use it sparingly.

Most compressors require several critical parameters to be set properly to achieve the desired sound. The MG compressor makes achieving great sound much easier: all you need to do is set a single "compression" control and all of the pertinent parameters are automatically adjusted for you.

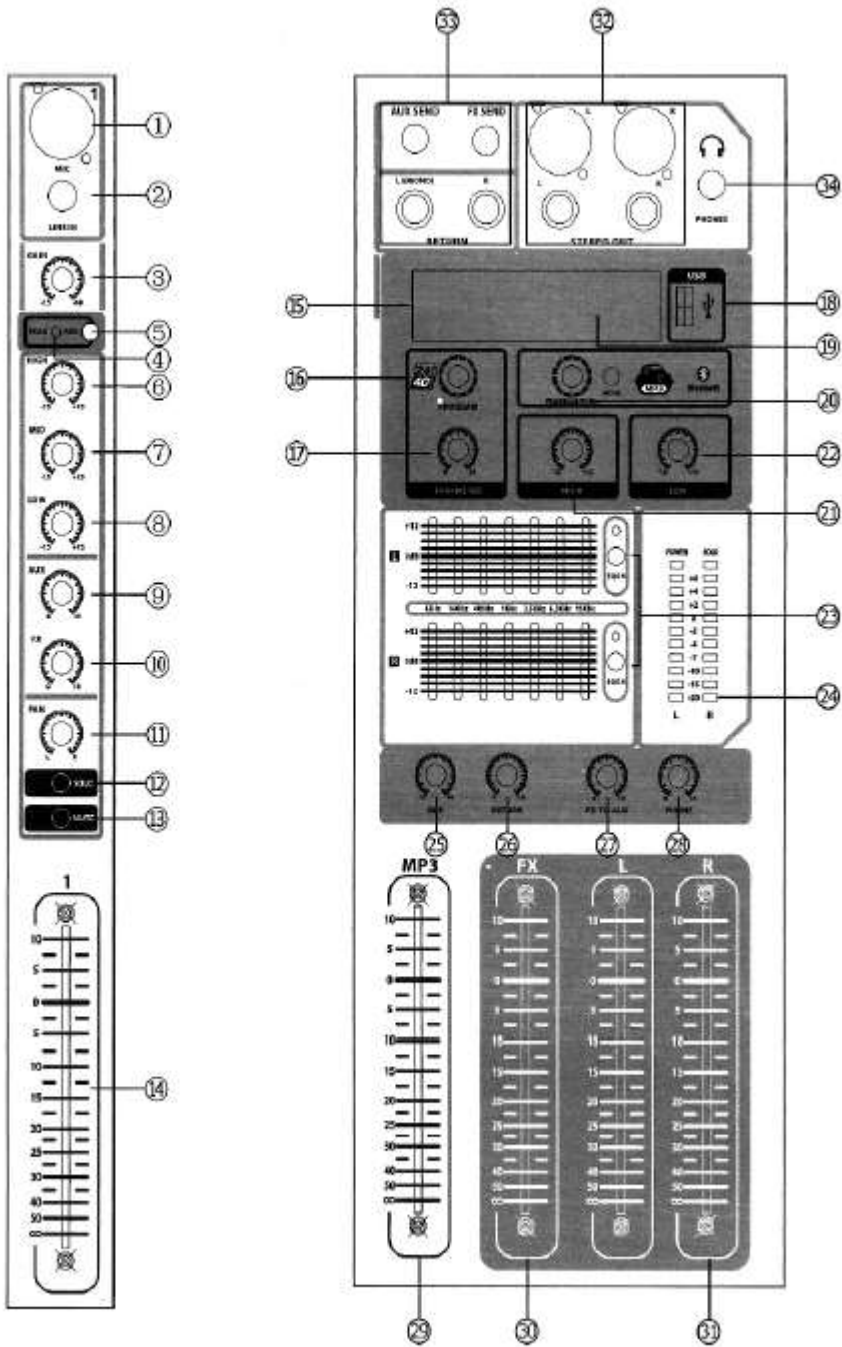




speakers



INSTRUCTIONS FOR USE



1. BALANCE INPUT

Electronically Balanced inputs acceptable a standard XLR male Connector.

2. LINE INPUT

The unbalanced Mic input is provided for the use of a unbalance mic and is designed to accept a unbalanced high impedance input signal.
(This use for connection Deck, Turntable, Keyboard etc..)

3. PEAK (PEAK LEVEL INDICATOR)

A red LED indicates a signal level at the insert return point, premaster Fader, It illuminates at approximately 5dB below clipping.

4. GAIN CONTROL

Adjusts input sensitivity from -60dB to -20dB with the -20dB pad switch in the out position, and -40dB to 0dB when the -20dB pad switch is pushed.

5. PAD (-20dB)

When push this switch, attenuates the input signal -20dB.

6. HIGH

Control the high frequency tone of each channel. Always set this control to the 12 o' clock position, but you can control the high frequency tone according to the speaker, the conditions of listening position and listener's taste. Clockwise rotation of the control increases level.

7. MID

Control the mid frequency tone of each channel. Always set this control to the 12 o' clock position, but you can control the mid frequency tone according to the speaker, the conditions of listening position and listener's taste. Clockwise rotation of the control increases level.

8. LOW

Control the low frequency tone of each channel. Always set this control to the 12 o'clock position, but you can control the low frequency tone according to the speaker, the conditions of listening position and listener's state. Clockwise rotation of the control increase the level.

9. AUX

Use this control to set the level of signal from external stereo source and the main signal control is recontrolled by STEREO or MONO section.

10. EFFECT

Use this control when you want to get effect sound by adjustment of input signal. When you don't use external source, digital delay will be working which installed inside.

11. PAN

The pan control sends continuously variable amounts of the post fader signal to either the left or right main busses. In the center position equal amounts of signal are sent to the left and right busses

12. PFL

You can monitor the signal of the only channel which PFL switch is turned "ON" by headphone. When PFL switch turned on, other channels will cut off automatically.

13. MUTE

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

14. CHANNEL FADER

This is the function to adjust the volume of signal connecting into each channel and adjust the volume of output, together with master fader. Normal operating position is at the "0" mark, providing 4dB of gain above that point, if required.

15. EFFECTOR DISPLAY

Show the kind of effector.

16. PROGRAM Dial

You can select the effect preset by turning the PROGRAM control. The display flashes with the number of the current preset. To recall the selected preset, press on the button; the flashing stops. You can also recall the selected preset with the foot switch.

17. EFFECT DELAY

This is used for adjusting the level of echo delay. You can achieve the delay echo by the rotation button.

18. MP3 player jack

USB: can be play through U-DISK

19. MP3 PLAY window

Show the Mp3 playing time, song name and other play instruction.

20. Mp3 player

21. adjust MP3 volume to higher with bluetooth function

22. adjust MP3 volume to lower

23. STEREO GRAPHIC EQUALIZER

2X7-band equalizer is provided for tone control over each frequency, and for precise high quality sound by final tone control.

24. Level Meter

Show the level signal's strength

NOTE: The "0" segment corresponds to the nominal output level. The PEAK indicator lights red when the output reaches the clipping level.

25. STEREO AUX RETURN1 Control

Adjusts the level at which the signal received at the RETURN jacks (L (MONO) and R) is sent to the STEREO L/R bus.

26. AUX SEND Control

Use this fader can control the AUX output jack

27. EFX SEND

When you use STEREO board, you can adjust the sound olume of all kinds of effects outside.

28. PHONES/CTRL ROOM ONLY Control

Use this control to adjust the control room output level and the headphones volume.

29. MP3 VOL Fader

Change VOL button can be control the VOL of Mp3.

30. FX SEND Fader

Control effect input signal level.

31. MAIN MIX FADER

You use the high-precision quality faders to control the output level of the main mix.

32. OUTPUT JACK (LEFT/RIGHT)

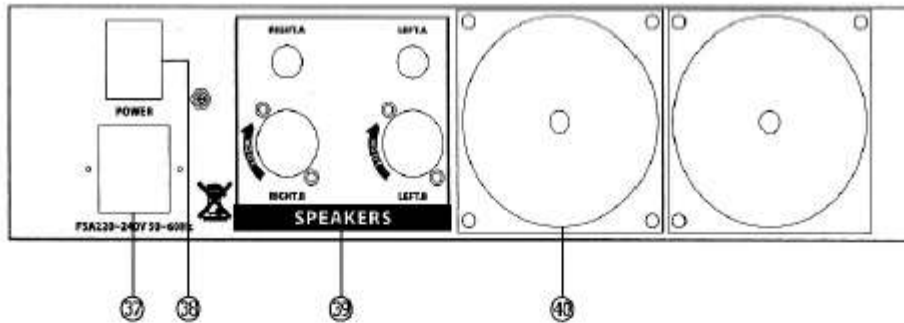
In this product, the final confirmed sound can be sent to main amplifier through XLR & 1/4 jack.

33. EFX SEND, RETURN

This can be used to connect all kinds of effects from outside.

34. HEADPHONE JACK

You can monitor working condition by sound through the headphone.



37. FUSE HOLDER/IEC MAINS RECEPTACLE

The console is connected to the mains via the cable supplied, which meets the required safety standards. Blown fuses must only be replaced by fuses of the same type and rating. The mains connection is made via a cable with IEC mains connector. An appropriate mains cable is supplied with the equipment.

38. POWER Switch

Use the POWER switch to turn on the mixing console. The POWER switch should always be in the "Off" position when you are about to connect your unit to the mains. To disconnect the unit from the mains, pull out the main cord plug. When installing the product, ensure that the plug is easily accessible.

39. AMPLIFIER OUTPUT

Can be contact with speaker to here

Minimum impedance 4 ohm per channel. Connecting two 4 ohm speakers per channel will overload the amp and void the warranty. Always use 8 ohm speakers or higher if connecting two speakers per channel.

40. COOLING FAX

Cooling the amplifier to avoid the amplifier too hot to be broken.

INSTALLATION

Rack mounting

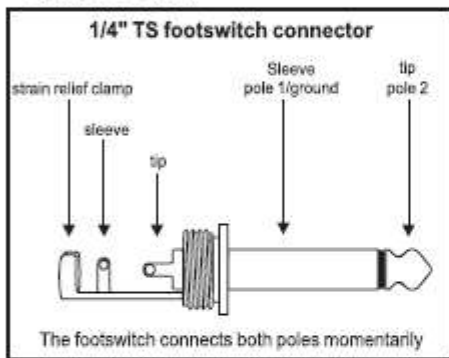
The packaging of your mixing console contains two 19" rack mounts for installation on the side panels of the console.

Before you can attach the rack mounts to the mixing console, you need to remove the screws holding the left and right side panels. Then, use these screws to fasten the two rack mounts, each specifically to one side. With the rack mounts installed, you can mount the mixing console in a commercially available 19" rack. Be sure to allow for proper air flow around the unit, and do not place the mixing console close to radiators or power amps, so as to avoid overheating.

- Only use the screws holding the mixing console side panels to fasten the 19" rack mounts.

Cable connections

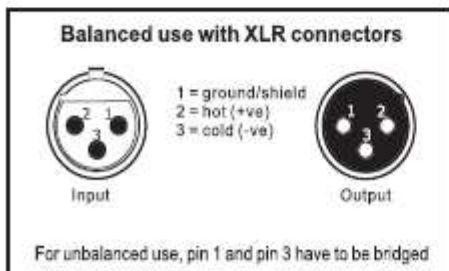
You will need a large number of cables for the various connections of the console. The illustrations below show the wiring of these cables. Be sure to use only high-grade cables.



Foot switch connector

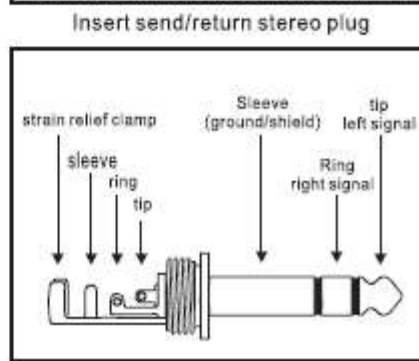
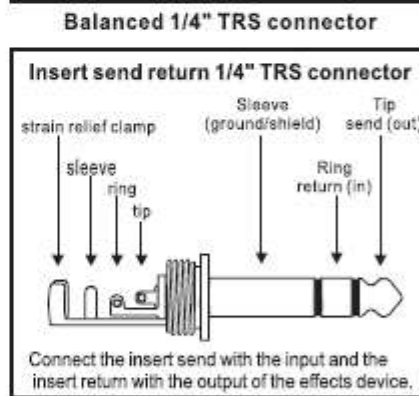
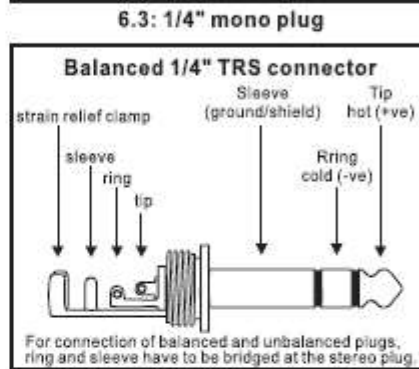
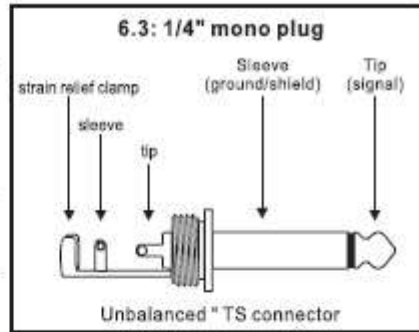
Audio connections

Please use commercial RCA cables to wire the 2-track inputs and outputs. You can, of course, also connect unbalanced devices to the balanced input/outputs. Use either mono plugs, or use stereo plugs to link the ring and shaft (or pins 1 & 3 in the case of XLR connectors).



XLR connections

- Caution! You must never use unbalanced XLR connectors (PIN 1 and 3 connected) at the MIC input jacks if you want to use the phantom power supply.



Specifications

MICROPHONE INPUTS (XENYX MIC PREAMP)

Type	XLR, electronically balanced, discrete input circuit
Mic E.I.N. (20 Hz - 20 kHz)	
@ 0 Ω source resistance	-134 dB / 135.7 dB A-weighted
@ 50 Ω source resistance	-131 dB / 133.3 dB A-weighted
@ 150 Ω source resistance	-129 dB / 130.5 dB A-weighted
Frequency response	<10 Hz - 150 kHz (-1 dB), <10 Hz - 200 kHz (-3 dB)
Gain range	+10 to +60 dB
Max. input level	+12 dBu @ +10 dB Gain
Impedance	approx. 2.6 k Ω balanced
Signal-to-noise ratio	110 dB / 112 dB A-weighted (0 dBu in @ +22 dB gain)
Distortion (THD+N)	0.005% / 0.004% A-weighted
Line input	
Type	1/4" TRS connector electronically balanced
Impedance	approx. 20 k Ω balanced 10 k Ω unbalanced
Gain range	-10 to +40 dB
Max. input level	30 dBu

FADE-OUT ATTENUATION¹ (CROSSTALK ATTENUATION)

Main fader closed	90 dB
Channel muted	89 dB
Channel fader closed	89 dB

FREQUENCY RESPONSE

Microphone input to main out	
<10 Hz - 90 kHz	+0 dB / -1 dB
<10 Hz - 160 kHz	+0 dB / +3 dB

Stereo inputs	
Type	1/4" TRS connector, electronically balanced
Impedance	approx. 20 k Ω
Max. input level	+22 dBu

EQ mono channels	
Low	80 Hz / ± 15 dB
Mid	100 Hz - 8 kHz / ± 15 dB
High	12 kHz / ± 15 dB

EQ stereo channels	
Low	80 Hz / ± 15 dB
Low Mid	500 Hz / ± 15 dB
High Mid	3 kHz / ± 15 dB
High	12 kHz / ± 15 dB

Aux sends	
Type	1/4" TS connector, unbalanced
Impedance	approx. 120 Ω
Max. output level	+22 dBu

Stereo aux returns	
Type	1/4" TRS connector, electronically balanced
Impedance	approx. 20 k Ω bal. / 10 k Ω unbal.
Max. input level	+22 dBu

Main outputs	
Type	XLR, electronically balanced and 1/4" TRS balanced
1622FX only:	1/4" TS connector unbalanced
Impedance	approx. 240 Ω symm. / 120 Ω unbalanced
Max. output level	+28 dBu +22 dBu

Control room outputs	
Type	1/4" TS connector unbalanced
Impedance	approx. 120 Ω
Max. output level	+22 dBu

Headphones outputs	
Type	1/4" TRS connector, unbalanced
Max. output level	+19 dBu / 150 Ω (+25 dBm)

DSP

Converter	24-bit Sigma-Delta, 64/128-times oversampling
Sampling rate	40 kHz

MAIN MIX SYSTEM DATA²

Noise	
Main mix @ - ∞ Channel fader @ - ∞	-101 dB -100 dB
Main mix @ 0 dB, Channel fader @ - ∞	-93 dB -96 dB -87 dB
Main mix @ 0 dB, Channel fader @ 0 dB	-81 dB -83 dB -80 dB

Power supply	
Mains voltage	230 V~, 50/60 Hz

Power consumption 4 ohm					
CH8. 300W	<input type="checkbox"/>	CH8. 450W	<input type="checkbox"/>	CH8. 650W	<input type="checkbox"/>
CH12. 300W	<input type="checkbox"/>	CH12. 450W	<input type="checkbox"/>	CH12. 650W	<input type="checkbox"/>
CH16. 300W	<input type="checkbox"/>	CH16. 450W	<input type="checkbox"/>	CH16. 650W	<input type="checkbox"/>

Fuse	230 V ~: T 12 A H 250V
Mains connection	Standard IEC receptacle

- Measuring conditions:
- 1 kHz @ 0 dBu; 20 Hz - 20 kHz; line input, main output, unity gain.
 - 20 Hz - 20 kHz; measured at main output. Channel 1 - 4 unity gain; 10 Hz; all channels on main mix; channel 1/3 as far left as possible; channel 2/4 as far right as possible. Reference = +6 dBu.

Troubleshooting

Power doesn't come on.	<ul style="list-style-type: none"> • Is the power line properly plugged into an AC wall outlet? • Are the power line and AC wall outlet connected correctly?
No sound	<ul style="list-style-type: none"> • Are microphone, external devices, and speakers connected correctly? • Are the channel GAIN controls, channel fader, STEREO OUT Master fader and GROUP fader set to appropriate levels? • Are the speaker cables connected properly, or are they shorted? • If the above checks do not identify the problem, please contact the service center.
Sound is faint, distorted, or noisy	<ul style="list-style-type: none"> • Are the channel GAIN controls, channel fader, STEREO OUT Master fader and GROUP fader set to appropriate levels? • Are two different instruments connected to the XLR-type and phone jacks, or to the phone and RCA pin jacks on one channel? Please connect to only one of these jacks on each channel. • Is the input signal from the connected device set to an appropriate level? • Are you applying the effects at an appropriate level? • Are microphone connected to the MIC input jacks? • If you are using condenser microphone, is the PHANTOM +48V switch turned on?
No effect is applied	<ul style="list-style-type: none"> • Check that the EFFECT control on each channel is correctly adjusted. • Be sure that the FX control and EFFECT fader are correctly adjusted.
I want spoken words to be heard more clearly.	<ul style="list-style-type: none"> • Adjust the equalizers on each channel.
I want to output a monitor signal through speakers.	<ul style="list-style-type: none"> • Connect a powered speaker to the AUX jack, or to the AUX1 or 2 jack and turn the PRE switch on each channel on. Then adjust the output signal by using the AUX controls on each channel.
The level meter doesn't show the output signal level.	<ul style="list-style-type: none"> • Are the PEL switches for the channels that you are not using turned on?

ARISTON AMP25

