



# Toft Audio Designs®



# ATB

Series™ Console Manual



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## INTRODUCTION

The Toft Audio Series ATB is the creation of Malcolm Toft, the founder of console manufacturer Trident Audio Developments in 1972. Prior to that, Malcolm was a recording engineer. His credits include the David Bowie 'Space Oddity' album, three albums with T-Rex and James Taylor's first album. He was also involved with the mixing of the Beatles 'Hey Jude' single at Trident Recording Studios in 1968. Trident Audio Developments became an acclaimed manufacturer of recording consoles. Artists who have recorded hits on a Trident console include: Stevie Wonder, Herbie Hancock, Rod Stewart, Queen, David Bowie and Elton John. More recently, Dire Straits, Coldplay and Oasis have added their name to the list. Radiohead has in fact, recorded all of their albums from OK Computer onwards on an MTA console. Malcolm sold Trident in 1988 but continued to make consoles under the MTA (Malcolm Toft Associates) banner until just a few years ago. It is this pedigree that is continued with his new range under the Toft Audio Designs marque.

The Series ATB is an extremely versatile and sophisticated 8 bus console in a compact frame. It has been designed with today's DAW user in mind and offers extremely high audio quality and features never before found in a console in its price range. These features include an optional digital I/O card which provides the eight subgroup sends and returns as either ADAT optical or firewire and the master left/right outputs as firewire or S/PDIF. ASIO drivers are provided as well as MIDI connectivity to complete the versatility of this option.

In a multi-track studio environment the console will perform both tracking and remix functions with consummate ease. Because of the channel monitor and direct output facilities, recording instruments to tape is simple and of very high quality with little outboard equipment being required. When mixing down, the large number of inputs to the stereo bus (56 on the 16 channel, 72 on the 24 channel and 88 on the 32 channel) makes the job very easy. The large number of auxiliary sends (six) also adds flexibility.

The Series ATB has been designed for easy servicing. The console is constructed from individual circuit boards mounted at right angles to an 8 way front panel. It is therefore only a matter of minutes to remove any channel board from its front panel should a component need replacing. All active circuitry (IC.'s etc) are socketed, again to aid servicing. All passive components (resistors, capacitors etc.) are of conventional types; not surface mount which are difficult to remove and replace. The modules connect to each other via ribbon cables which ensure maximum reliability and freedom from wiring errors.

Connections to the console are via the rear panel where the cables can be conveniently hidden from view, which is again a feature normally only found in consoles costing considerably more. The console is fitted with extremely accurate 12 segment bar graph displays on the groups and master left/right outputs and high quality illuminated VU meters operate in parallel with the master left/right bar graphs.

By carefully following the installation procedures detailed in the following pages, your Series ATB console will give years of reliable service and professional audio quality.



## POWER & AUDIO CONNECTIONS

### D.C. POWER:

The Series ATB is supplied with a high quality D.C. power supply which should be connected to a suitable A.C. power source via an earthed cable connected to the 3 pin euro socket at the back of the power supply.

The supply has been fitted with the highest quality 'toroidal' type transformer in order to minimise stray hum radiation, but should still be located as far away from the console as possible to avoid any possibility of hum pick-up.

Before connecting to a power socket, make sure that the correct voltage is selected on the rear of the power supply, either 115 or 230 volts.

Make sure that the supply is placed in a well ventilated area, with at least one rack space between the supply and other units.

Low voltage D.C is supplied to the console via a circular locking connector. With the power supply switched off, this should be pushed firmly into the mating socket on the rear of the console located behind the Master Module. A locking ring insures that the cable cannot be accidentally pulled from the console.

The D.C power connections are as follows:

Pin 1	+17V
Pin 2	Chassis Ground
Pin 3	Electronics Ground
Pin 4	+48V
Pin 5	-17V

Once the power supply has been connected to the console, the correct voltage selected and the A.C. power applied, the power supply can be switched on. The power supply is fitted with three red LED's which indicate that the +17, -17 and +48 voltages are functioning. If for any reason any of the LED's fail to illuminate, do not operate the console. Turn the power supply off immediately and check that the fuses (located inside the unit) are not blown. If any fuse has blown, replace with one of the same rating and try again. If the fuse has not blown or a replacement subsequently fails, seek qualified help.



**AUDIO CONNECTIONS:**

Two types of audio connector are provided on the rear of the unit, XLR and 1/4" jack.

The mic input XLR is female and therefore the microphone lead requires a male plug. The wiring convention is as follows:

<b>Pin 1</b>	<b>Earth</b>
<b>Pin 2</b>	<b>+ (positive phase)</b>
<b>Pin 3</b>	<b>- (negative phase)</b>

Various connections to the console use balanced (stereo) 1/4" jacks and are wired as follows:

<b>Sleeve</b>	<b>Earth</b>
<b>Tip</b>	<b>+ (positive phase)</b>
<b>Ring</b>	<b>- (negative phase)</b>

The balanced connections are: Channel 'LINE' Input, Channel 'MON', 'MONITOR RETURNS', 'SUBMASTER OUTPUTS', 'MASTER O/P' and 'AUX MASTERS'.

The 'STEREO FX RETURNS' and '2 TRACK RET.' 1 and 2 utilise a balanced (stereo) jack wired in the following manner:

<b>Sleeve</b>	<b>Earth</b>
<b>Tip</b>	<b>Right (unbalanced)</b>
<b>Ring</b>	<b>Left (unbalanced)</b>

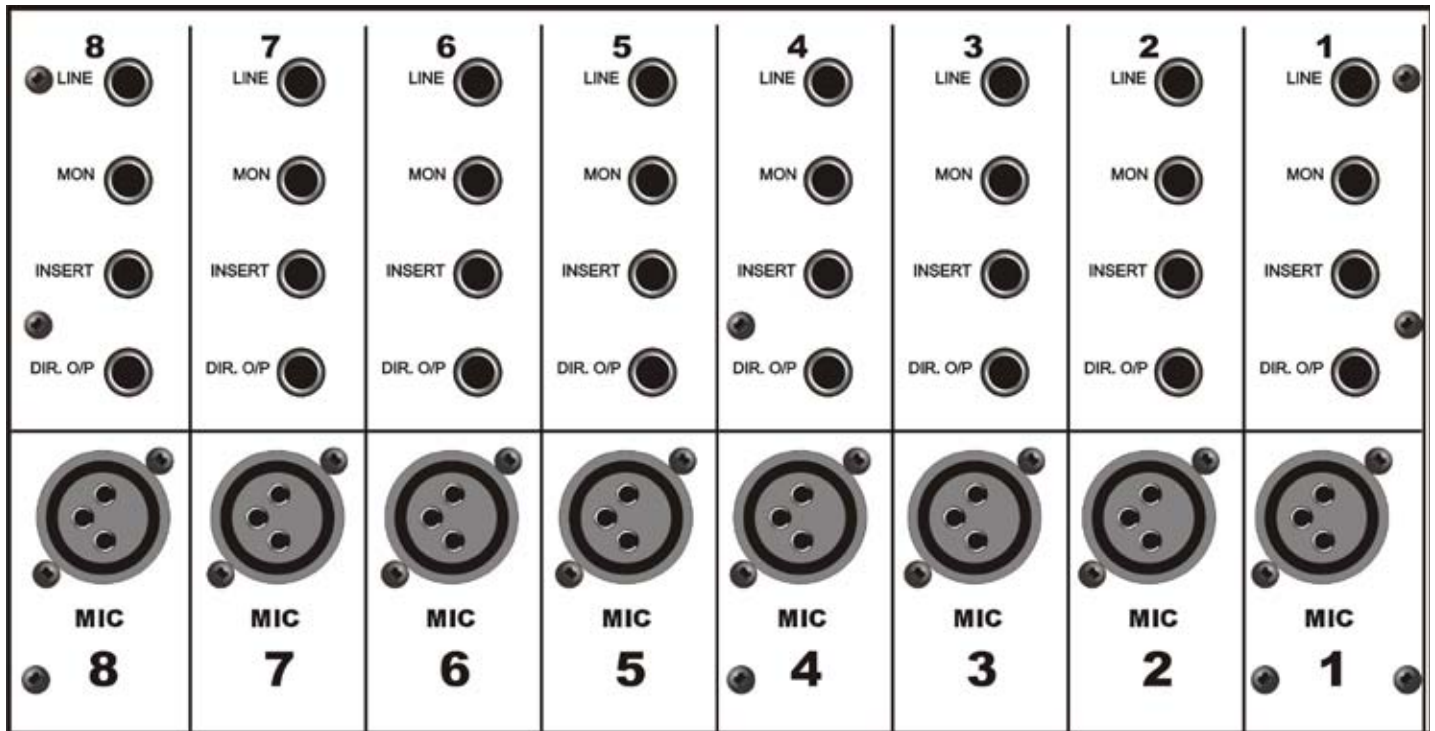
Various connections to the console are unbalanced (mono) 1/4" jacks and are wired as follows:

<b>Sleeve</b>	<b>Earth</b>
<b>Tip</b>	<b>+ (positive phase)</b>

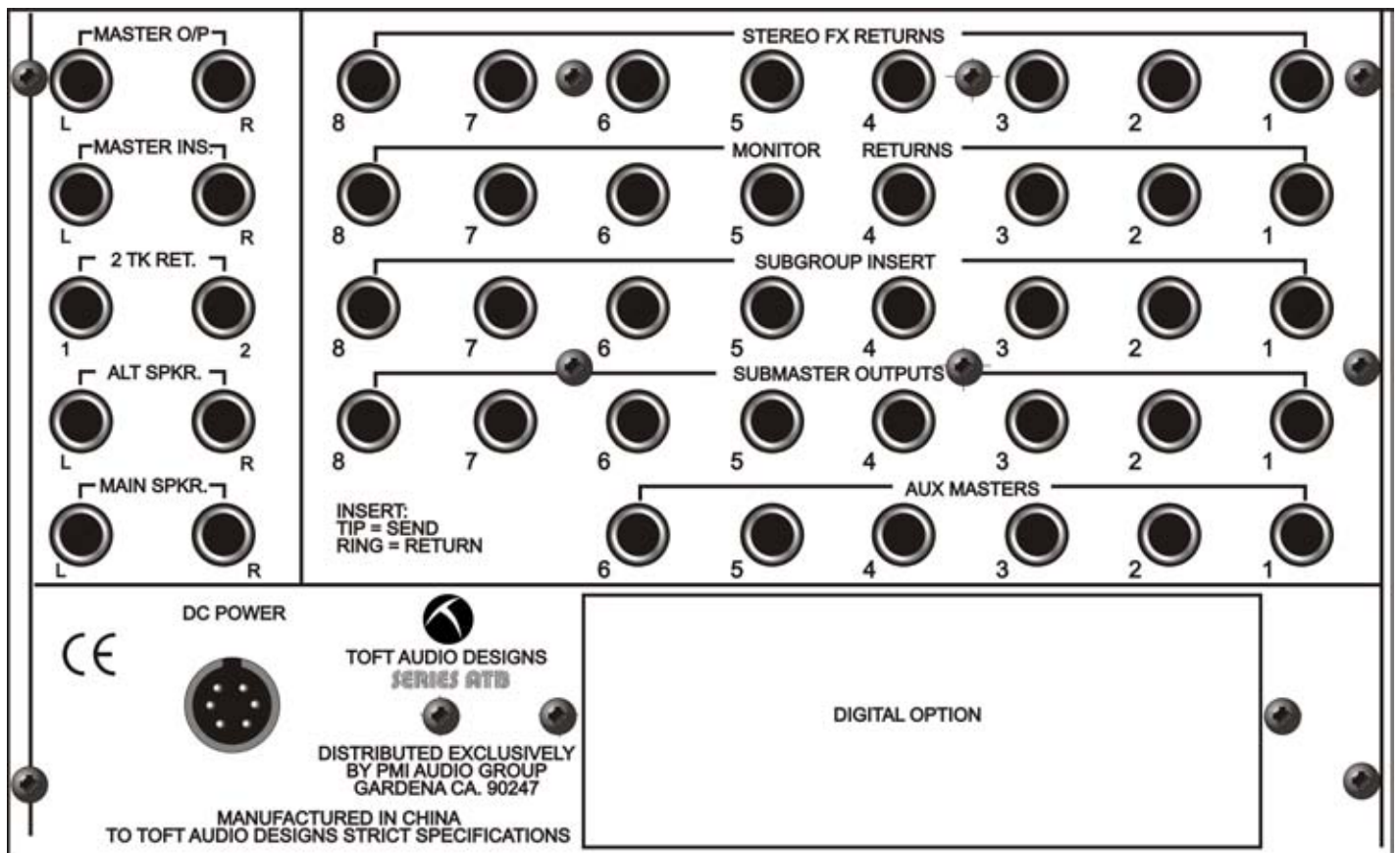
The unbalanced connections are: Channel 'DIR O/P', 'ALT SPKR', 'MAIN SPKR'.

The Channel 'INSERT' Sends, 'SUBGROUP INSERT' Sends and 'MASTER INS.' Sends Utilise 1/4" stereo jacks wired in the following manner:

<b>Sleeve</b>	<b>Earth</b>
<b>Tip</b>	<b>Send</b>
<b>Ring</b>	<b>Return</b>



SERIES ATB REAR PANEL - INPUT SECTION



SERIES ATB REAR PANEL - OUTPUT SECTION



## STAR GROUNDING

By connecting equipment together using the 'star grounding' principle it is possible to virtually eliminate hum loops from studio wiring.

Since the system is quite easy to implement, it only requires following a certain procedure that will soon become second nature. A few moments spent studying this chapter and following the steps listed below, could pay dividends in the future.

Step 1: Choose a particular piece of equipment that will be made 'technical' earth (in most cases the mixing console is the ideal choice as it is usually the item that everything else is connected to).

Step 2: On all equipment that is connected to the console inputs, connect the earth (sleeve) of the cable carrying the signal at the console end but not the end that connects to the equipment. For example, if it is a jack lead, connect the sleeve at the mixer end but not at the equipment end. The only exception to this should be a microphone as it cannot get an electrical earth by any other means. The earth should therefore be connected at both ends of a microphone cable.

Step 3: On all equipment that is connected to the console outputs, connect the earth (sleeve) at the console end but not at the equipment end (input).

Step 4: Connect the chassis or mains earth of all equipment to the same point that the console chassis and power supply earth are connected to. This point is then known as the 'technical' earth and should be as good as it possibly can be. In many cases this will be the earth of the mains socket feeding the console, but sometimes better results and freedom from refrigerator clicks and pops etc. can be achieved by making an independent 'technical' earth. In many professional recording studios this is done by burying a large copper plate at least three feet underground and connecting the 'technical' earth point to it. A less drastic measure is to use the copper pipe of a radiator as the earth reference, as this often runs underground.

It is essential that if a separate technical earth is generated, this is always used as the earth point for all equipment. Connecting some equipment to mains earth and some to 'technical' earth could result in a shock hazard as it is sometimes possible for a quite high A.C. voltage difference to be generated across the two earth points.





## INPUT MODULE



### INPUT MODULE General Description

The Series ATB Input Module is an extremely sophisticated and flexible channel that can be used for recording, mixing and monitoring multi-track recordings. It will accept the signal from a low impedance balanced microphone, high or low level balanced line input, or even the output from a musical instrument.

The signal can be shaped via the extremely musical four band equaliser section that allows precise control over the audio spectrum. Boost and cut of 15dB is provided by centre detent controls and the two sweep frequency ranges overlap to increase the flexibility of the equaliser section. An 80Hz (high pass) filter is also incorporated which can be switched in or out of circuit. The equaliser is extremely useful in both recording and mixing modes. The equaliser can also be switched from the input channel to the separate monitor channel incorporated into each input module.

Six auxiliary sends are provided. Aux 1 is permanently pre-fade while auxes 3-6 can be switched pre or post the channel fader. Auxes 5-6 can also be selected between the channel or monitor path, greatly adding to the flexibility of the module.

Each module incorporates a monitor section which is used to monitor a replay channel from a multi-track recording device. This is in effect a separate signal path which ultimately feeds the master remix outputs. By making the equaliser and auxiliary sends 5-6 available to this path, this becomes an extremely versatile feature of the console. When monitoring the output of a multi-track recorder or DAW, it is possible to create a monitor mix with equalisation and reverb effects completely independent from the channel. These effects would also remain when the recording is played back through the same monitor section.

When the console is used for mixing purposes, the monitor section functions as an additional line input return to the stereo mix busses, again with the ability to route the equalisation and auxiliary sends 5-6 through the monitor section. This effectively doubles the number of inputs available on remix. As there are also eight dedicated stereo effects returns and 8 monitor returns on the sub group section, at mixdown a 16 channel Series ATB becomes a 56 input mixer whilst a 24 channel becomes a 72 input mixer and a 32 channel becomes an 88 input mixer.



The input channel is provided with a stereo, non destructive after fade listen system (AFL). By depressing this button, it is possible to hear the selected signal in isolation (or mixed with other selected AFL signals). This signal follows the position of the channel pan control so that the signal can be monitored with the stereo perspective intact. As it is a monitor function only, the AFL can be depressed when recording or mixing without it harming the signal path. It is however, possible to change the function of the switch on a channel by channel basis so that the signal becomes a mono, pre-fade source (PFL). This is useful if the console is used in a live sound application where a simple check to see if a channel is 'live' is required without having to push up the fader. By removing the base panel of the mixer and accessing the individual channel PCB'S, it will be possible to see a push button switch mounted approximately halfway down the circuit board. When the button is depressed (factory default setting) the signal is in AFL mode. If the button is not depressed, the signal from that channel will become PFL.

Centre detent pan controls are provided for both the channel and monitor sections for accurate placement of signals in the stereo spectrum. The channel path is provided with a green LED which illuminates when the signal reaches -20dB at the channel output and a red LED which illuminates when the signal reaches +10dB at the channel output. These are extremely useful features and provide a constant indication that a signal is present in the module (green LED), and that signal peak is being reached (red LED).

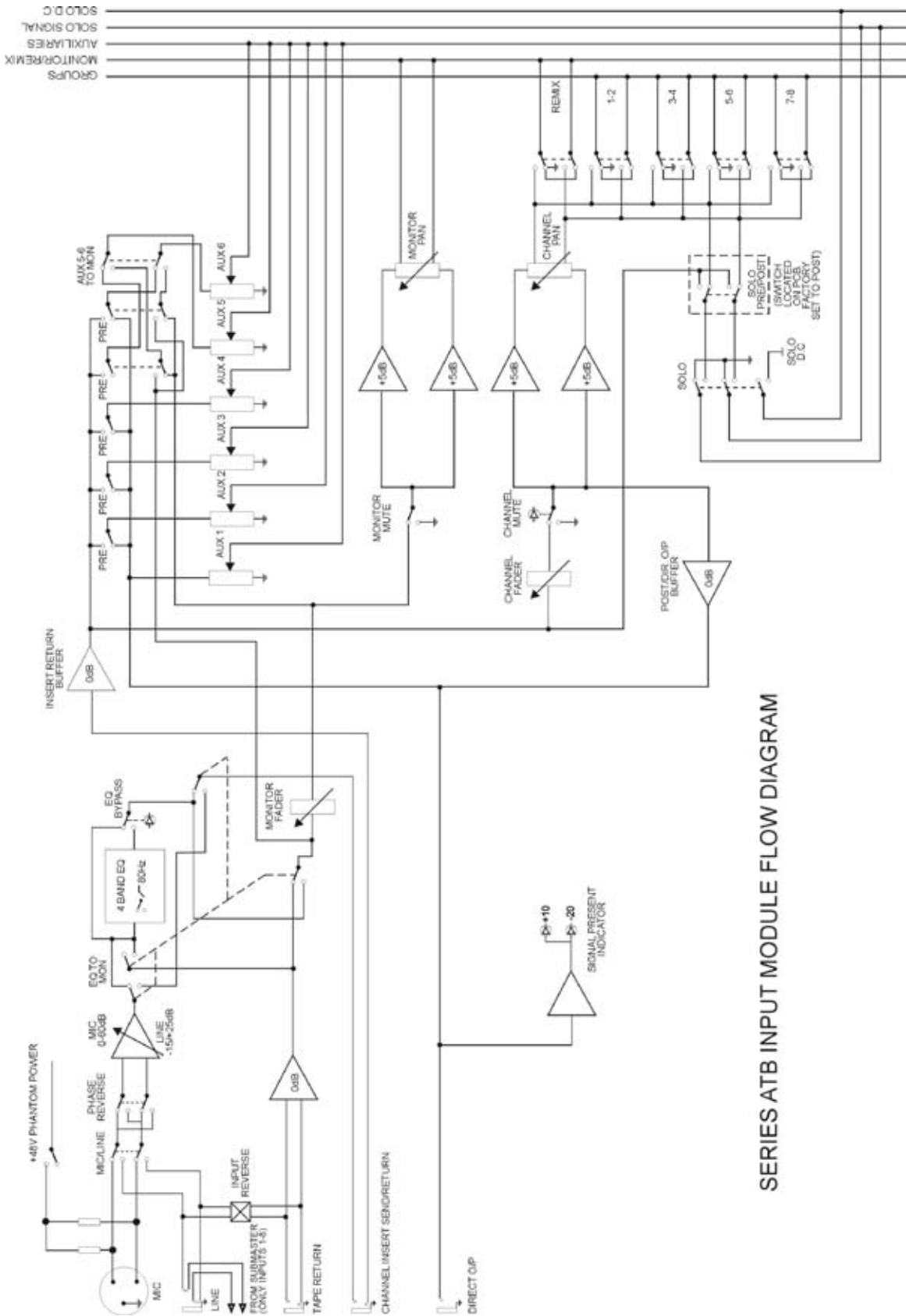
Routing to the multi-track groups is achieved by means of pushbuttons located beside the channel fader. These are arranged in pairs and work in conjunction with the pan control so that if for example groups 1-2 are selected and the pan control is in the mid position, the signal will be fed to groups 1 and 2 in equal amounts. If however the pan control is positioned fully anti-clockwise (left) the signal will only feed group 1. If the pan control is positioned fully clockwise (right), the signal will only feed group 2

A high quality, 100mm metal bodied fader is provided which gives precise and smooth control of the channel level.

Reference to the Input module signal flow diagram will help to give a better understanding of the way in which the signal is routed from the input to the group output



# INPUT MODULE FLOW DIAGRAM



SERIES ATB INPUT MODULE FLOW DIAGRAM



**SUBMASTER MODULE**

**SUBMASTER MODULE General Description**

The Submaster section of the console greatly enhances the flexibility of the Series ATB console and as a consequence, provides a number of different functions.

Each Submaster channel provides a method of adjusting the output level of each of the 8 subgroup outputs via a 100mm high quality fader. The level is also indicated by a high quality 12 segment bar graph display.

The submaster also provides a method of monitoring and panning the subgroup output on the control room speakers. By using these controls it is possible to build up a monitor mix of the recorded tracks and any that are about to be recorded. To enhance this facility, level controls for auxiliary sends 5 and 6 are included on each submaster section making it possible to send a headphone mix from the subgroups together with reverb or echo effects. Auxiliary send 5 can also be switched pre or post the submaster monitor level control, making it usable as either a headphone feed (in 'pre' mode), or a reverb send (in 'post' mode).

A balanced line level input is provided on the rear panel behind the submaster section so that the replay from an 8 track DAW system can be connected to the submaster section. Individual 'TAPE' switches make it possible to select either the send to the DAW or its output onto the monitoring system. The bar graph metering and auxiliary sends follow this switch so that they automatically switch over according to the position of the 'TAPE' switch. By this method it is very easy to build a monitor mix while tracking instruments on a DAW.

A further useful feature is that the balanced line level inputs connected to the submaster 'MONITOR RETURNS' are also connected to the 'LINE INPUT' jacks of channels 1-8. This means that when you want to mix from the DAW on channels with equalisation and more auxiliary sends, selecting 'LINE' input on channels 1-8 will automatically connect the DAW outputs to these channels.

Each of the 8 submaster channels is also provided with a 'SOLO' push button. This makes it possible to listen to the signal on any selected submaster in isolation and in stereo, depending on the position of the submaster 'PAN' control.





Since the 'SOLO' function is a 'post fader' signal, its volume will be dependent on the position of the Submaster 'MON LEVEL' control. This makes it possible to solo a number of Submaster channels and create a balanced mix of the soloed signals.

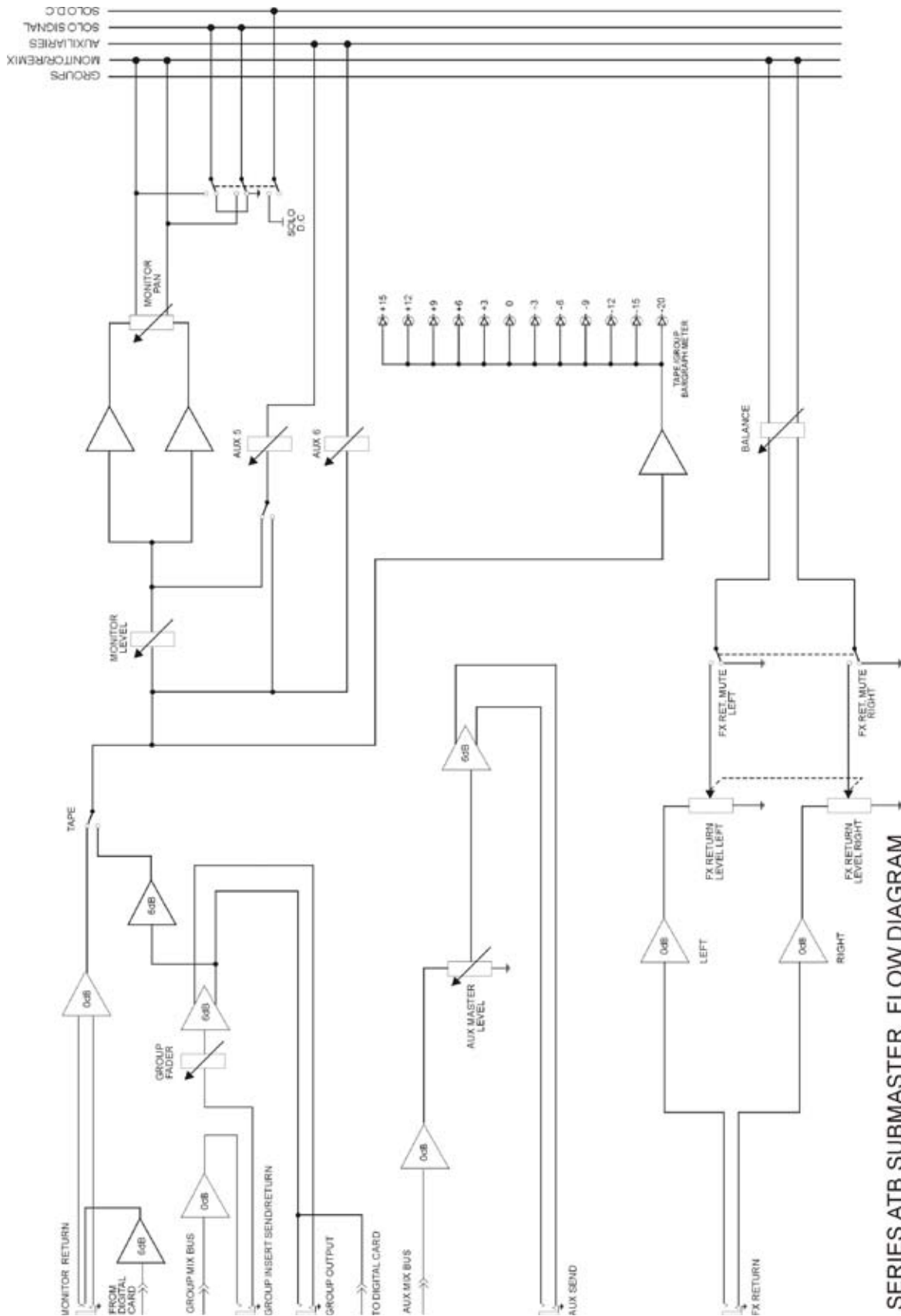
A separate facility on each Submaster channel is the inclusion of a stereo Effects Return section located directly above the 12 segment bar graph meters. Individual jack inputs are provided on the rear panel behind the Submaster section and are wired in stereo. This therefore adds another 16 input capability to the console in remix mode. An effects return level control (which adjusts both left and right inputs simultaneously) is combined with a balance control which makes it possible to attenuate either of the left or right signals in relation to each other. In the centre position the signals will be of equal strength, but when the balance control is turned anti-clockwise (left) the right signal is attenuated by a varying degree and when the balance control is turned clockwise (right), the left signal is attenuated by a varying degree. This allows for adjustment if the stereo signal connected is not equally matched.

A final facility included only on the first 6 Submaster sections, is a master control for each of the balanced auxiliary sends. This makes it possible to adjust the overall level of signal being sent, to either a reverb (or other such signal processing device), or a headphone amplifier if the auxiliary sends are being used as a headphone cue (or foldback) system.

Reference to the Submaster module signal flow diagram will help to give a better understanding of the way in which the signal is routed.



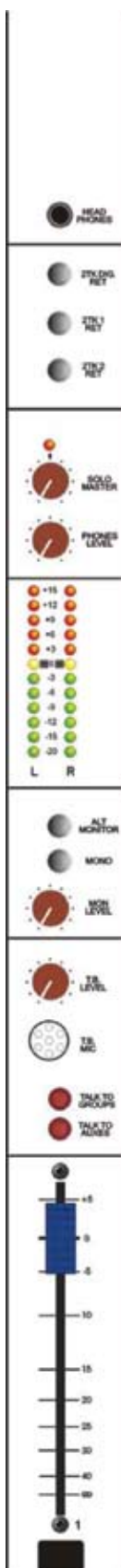
**SUBMASTER FLOW DIAGRAM**



**SERIES ATB SUBMASTER FLOW DIAGRAM**



## MASTER SECTION



### MASTER SECTION General Description

The Master section of the ATB console contains all of the controls that affect the overall functionality of the console.

An accurately matched 100mm stereo fader controls the level of the master stereo balanced outputs.

A comprehensive talkback system is provided which incorporates a high quality electret microphone and amplifier with a continuously variable level control. The signal can be routed to either the 8 subgroups and master left/right outputs for 'slate' announcements (so that the title of a song can be recorded at the beginning), or to the auxiliary sends so that the engineer can communicate with the musicians when they are in a separate room. When the 'TALK TO GROUPS' button is used, the monitor signal is dimmed by 25db to avoid feedback in the control monitor speakers.

Above the talkback section is the master monitor level control for the control room speakers. This is a very important control as without this turned up, it will not be possible to hear any signal on the control room monitor speakers. Coupled with this control is a 'MONO' push button which combines the left/right monitor signals so that it is possible to check mono compatibility. It should be noted that the mono facility is purely a monitor function and does not affect the main stereo left/right output which remains as a true stereo image. A very useful further facility in this section is the provision of an 'ALT MONITOR' push button which routes the monitor signal to a second set of output jacks on the rear of the master section so that an alternative monitor system can be set up. This makes it possible to listen to the signal on (for example), a different set of monitor speakers so that comparisons can be made.

A pair of very accurate 12 segment bar graph meters indicate the signal being sent from the master left/right stereo outputs. As an extremely useful additional feature to these electronic meters, a pair of conventional moving coil VU meters are provided which read in parallel with the bar graph meters. This makes it possible to compare the signal measured on the two different types of meters. By their nature, bar graph meters such as the ones fitted to the Series ATB console, respond very quickly to signal peak and therefore are useful when used on instruments with complex dynamic waveforms such as pianos etc.



This type of meter will therefore greatly assist the engineer when it comes to avoiding distortion during recording. However, when used on less complex waveforms generated by an electric guitar for example, it is possible to under record the instrument using a bar graph display as they will sometimes react too much to the peak amplitudes of the signal and not the average amplitude. They will however, generally result in a better use of the available dynamic range of the recording medium. An analogue meter such as the conventional V.U. meter also fitted to the Series ATB console, responds mostly to the average level of the programme, so can therefore be considered more useful when recording instruments such as an electric guitar. The combination of electronic bar graph and analogue metering therefore provides the user with the best of both worlds.

A headphone jack with a high quality stereo amplifier is provided so that if the engineer wishes to monitor the console only on headphones, they can be connected to the jack socket at the top of the master section. A separate stereo volume control is provided so that the level of the headphones can be adjusted independently.

A stereo master level control is also provided so that the volume of the AFL/PFL system can be adjusted in the control room monitor speakers.

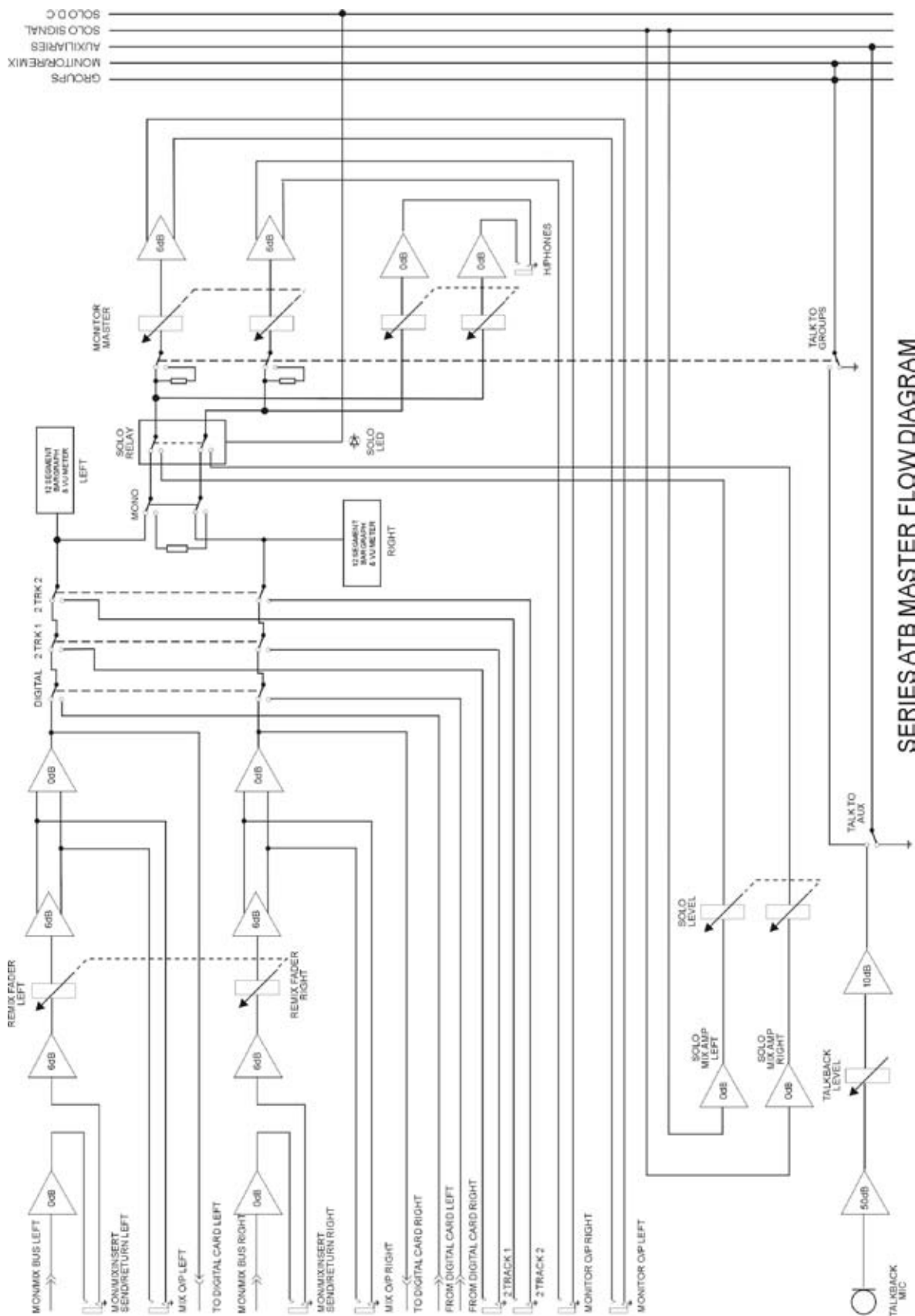
The final facility provided by the master section is the ability to monitor any one of three stereo playback devices. '2TK 1 RET' and '2TK 2 RET' are conventional analogue connections located on the rear of the submaster section so that the playback of a stereo device such as a cd player or output of a computer sound card can be selected to the control room monitor speakers. '2TK DIG RET' is provided for use when the optional digital I/O card is connected. This routes the converted analogue output of a digital playback device connected through the I/O card onto the control room monitor speakers. The bar graph displays will meter the playback level and, when any of the three buttons are depressed, it is no longer possible to listen to the master stereo output from the console. This is to make sure that when listening to an important final mix, nothing else will be monitored. The signal will however still appear at the balanced master output jacks at the rear of the master section and will also be controlled by the stereo master fader. By selecting the appropriate pushbutton, it is therefore possible to listen to a stereo mix after it has been recorded, providing an easy A/B comparison method.

Reference to the Master Section signal flow diagram will help to give a better understanding of the way in which the signal is routed.





# MASTER FLOW DIAGRAM



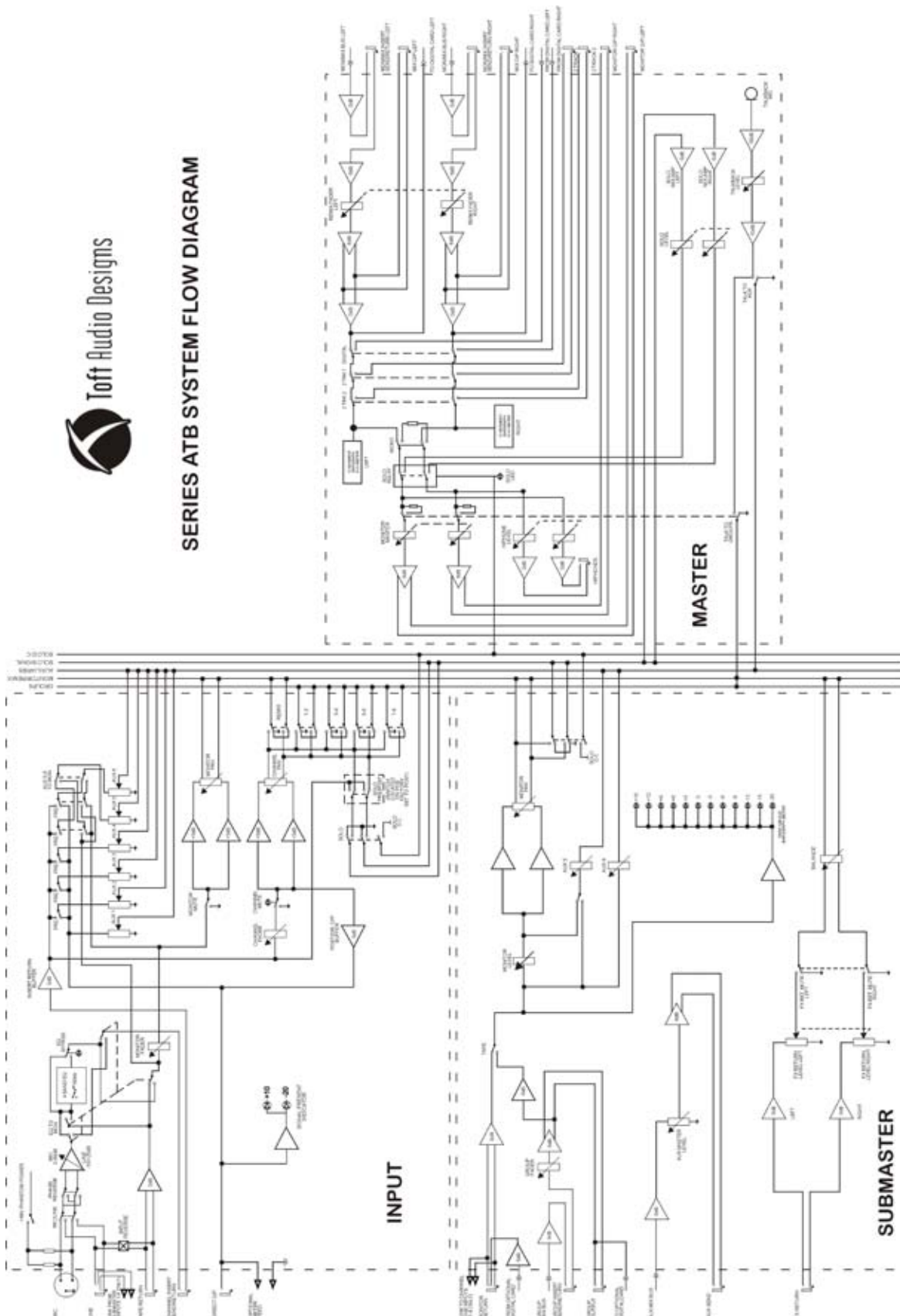
SERIES ATB MASTER FLOW DIAGRAM



OVERALL FLOW DIAGRAM



SERIES ATB SYSTEM FLOW DIAGRAM





## OPERATIONAL DESCRIPTION

### Multi-Track Recording Set Up

Make sure that all buttons are up and that all controls are in their default positions, i.e.: centre detent potentiometers such as pan and equaliser boost/cut are in their centre positions, level controls and equaliser frequency sweeps are at minimum (anti-clockwise). The channel fader should be set to minimum.

The output from a balanced microphone should be wired in accordance with the information given in section 2 of this manual and connected to the XLR input at the rear of the appropriate channel. If the microphone requires +48 phantom power, this should be selected by depressing the '+48' red button at the top of the module.

The channel fader should be advanced to the top of its' travel (maximum) and the mic/line level control (at the top of the module) should be advanced until the green '-20' LED illuminates to a steady state. This will indicate that there is signal present at the module output.

There are two ways to set up the console to record in a multi-track setup. If the console is only being used in conjunction with an 8 track recorder or DAW, the signal can be sent via the 'Submaster Outputs' to the inputs of the recorder. The output of the 8 track recorder or DAW should be connected to the 'Monitor Returns'. If for example, you wish to route to track 5 of the multi-track recorder, depress the appropriate channel button marked '5-6' and turn the channel pan control fully anti-clockwise (left). The signal will now appear at the input of the 8 track recorder or DAW and the subgroup bar graph display will indicate the level being sent to the recorder. To monitor the signal being replayed from the DAW, depress the red 'TAPE' button on the appropriate submaster channel, turn up the 'MON LEVEL' control on the appropriate submaster channel and the signal will appear on the monitor speakers. The 'MONITOR MASTER' control in the Master section must be turned up. Since a lot of DAW's (and of course tape based recording systems) have automatic input/output monitoring systems such that when a track is set to record the input is monitored through the playback outputs and then automatically switches to playback when the track is not in record, the 'TAPE' button can be left depressed and the DAW will provide the necessary monitor switching.

The second way of setting the console up is by using the 'in-line' monitoring facility provided on every input. This system is to be favoured when using the console with anything more than an 8 track recording system.

As each input module is provided with a direct output signal derived directly after the channel fader, this can be used to feed the individual tracks of a multi-track recording device. Therefore the 'DIR O/P' jack located on the rear panel behind each input should be connected to the appropriate input of the recording device. It should be noted that connecting to the 'DIR O/P' jack does not break the signal path of the input module, so all other facilities such as routing and panning to the 8 submaster groups is still possible. This is especially useful since using the direct outputs only allows one signal to be routed to each track. It is therefore necessary to use one of the sub groups only if a mix of a number of channels is required. As the 'DIR O/P' jack does not break the signal path from the channel, it is a simple job to connect from this jack directly to the input of the multi-track recorder or DAW.



To monitor the signal being recorded (via the multi-track recorders input/output monitoring system, the recorder's output should be connected to the balanced 'MON' jack located behind each input module. This then connects the signal to the separate monitor section incorporated within each input module.

By turning up the appropriate 'MON LEVEL' control on an input module and adjusting the associated 'MON PAN' control, the signal can be listened to and placed anywhere between the left and right control room monitors. This of course is independent of any main channel panning that is taking place. By this means a monitor mix of all 16, 24 or 32 channels can be built up depending on the mixer configuration. The multi-track recorder's own meters can be used to control the record level and monitor the playback of the signal.

A further very useful feature is the ability to select auxiliary sends 5 and 6 to the monitor circuit by means of the 'AUX 5-6 TO MON' push button located next to the monitor level control. This makes it possible to supply a headphone mix to artists during recording (by using for example auxiliary 5 as a headphone feed) and also monitor reverb (or some other processing device) by using (for example) auxiliary 6. In this mode, pressing the 'PRE' push button selects the auxiliary send signal either pre or post the monitor level control. It is therefore extremely quick and easy to record and overdub using this section of the console. An 'EQ TO MON' push button completes the versatility of this section by making it possible to route the entire four band equaliser onto the monitor section. This makes it possible to add equalisation only on monitor to the track being recorded or playing back, so that it is possible to hear how the mixed track might sound without actually recording it. In this mode, the EQ is now bypassed on the channel section.

By using a combination of the channel direct outputs and the subgroups coupled with the channel and subgroup monitor returns, it is of course possible to record a larger number of tracks than the console has inputs. Therefore the 16 input console can be connected to a 24 track recorder, the 24 input console can be connected to a 32 track recorder and the 32 input console can be connected to a 40 track recorder.

When it is required to use the console to mix down the recorded tracks, it is not necessary to re-cable. By depressing the 'I/P REV' push button at the top of the input module, the signal connected to the 'MON' jack is routed to the channel line input. The signal therefore appears at the main channel fader and can be routed to the stereo mix via the 'L-R' push button and panned via the channel pan control. The normal channel facilities such as AFL, equalisation and auxiliary sends can now be used. Whatever was plugged into the channel line input now appears at the monitor section and can also be routed to the stereo mix via the monitor level control and pan. In this way, the number of input channels to the stereo mix is doubled which greatly enhances the versatility of the console.

## **Recording With A Microphone**

After routing the signal to the desired output, the channel fader should be put to the top of its travel (maximum position) and the 'INPUT GAIN' control should be adjusted so that either the input meter of the device that the microphone is being sent to reads an acceptable level or the channel input



level LEDs (situated next to the channel fader), show the signal level. Using these meters, the level should be adjusted until the red LED just illuminates on peaks and the level control should then be brought down so that the red LED is just extinguished.

The phase reverse button located at the top of the module works in conjunction with the mic and line inputs and can be used to reverse the signal phase of either input. Phase differences can occur when two microphones are placed in close proximity (for example when recording acoustic guitar) and causes a cancellation of certain frequencies (usually bass) resulting in a 'thin' sound. By using the phase switch, this effect can often be minimised or eliminated altogether. In the line mode, phase errors can often occur due to incorrect phase wiring of a piece of equipment connected to the console. Operation of the phase switch under these circumstances will often solve the problem.

Very often at this stage in the recording process, musicians may require a headphone feed so that they can hear their instrument or a mix of the other musicians. A suitable headphone amplifier should be connected to the auxiliary sends) that is to be used for this purpose. To provide a headphone feed (foldback or cue) signal, it is usual that the signal is taken pre-fader. This is so that if the engineer changes the fader settings during recording, it will not affect the musicians balance. On the ATB console, auxiliary send 1 is designated as a pre-fader send for this purpose. To provide a musicians headphone feed therefore, all that is necessary is to turn the appropriate master auxiliary send level control at the top of the Master Module to just below maximum, and advance the relevant channel module auxiliary send level until the musician is happy with the level.

To provide auxiliary send to a reverberation/echo device, the same procedure as above should be followed, except that it is recommended that auxiliary sends 2-6 are used as they can be selected post fader. This is necessary because unlike headphone feeds, the reverb/echo signal will need to follow the fader level.

Equalisation can now be added to the microphone signal if required and this is done by simply switching the equaliser into circuit by means of the illuminated switch in the equaliser section. The ATB Series equaliser is very comprehensive and consists of four separate bands each offering +/- 15dB boost or cut. The high and low sections are of the 'shelving' type whilst the two mid sections are of the 'peaking' type and all ranges have been chosen for their musical response and character. An 80Hz roll off filter is also incorporated which is particularly useful in eliminating microphone rumble etc.

If it is required to introduce further signal processing such as a limiter compressor etc. into the signal chain, this can be accomplished by connecting the device via the individual 'INSERT' jacks on the rear of each input module. The send (input to the device) is connected to the tip of the jack and the return (output of the device) is connected to the ring of the jack. The device should be adjusted so that the required amount of signal processing takes place and care should be taken to make sure that additional gain is not introduced that may cause distortion. If this is the case it may be necessary to adjust the channel module input level control to allow for this.

Once all of the microphone channels have been adjusted in the above fashion, the recording can be



carried out. If line level signals are also to be recorded, the procedure is identical to that above except that a signal is connected to the channel line input jack and the 'LINE' button at the top of the module is depressed. It should be noted that when the input level control is in its' midway position, the line amplification is 0dB. Decreasing the level will attenuate the signal by up to -15dB and increasing it from midway will amplify it by up to 25dB.

Once the recording has been made, it can be played back either via the Submaster section or Monitor sections of the input channels. This is explained in detail in the previous 'Multi-Track Recording' section.

### Mixing Down

Changing from the recording mode to mixing on the Series ATB is very easy to accomplish as a lot of the functions are selectable by push buttons rather than re-patching connections.

If the replay lines from the multi-track recorder have been connected to the 'LINE' jacks on each Input module, depressing the 'LINE' button at the top of the module will bring the signal through the channel fader. If the replay lines from the multi-track recorder have been connected to the 'MON' jack on the rear of the input module (as would be the case when using the channel monitor section for multi-track monitoring), depressing the 'I/P REV' button will bring those signals through the input channel via the line input. Any signal that was connected to the line input will now be routed through the channel monitor section.

The input level control at the top of the module should be at it's midway position and the channel fader advanced until the green -20db illuminates and an appropriate signal level is metered on the master left/right bar graph displays.

To select a channel to mix it is only necessary to depress the 'L-R' button on the appropriate channel located next to the channel fader. The channel pan control will position the signal in the stereo image. The channel fader will control the level to the stereo mix which will be metered on both the left and right bar graph displays located in the master section as well as the two analogue meters. The stereo master fader located in the Master Module should be put to the top of it's travel and the master monitor level control should be adjusted so that the signal from the control room monitor speakers is at the appropriate level. When the final mix has been recorded on one of the two analogue 2 track recorders, or digital recorder that can be connected to the console, replay of the final stereo mix can be made via the '2 TK 1 RET', '2 TK 2 RET', or '2 TK DIG RET' selection buttons on the master module.

When building up the mix, it is important to work with the relationship between the input level control and the channel fader such that the fader is always as close to the top of it's travel as possible yet still allowing enough in hand to increase gain for solo passages etc. If the fader is operated too low in its travel and the input level control too far advanced, overload/headroom problems are more likely to occur, especially when introducing equalisation.



As described in the previous sections on recording and overdubbing, signal processing such as limiter/compressors can be added to individual channels via the 'INSERT.' send and return jacks. Alternatively, overall limiting/compression of the stereo mix can be accomplished by means of the 'MASTER INS.' left and right jacks located behind the master module.

Reverb/echo effects can be added by using any of the six auxiliary send outputs connected to an appropriate reverb device. The output of these devices can either be connected to any of the eight dedicated Stereo Effects channels located in the Sub Group section of the console, or connected as a line input to any input module or channel Monitor section.

As described previously, one of the main features of the console is the ability to use the channel Monitor sections as additional line inputs during mixdown. This virtually doubles the amount of channels available to the mix with the possibility of adding equalisation, auxiliary sends, panning and level controls.

To use this facility, let's assume that the replay from the multi-track is connected to channels 1-24 line inputs, line has been selected and the signal routed through the 'L-R' buttons to the stereo mix. Additional line level inputs can now be connected to the 'MON' jack on any input module and that signal will also feed the stereo mix. Adjusting the 'MON Level' control and pan will place the signal across the stereo mix. Selecting 'EQ' on the channel monitor section will make it possible to add equalisation to it. If reverberation/echo is required, this can be achieved by selecting the appropriate 5-6 auxiliary send to be from 'MON' and the signal will now be sent either pre or post the appropriate monitor level control. The signal can be muted via the illuminated 'MUTE' button in this section.



## DIGITAL I/O OPTION

A unique and extremely versatile feature of the Series ATB console is the optional digital I/O card that can be plugged into the rear panel of the console.

Extremely high quality analog to digital and digital to analog convertors are used so that at no time is either the analog or digital signal quality compromised. Sample rates up to 192kHz are supported with 24 bit conversion.

A total of ten digital sends and ten digital returns are provided. The mixed subgroup outputs provide eight of the sends and the master left right outputs provide the last two. The eight subgroups are available as firewire and ADAT optical signals whilst the master left right outputs are available as firewire and SPDIF. Similarly, the subgroup returns (accessed when the subgroup 'TAPE' buttons are depressed), provide eight of the returns from either a firewire or ADAT optical source. Whilst depressing the 'DIG'. button as a replay source (located on the master section), returns digital signal from either a firewire or SPDIF source.

Wordclock I/O is provided on BNC connectors.

Detailed interface instructions and further information are supplied with the I/O card.





## TECHNICAL SPECIFICATIONS

### Input impedance:

Microphone: >1.2K $\Omega$  electronically balanced  
Line: >10K $\Omega$  electronically balanced

### Output Impedance:

Groups and Auxiliaries: <100 $\Omega$  electronically balanced  
Others: <100 $\Omega$  unbalanced

### Gain:

Microphone: 70dB (channel input to group output)  
Gain: 6dB to 65dB  
Line: -15 to +25dB

### Noise:

Microphone: <-127dBu ref 200 $\Omega$ , 20Hz to 20kHz  
Line: <-75dBu (Eq in, direct output) 20Hz to 20kHz  
Overall system <-75dBu (Line input to group output) 20Hz to 20kHz

### Maximum Levels:

Mic Input: +20dBu at all frequencies  
Line Input: +22dBu at all frequencies  
Group outputs: +26dBu balanced, +22dBu unbalanced into 2K $\Omega$

### Distortion:

Microphone: <0.03% T.H.D.(-50dBu input, +4dBu output)  
Line: <0.02% T.H.D.(+4dBu input, +4dBu output)

### Frequency response:

Microphone:  $\pm$ 1dB 20Hz to 40kHz  
Line:  $\pm$ 0.5dB 20Hz to 30kHz

### Crosstalk:

Adjacent channel: -70dB @ 1kHz, -60dB @ 15kHz  
Adjacent group: -70dB @ 1kHz, -60dB @ 15kHz

Nominal level: +4dBu

In accordance with our policy of product improvement, we reserve the right to alter specifications without prior notice.



**IMPORTANT SAFETY INFORMATION**



**CAUTION: TO REDUCE THE RISK OF ELECTRIC SHOCK, DO NOT REMOVE COVER. NO USER-SERVICEABLE PARTS INSIDE. REFER SERVICING TO QUALIFIED SERVICE PERSONNEL.**



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



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. accordance with manufacturers’ instructions.

**WARNING: TO AVOID FIRE OR ELECTRIC SHOCK HAZARD, DO NOT EXPOSE THIS APPARATUS TO WATER, RAIN OR MOISTURE.**

**NOTE** — This apparatus does not exceed the Class A/Class B (whichever is applicable) limits for radio noise emissions from digital apparatus as set out in the radio interference regulations of the Canadian Department of Communications.

**ATTENTION** — *Le présent appareil numérique n’émet pas de bruits radioélectriques dépassant les limites applicables aux appareils numériques de class A/de class B (selon le cas) prescrites dans le règlement sur le brouillage radioélectrique édicté par les ministere des communications du Canada.*

These limits are designed to provide reasonable protection against harmful interference in a commercial/residential installation respectively. 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. There is no guarantee that interference will not occur in a particular installation. If this equipment does cause interference to radio or television equipment reception, which can be determined by turning the equipment off and on, the user is encouraged to try to correct the interference by any combination of the following measures: (1) Relocate or reorient the receiving antenna (2) Increase the separation between the equipment and the receiver (3) Plug the equipment into an outlet on a circuit different from that to which the receiver is connected (4) Consult your dealer or experienced radio/television technician for additional assistance

**CAUTION** — Changes or modifications to this equipment not expressly approved by the party responsible for compliance could void the user’s authority to operate this equipment.

**IMPORTANT SAFETY INSTRUCTIONS**

1. Read these instructions.
2. Keep these instructions.
3. Heed all warnings.
4. Follow all instructions.
5. Do not use this apparatus near water. Do not expose to drips or splashes. Do not place any objects filled with liquids, such as vases, on the apparatus.
6. Clean only with dry cloth.
7. Do not block any ventilation openings. Do not install this apparatus in a confined space such as a book case or similar unit. Install only in racks designed for the purpose and in accordance with manufacturers’ 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. Protect the power cord from being walked on or pinched particularly at plugs, convenience receptacles, and the point where they exit from the apparatus.
11. Only use attachments/accessories specified by the manufacturer.
12. Use only with a 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.
13. Unplug this apparatus during lightning storms or when unused for long periods of time.
14. 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.
15. Apparatus designed with Class-I construction must be connected to a mains socket outlet with a protective earthing connection (the third grounding prong).
16. This apparatus may be equipped with a single-pole, rocker style AC mains power switch. If so this switch is located on the front panel and should remain readily accessible to the user.
17. The manufacturer reserves the right to change the technical specification of the product without prior notice.

**STATEMENT OF RoHS COMPLIANCE**



PMI Audio Group manufactures complete electronic products which are covered by the European Union’s “Removal of Hazardous Substances” directive 2002/95/EC (RoHS). This directive seeks to eliminate toxic substances from the manufacturing process, such that when equipment is disposed of at the end of its life cycle, the materials it contains do not contaminate the environment and pose health risks. Banned substances are lead, mercury, cadmium, hexavalent chromium, polybrominated biphenyls (PBB) and polybrominated diphenyl ethers (PBDE). Lead is used together with tin in solder connections to reduce the melting point of solder. Lead-free solder requires higher soldering temperatures which in turn places greater thermal stress on components.

PMI Audio Group takes seriously its obligations under the RoHS directive and insists that its factories use only components that are certified RoHS compliant, as well as leadfree solder. In a very few cases the necessary components may not yet be available to the world market but we work continuously to eliminate any such exceptions at the earliest opportunity. Our printed Circuit Boards (PCB’s) and all soldered joints have been lead-free since 2005.

**STATEMENT OF WEEE POLICY**



PMI Audio Group manufactures many complete electronic products which are covered by the European Union’s “Waste Electric and Electronic Equipment” directive 2002/96/EC (WEEE). This directive seeks to ensure that waste electric and electronic equipment is disposed of in an environmentally responsible manner, at the end of its life cycle. PMI Audio Group takes seriously its obligations under this directive to take back WEEE-affected products and, from 13th August 2005, will mark all such products with the crossed-out wheeled bin symbol.

**Business to Business products:** PMI Audio Group will cost-neutrally take back WEEE-affected electric and electronic equipment in this category, from 1st January 2006. PMI Audio Group will work with disposal and recycling partners working within the EU. The waste electric and electronic equipment can then be turned over to a disposal and recycling companies in the countries concerned.

**Business to Customer products:** emerging electric and electronic equipment will be disposed of by local authorities’ collection systems. Dual Use products: this equipment will be disposed of by local authorities’ collection systems.



**WARRANTY**

**TOFT AUDIO DESIGNS LIMITED WARRANTY**

PMI Audio Group warrants that all products will be free from defects in material or workmanship:

A: For a period of (3) three years from the date of purchase (hereinafter the labor warranty period), PMI Audio Group will repair or replace this Product if determined to be defective. After the expiration of the labor warranty period, the Purchaser must pay labor charges.

B: In addition, PMI Audio Group will supply, at no charge, replacements for defective parts for a period of (three years) from the date of purchase. During the labor warranty period, to repair the Product, Purchaser must return the defective Product, freight prepaid, or deliver it to PMI Audio Group Service Center. The product to be repaired is to be returned in either its original carton or a similar package affording an equal degree of protection. PMI Audio Group will return the repaired Product freight prepaid to the Purchaser. PMI Audio Group is not obligated to provide Purchaser with a substitute unit during the warranty period or at any time.

**CONDITIONS**

1. *Notification of claims: Warranty Service:* If Purchaser discovers that the Product has proven defective in material or workmanship, then written notice with an explanation of the claim shall be given promptly by Purchaser to PMI but all claims for warranty service must be made within the warranty period. If after investigation PMI determines that the reported problem was not covered by the warranty, Purchaser shall pay PMI for the cost of investigating the problem at its then prevailing time-and-materials rate. No repair or replacement by Purchaser of any Product or part thereof shall extend the warranty period as to the entire

Product. The specific warranty on the repaired part only shall be in effect for a period of ninety (90) days following the repair or replacement of that part or the remaining period of the Product warranty, whichever is greater.

2. *Exclusive Remedy: Acceptance:* Purchaser's exclusive remedy and PMI's sole obligation is to supply (or pay for) all labor necessary to repair any product found to be defective within the warranty period and to supply, at no extra charge, new or rebuilt replacements for defective parts. If repair or replacement fails to remedy the defect, then and only in such an event, shall PMI exchange to Purchaser a new or reconditioned unit. Purchaser's failure to make a claim as provided in paragraph 1 above or continued use of the product shall constitute an unqualified acceptance of such Product and a waiver by Purchaser of all claims thereto.

3. *Exceptions to Limited warranty:* PMI shall have no liability or obligation to Purchaser with respect to any Product subjected to abuse, improper use, negligence, accident, modification, failure of the end-user to follow the operating and maintenance procedures outlined in the users manual, attempted repair by non-qualified personnel, operation of the unit outside of the published environmental and electrical parameters, or of such products original identification (trademark, serial number) markings have been defaced, altered, or removed. PMI excludes from warranty coverage, Products sold AS IS and/or WITH ALL FAULTS and excludes used products which have not been sold by PMI to the Purchaser. PMI also excludes from warranty coverage consumables such as fuses and batteries, tubes, etc.

4. *Proof of purchase:* The dealer's dated bill of sale must be retained as evidence of the date of purchase and to establish warranty eligibility.

**DISCLAIMER OF WARRANTY**

EXCEPT FOR THE FORGOING WARRANTIES, PMI HEREBY DISCLAIMS AND EXCLUDES ALL OTHER WARRANTIES, EXPRESS OR LIMITED, INCLUDING, BUT NOT LIMITED TO ANY/OR ALL IMPLIED WARRANTIES OF MERCHANTABILITY, FITNESS FOR A PARTICULAR PURPOSE AND/OR ANY WARRANTY WITH REGARD TO ANY CLAIM OF INFRINGEMENT THAT MAY BE PROVED IN SECTION 2-312(3) OF THE UNIFORM COMMERCIAL CODE AND/OR IN ANY COMPARABLE STATE STATUE. PMI HEREBY DISCLAIMS ANY REPRESENTATIONS OR WARRANTY THAT THE PRODUCT IS COMPATIBLE WITH ANY COMBINATION OF NON-PMI AUDIO PRODUCTS PURCHASER MAY CHOOSE TO CONNECT TO THE PRODUCT.

**LIMITATION ON LIABILITY**

THE LIABILITY OF PMI, IF ANY, AND PURCHASER'S SOLE AND EXCLUSIVE REMEDY FOR DAMAGES FOR ANY CLAIM OF ANY KIND WHATSOEVER, REGARDLESS OF THE LEGAL THEORY AND WHETHER ARISING IN TORT OR CONTRACT, SHALL NOT BE GREATER THAN THE ACTUAL PURCHASE PRICE OF THE PRODUCT WITH RESPECT TO WHICH SUCH CLAIM IS MADE. IN NO EVENT SHALL PMI BE LIABLE TO PURCHASER FOR ANY SPECIAL, INDIRECT, INCIDENTAL, OR CONSEQUENTIAL DAMAGES OF ANY KIND INCLUDING, BUT NOT LIMITED TO, COMPENSATION, REIMBURSEMENT OR DAMAGES ON ACCOUNT OF THE LOSS OF PRESENT OR PROSPECTIVE PROFITS OR ANY OTHER REASON WHATSOEVER.



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TO BE COMPLETED AT THE TIME OF PURCHASE

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