## OPERATORS MANUAL

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## DECLARATION OF CONFORMITY

The Directive Covered by this Declaration
89/336/EEC Electromagnetic Compatibility Directive, amended by 92/31/EEC \& 93/68/EEC
Equipment TypeProduct Name
Audio Mixing Console Heritage 3000
The Basis on which Conformity is being Declared
The Products named above comply with the requirements of the listed EU directives by meeting the following standards:

EN 55013: 1990
EN 50082: 1992
EN50081/1 and /2 Generic Standard Using En55022 Limits and Methods
EN55022 Class B Conduct Emissions
EN55022 Class B Radiated Emissions
EN61000-4-4 Fast Transient Burst Level 4
EN61000-4-2 Static Discharge Level 4
EN60204 Earth Continuity, Insulation at 500V

Signed:.. 4 .................Alex Cooper
Authority:Project Leader
Date:1st April, 1999
Attention!
The attention of the specifier, purchaser, installer or user is drawn to special limitations of use which must be observed when these products are taken into service to maintain compliance with the above directives. Details of these special measures and limitations to use are available on request and are also contained in product manuals.

## ATTENTION!

The following special limitations apply to the console and must be observed in order to maintain safety and electromagnetic compatibility performance:

## POWER CONNECTION

The console should only be operated with the power supply connected to ground via its mains supply connector.

## CONTROL CONNECTIONS

The console should only be operated with high quality screened control cables. All connector shells should be of metal construction so that they provide a screen when they are plugged into the console. All DEE connector shells should be connected to the cable screen. All XLR and DIN connectors should have pin 1 connected to the cable screen.

## AUDIO CONNECTIONS

The console should only be operated with high quality screened twisted pair audio cables. All connector shells should be of metal construction so that they provide a screen when they are plugged into the console. All JACK connector shells should be connected to the cable screen. All XLR connectors should have pin 1 connected to the cable screen.

## ELECTRIC FIELDS

If the console is operated in an electromagnetic field that is amplitude modulated by an audio frequency signal, the signal to noise ratio may be degraded. Degradation of up to 60 dB may be experienced under extreme conditions ( $3 \mathrm{~V} / \mathrm{m}, 90 \%$ modulation).

## INSTALLATION

There are a number of points to consider when installing a mixing console. Many of these points will have been addressed before the console is even unpacked but it is worth repeating them.

## POSITION

The console should be located in a convenient space commensurate with the use to which the console is being put. Ideally a cool area is preferred not in close proximity to power distribution equipment or other potential sources of interference. Provision should be made for some flat surface surrounding the console to prevent people using it as a table top.

## POWER

The power supply should be located as far from the console as the connecting cable will allow. It should be set for the appropriate line voltage and plugged into the mains outlet using the supplied cable.


RS-232

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MIDAS HS0001 Mono Input Module

The 48 V switch connects 48 volt phantom power to the input connector which is suitable for a condenser microphone or DI box.

The PAD switch gives 25 dB of attenuation to the input signal which will allow the connection of high output microphones or line level signals. If the input amplifier is transformer coupled (option) the pad greatly reduces the risk of saturation at very low frequencies.

The PHASE switch activates a 180 degrees phase change within the input amplifier.

The PRE switch re configures the direct output to derive signal from the input channel pre insert and equaliser. It is important to note that pre insert direct outputs are also pre mute.

The treble FREQ control gives continuous adjustment of the frequency range that the treble equaliser acts on from 1 k to 20 k .

The treble BELL switch converts the treble equaliser from traditional MIDAS shelving response to full parametric operation.

The hi mid FREQ control gives continuous adjustment of the frequency range that the hi mid equaliser acts on from 400 Hz to 8 k .

The INS switch connects the input insert return signal to the input channel signal path.

The EQ switch connects the equaliser into the input channel signal path.

The GAIN control gives continuous adjustment of the input amplifier gain from +15 dB to +60 dB .

The DIRECT output control gives continuous adjustment of the direct output level from +10 dB to off. The output is derived from the input channel post equaliser pre fader signal.

The TREBLE (dual concentric top) control gives continuous adjustment of boost and cut from +15 dB to -15 dB with a 0 dB centre detent.

The treble WIDTH (dual concentric bottom) control gives continuous adjustment of bandwidth from 0.1 to 2 octaves with a 0.5 octave centre detent. This only operates when the BELL switch is activated.

The HI MID (dual concentric top) control gives continuous adjustment of boost and cut from +15 dB to -15 dB with a 0 dB centre detent.

The hi mid WIDTH (dual concentric bottom) control gives continuous adjustment of bandwidth from 0.1 to 2 octaves with a 0.5 octave centre detent.

The insert PRE switch arranges the input channel signal to pass through the insert point before the equaliser when activated and after the insert point when not activated.

The lo mid FREQ control gives continuous adjustment of the frequency range that the lo mid equaliser acts on from 100 Hz to 2 k .

The bass BELL switch converts the bass equaliser from traditional MIDAS shelving response to full parametric operation.

The bass FREQ control gives continuous adjustment of the frequency range that the bass equaliser acts on from 20 Hz to 400 Hz .

The HI PASS switch connects the filter in the input channel signal path before the insert point and equaliser.

The LO MID (dual concentric top) control gives continuous adjustment of boost and cut from +15 dB to -15 dB with a 0 dB centre detent.

The lo mid WIDTH (dual concentric bottom) control gives continuous adjustment of bandwidth from 0.1 to 2 octaves with a 0.5 octave centre detent.

The BASS (dual concentric top) control gives continuous adjustment of boost and cut from +15 dB to -15 dB with a 0 dB centre detent.

The bass WIDTH (dual concentric bottom) control gives continuous adjustment of bandwidth from 0.1 to 2 octaves with a 0.5 octave centre detent. This only operates when the BELL switch is activated.

The HI PASS filter control is continuously adjustable from 20 Hz to 400 Hz .

The aux PRE switches only operate when in one of the AUX bus modes; they change the signals sent to the group busses from post fader to pre fader. When configured as stereo auxes only the right switches are active.


The configurable group MIX controls (1 to 24 ) have two functions:-
i. They operate as bus assign ON/OFF switches by way of a non latching push/push action with LED status indication.
ii. They adjust the levels sent from the input channel to the group busses when in one of the AUX bus modes.

The group mix busses can be configured in three modes:- MONO AUX, STEREO AUX or POST PAN GROUP. This is controlled on a bus by bus basis by the global BUS MODE switches on the GROUP modules. When configured as mono auxes the left and right controls give independent level adjustments from +6 dB to off. When configured as stereo auxes the left controls perform a pan function with a constant power ( -3 dB ) law while the right controls give continuous level adjustment from +6 dB to off. When configured as stereo groups the level control functions are disabled such that any assigned busses are sent at unity gain from the channel post fader, post pan signals.

The PAN defaults to control the channel placement within a group or master stereo mix and has a constant power law i.e. -3 dB at the centre position.

The SIS switch enables the spacial imaging system which operates in conjunction with the pan and image controls. It also acts as a left, centre, right master bus enable overriding any stereo and mono master bus assignments.

When the spacial imaging system is active the IMAGE control can modify the action of the pan control so as to place the channel within a three speaker system. When the image control is fully clockwise the pan control will operate in full left, centre, right such that a centre panned signal will route to the centre speaker only and will not appear in either of the left or right outputs. When the image control is fully anti-clockwise the pan control reverts to stereo such that a centre panned signal will route at equal power to the left and right speakers. All other Image control positions generate a composite blend of the stereo and LCR panning systems so that the optimum degree of center image focus and speaker power can be obtained. When the image control and pan control are both set central the channel will be routed with equal power to all three speakers. Constant power is maintained at all times so that the image can be adjusted during the show without any perceived level change.

The ST switch connects the post fader channel signal to the master stereo bus via the pan control.

The MONO switch connects the post fader channel signal to the mono master bus.

The METER monitors the peak signal level of the pre fader input channel.

The MUTE switch mutes the input channel at all points after the insert send. The switch can be controlled from snapshot automation and by automute scenes.

## MIDAS HS0003 Input Fader



The SAFE switches disable remote control of the channels as follows:-
i. The MUTE SAFE removes the channel mute from the snapshot automation and automute scenes.
ii. The FADER SAFE removes the channel fader from the virtual fader automation and VCA master fader control including vca mutes.
iii. The AUTO SAFE removes the channel from the snapshot automation system only; leaving the automutes, VCA masters and assignment systems active.

The SET switch is used to programme the channel automute and VCA master assignment. The central controller MODE and ASSIGN keys select the desired automute or VCA group and the SET switch will toggle the channel on and off with each alternate press.

The STATUS leds are used to show fader positions and the status of VCA and MUTE group assignments. The central controller MODE switches toggle through the four available states:- VCA group assignment, MUTE group assignment, FADER position manual recall and null, and full automated VIRTUAL FADER RECALL.

FADER MANUAL RECALL AND NULL In this mode, the STATUS LEDs are used to prompt the operator where to move the fader. If the fader is not at the position stored in the current recalled snap shot, one/two LEDs will flash to indicate where the fader should be. A single flashing LED indicates that the fader should positioned next to that LED, if two LEDs are flashing the fader should be between the two LEDs. As the fader is moved closer to the required position the LED(s) will stop flashing and will be replaced by a single continuously lit LED. Once the fader is at the correct position all LEDs will extinguish.

The SOLO switch sends the input channel signal to the PFL mono and AFL stereo busses. If the switch is pressed for a short time it will latch on or off, but if it is held on for more than 1 second the latching is disabled and when the switch is released the channel solo will turn off. As a default the solo system is auto cancelling so each new solo cancels the last. This function is time dependant which allows several solos to be active as long as they are switched on at approximately the same time. The SOLO ADD MODE switch on the MONITOR module defeats the auto cancelling and allows multiple channel monitoring. In this mode input solos have priority over outputs and will temporarily override any active output solos. The input solos also override any active VCA solos.

The FADER gives continuous adjustment of the input channel level from
+10 dB to off.

## FADER POSITION CHECK

When a scene's contents are being "checked" (see automation operation) the STATUS LEDs will indicate the fader position stored in the scene by continuously illuminating one or two LEDs as appropriate.

## VIRTUAL FADER RECALL

 When in VIRTUAL FADER mode (see automation operation) the automation system will generate a "virtual" fader, set to the level of the input fader at the time the snap shot was stored. The level of the virtual fader is added to the level of the physical input fader. In this mode the STATUS LEDs indicate the "position" of the virtual fader by illuminating a bar of LEDs starting at $-\infty$.

MIDAS HS0004 Stereo Input Module

The 48 V switch connects 48 volt phantom power to both input connectors and is suitable for condenser microphones or DI boxes.

The PAD switch gives 25 dB of attenuation in both input signals to allow the connection of high output microphones or line level signals. If the input amplifiers are transformer coupled (option) the pad greatly reduces the risk of saturation at very low frequencies.

The PHASE switches activate a 180 degree phase change within the input amplifiers. The upper switch acts on the left channel and the lower switch acts on right channel.

The treble FREQ control gives continuous adjustment of the frequency range that the treble equalisers act on from 1 k to 20 k .

The treble BELL switch converts the treble equalisers from traditional MIDAS shelving response to bell filters with a 1.5 octave bandwidth.

The hi mid FREQ control gives continuous adjustment of the frequency range that the hi mid equalisers act on from 400 Hz to 8 k .

The INS switch connects the left and right input insert return signals to the input channel.

The insert PRE switch arranges the input channel signals to pass through the insert points before the equalisers when activated and after the insert points when not activated.

The EQ switch connects the left and right equalisers into the input channel signal paths.

The GAIN control gives continuous adjustment of the input amplifier gains from +15 dB to +60 dB .

The BALANCE control gives continuous and reciprocal adjustment of the stereo left and right signal levels by +10 dB to -10 dB . This allows fine adjustment of the left and right signal levels and imaging.

The TREBLE control gives continuous adjustment of left and right boost and cut from +15 dB to -15 dB with a 0 dB centre detent.

The hi mid HI Q control changes the bandwidth of the hi mid equalisers from 1.5 octave to 0.5 octave.

The HI MID control gives continuous adjustment of left and right boost and cut from +15 dB to -15 dB with a 0 dB centre detent.

The lo mid FREQ control gives continuous adjustment of the frequency range that the lo mid equalisers act on from 100 Hz to 2 k .

The bass BELL switch converts the bass equalisers from traditional MIDAS shelving response to bell filters with a 1.5 octave bandwidth.

The bass FREQ control gives continuous adjustment of the frequency range that the bass equalisers act on from 20 Hz to 400 Hz .

The HI PASS switch connects the filters in the input channel signal path before the insert points.


The LO MID control gives continuous adjustment of left and right boost and cut from +15 dB to -15 dB with a 0 dB centre detent.

The lo mid HI Q control changes the bandwidth of the lo mid equalisers from 1.5 octave to 0.5 octave.

The BASS control gives continuous adjustment of the left and right shelving filters boost and cut from +15 dB to -15 dB with a 0 dB centre detent.

The HI PASS filter control is continuously adjustable from 20 Hz to 400 Hz .

The aux PRE switches only operate when in one of the AUX bus modes; they change the signals sent to the group busses from post fader to pre fader.


The configurable group MIX controls (1 to 24) have two functions:-
i. They operate as bus assign ON/OFF switches by way of a non latching push/push action with LED status indication.
ii. They adjust the levels sent from the input channel to the group busses when in one of the AUX bus modes.

The group mix busses can be configured in three modes:- MONO AUX, STEREO AUX or POST PAN GROUP. This is controlled on a bus by bus basis by the global BUS MODE switches on the GROUP modules. When configured as mono auxes the left and right controls give independent level adjustments of the levels sent from a mono sum of the channels left and right sides. The adjustment is from +6 dB to off. When configured as stereo auxes the left controls give level adjustment of the left channel signal levels and the right controls give level adjustment of the right channel signal levels. When configured as stereo groups the level control functions are disabled such that any assigned busses are sent at unity gain from the channel post fader, post pan signals.

The left and right PAN controls are used to place the input channel signals within a stereo group or stereo master mix. As well as image placement, the controls can also adjust the image width from stereo through mono to reverse stereo (left and right crossed over). The controls have a constant power law i.e. -3 dB at the centre position.

The ST switch connects the post fader channel signals to the stereo master bus via the pan controls.

The MUTE switch mutes the input channel at all points after the insert send. The switch can be controlled from snapshot automation and by automute scenes.

## MIDAS HS0003 Input Fader



The SAFE switches disable remote control of the channels as follows:-
i. The MUTE SAFE removes the channel mute from the snapshot automation and automute scenes.
ii. The FADER SAFE removes the channel fader from the virtual fader automation and VCA master fader control including vea mutes.
iii. The AUTO SAFE removes the channel from the snapshot automation system only; leaving the automutes, VCA masters and assignment systems active.

The SET switch is used to programme the channel automute and VCA master assignment. The central controller MODE and ASSIGN keys select the desired automute or VCA group and the SET switch will toggle the channel on and off with each alternate press.

The STATUS leds are used to show fader positions and the status of VCA and MUTE group assignments. The central controller MODE switches toggle through the four available states:- VCA group assignment, MUTE group assignment, FADER position manual recall and null, and full automated VIRTUAL FADER RECALL.

FADER MANUAL RECALL AND NULL In this mode, the STATUS LEDs are used to prompt the operator where to move the fader. If the fader is not at the position stored in the current recalled snap shot, one/two LEDs will flash to indicate where the fader should be. A single flashing LED indicates that the fader should positioned next to that LED, if two LEDs are flashing the fader should be between the two LEDs. As the fader is moved closer to the required position the LED(s) will stop flashing and will be replaced by a single continuously lit LED. Once the fader is at the correct position all LEDs will extinguish.

The SOLO switch sends the input channel signal to the PFL mono and AFL stereo busses. If the switch is pressed for a short time it will latch on or off, but if it is held on for more than 1 second the latching is disabled and when the switch is released the channel solo will turn off. As a default the solo system is auto cancelling so each new solo cancels the last. This function is time dependant which allows several solos to be active as long as they are switched on at approximately the same time. The SOLO ADD MODE switch on the MONITOR module defeats the auto cancelling and allows multiple channel monitoring. In this mode input solos have priority over outputs and will temporarily override any active output solos. The input solos also override any active VCA solos.

The FADER gives continuous adjustment of the input channel level from
+10 dB to off.

## FADER POSITION CHECK

When a scene's contents are being "checked" (see automation operation) the STATUS LEDs will indicate the fader position stored in the scene by continuously illuminating one or two LEDs as appropriate.

## VIRTUAL FADER RECALL

When in VIRTUAL FADER mode (see automation operation) the automation system will generate a "virtual" fader, set to the level of the input fader at the time the snap shot was stored. The level of the virtual fader is added to the level of the physical input fader. In this mode the STATUS LEDs indicate the "position" of the virtual fader by illuminating a bar of LEDs starting at $-\infty$.


The METERS monitor the peak signal levels of the sub group outputs (post fader).

The direct SOLO switches send direct input to the PFL mono and AFL stereo busses (AFL is selected as stereo or mono depending on the group SPLIT switch settings). If a SOLO switch is pressed for a short time it will latch on or off, but, if it is held on for more than 1 second the latching is disabled and when the switch is released the solo will turn off. As a default the solo system is auto cancelling so each new solo cancels the last. This function is time dependant which allows several solos to be active as long as they are switched on at approximately the same time, i.e. to solo both sides of a stereo mix press both solo switches at the same time. Alternatively the SOLO ADD MODE switch on the MONITOR module can be used to defeat the auto cancelling and allow multiple channel monitoring. In this mode input channel solos have priority over all other solos and will temporarily override them.


The global BUS MODE switches configure input module MIX controls to act as either mono aux (level and level), stereo aux (level and pan) or as audio sub groups (post fader and main pan). This makes the console extremely flexible and quick to reconfigure.

The DIRECT input controls give continuous adjustment of the direct input levels from +10 dB to off. The direct signals are summed into the sub group signals and can be used as effects returns etc. or for console bus linking.

The direct PRE switches move the point at which the direct signals are summed into the sub groups. The default is post insert but when the PRE switches are active the signals are summed at the sub group mix busses.

The direct MUTE switches mute the sub group direct inputs at all points.

The PRE insert switches change the signals sent to the matrix mixes from post group insert to pre group insert and override the pre fader switches. It is important to note that pre insert matrix sends are also pre the group mutes.


The matrix MIX controls (1 to 8) give continuous adjustment of the sub group levels sent to the matrix mixes from +6 dB to off.

The PRE fader switches change the signals sent to the matrix mixes from post group fader to pre group fader.

The VCA switches assign the audio sub groups to VCA control from VCA masters 9 and 10.

The MONO switches connect the post fader sub group signals to the mono master bus.

The TALK switches connect the sub groups to the MONITOR module. When the TALK INTERNAL or GENERATOR INTERNAL are active on the MONITOR module the oscillator, pink noise and talk mic can be routed to the sub groups.

The MUTE switches mute the sub group signals at all points after the insert send. The switches can be controlled from snapshot automation.

The SPLIT switch changes the sub group AFL solos from mono to stereo.

The GROUP FADERS give continuous adjustment of the sub group output levels from +10 dB to off.


The SOLO switches send sub group signals to the PFL mono and AFL stereo busses. If a SOLO switch is pressed for a short time it will latch on or off, but, if it is held on for more than 1 second the latching is disabled and when the switch is released the channel solo will turn off. As a default the solo system is auto cancelling so each new solo cancels the last. This function is time dependant which allows several solos to be active as long as they are switched on at approximately the same time, i.e. to solo both sides of a stereo mix press both solo switches at the same time. Alternatively the SOLO ADD MODE switch on the MONITOR module can be used to defeat the auto cancelling and allow multiple channel monitoring. In this mode input channel solos have priority over the sub group solos and will temporarily override them. When the input solos are cancelled the mix group solos will be active again. Sub group solos can also be temporarily overridden by activating the corresponding direct input solos.

## MIDAS HS0013 VCA Master Fader



The vca MUTE switches act on any post fader input channels or audio sub groups which are assigned to be controlled from the corresponding VCA masters. The switches can be controlled from snapshot automation.

The AUTO SAFE switches disable snapshot automation control of the VCA master faders and VCA mutes.

The STATUS LEDs are off when the console is in VCA or mute assignment modes (see assignment operation). When the console is in FADER mode indication the STATUS LED's can indicate one of three states: -

FADER MANUAL RECALL AND NULL In this mode, the STATUS LEDs are used to prompt the operator where to move the fader. If the fader is not at the position stored in the current recalled snap shot, one/two LEDs will flash to indicate where the fader should be. A single flashing LED indicates that the fader should positioned next to that LED, if two LEDs are flashing the fader should be between the two LEDs. As the fader is moved closer to the required position the LED(s) will stop flashing and will be replaced by a single continuously lit LED. Once the fader is at the correct position all LEDs will extinguish.

## FADER POSITION CHECK

When a scene's contents are being "checked" (see automation operation) the STATUS LEDs will indicate the fader position stored in the scene by continuously illuminating one or two LEDs as appropriate.

The vca SOLO switches are used to monitor the VCA master faders by creating a mix on the solo busses which consists of all input channels and audio sub groups which are assigned to control from the corresponding VCA masters. If a VCA solo switch is pressed for a short time it will latch on or off, but if it is held on for more than 1 second the latching is disabled and when the switch is released the solo will turn off. When the console is operating in SOLO ADD MODE input channels have priority over VCA solos and will temporarily override them.

VIRTUAL FADER RECALL When in VIRTUAL FADER mode (see automation operation) the automation system will generate a "virtual" fader, set to the level of the input fader at the time the snap shot was stored. The level of the virtual fader is added to the level of the physical input fader. In this mode the STATUS LEDs indicate the "position" of the virtual fader by illuminating a bar of LEDs starting at $-\infty$.


The METERS monitor the peak signal levels of the three master outputs (post fader).

The SOLO IN PLACE switch sets the console to solo in place mode. In this mode any input solo that is pressed activates a mute of all the other channels. The mute safe switches on the input channels can be used to protect channels from this function if desired.

The DIRECT input controls give continuous adjustment of the direct input levels from +10 dB to off. The direct signals are summed into the master left and right signals and can be used as effects returns etc. or for console bus linking. The master mono also has a direct input XLR on the rear of the console for which is intended for console linking only.

The direct PRE switches move the point at which the direct signals are summed into the masters. The default is post insert but when the PRE switches are active the signals are summed at the master mix busses.

The direct MUTE switches mute the master direct inputs at all points.


The direct SOLO switches send direct input to the PFL mono and AFL stereo busses. If a SOLO switch is pressed for a short time it will latch on or off, but, if it is held on for more than 1 second the latching is disabled and when the switch is released the solo will turn off. As a default the solo system is auto cancelling so each new solo cancels the last. This function is time dependant which allows several solos to be active as long as they are switched on at approximately the same time, i.e. to solo both sides of a stereo mix press both solo switches at the same time. Alternatively the SOLO ADD MODE switch on the MONITOR module can be used to defeat the auto cancelling and allow multiple channel monitoring. In this mode input channel solos have priority over all other solos and will temporarily override them.

The matrix STEREO switches select the source for the lower matrix mix controls as either left channel, right channel or a sum of both.

The PRE fader switches change the signals sent to the matrix mixes from post group fader to pre group fader.

The MONO level control gives continuous adjustment of the mono master output level from +10 dB to off.

The mono TALK switch connect the mono master to the MONITOR module. When the TALK INTERNAL or GENERATOR INTERNAL are active on the MONITOR module the oscillator, pink noise and talk mic can be routed to the mono master outputs.

The mono INST switch connects the mono insert return signals to the mono master signals.

The matrix MIX controls (1 to 8) give continuous adjustment of the master levels sent to the matrix mixes from +6 dB to off. The top control adjusts the feed from the mono master and the lower control adjusts the feed from the stereo masters.

The PRE insert switches change the signals sent to the matrix mixes from post group insert to pre group insert and override the pre fader switches. It is important to note that pre insert matrix sends are also pre the master mutes.

The MUTE switch mutes the mono master signals at all points after the insert send. The switch can be controlled from snapshot automation.

The mono mute SAFE removes the mono mute from snapshot automation.

The TALK switches connect the stereo masters to the MONITOR module. When the TALK INTERNAL or GENERATOR INTERNAL are active on the MONITOR module the oscillator, pink noise and talk mic can be routed to the stereo masters.

The mute SAFE switches remove the stereo master mutes from snapshot automation.

The VCA link to mono switch connects the mono master level control to the stereo master fader so that the mono output tracks any change of the stereo master fader.


The MASTER FADER gives continuous adjustment of the stereo master output levels from +10 dB to off.


The METERS monitor the peak signal levels of the stereo left and right monitor paths.


The 1 kHz switch overrides the swept frequency control giving a fixed 1 k tone.

The PINK switch overrides the oscillator by giving a pink noise output.

The GENERATOR TO INTERNAL switch connects the signal generator output to the console's internal talk all and talk select busses.

The TALK TO ALL switch overrides all output talk switches so that the generator or talk mic can be routed to


The FREQ. control gives continuous adjustment of the oscillator frequency from 100 Hz to 10 k .

The generator LEVEL control gives continuous adjustment of the signal generator peak output signals from +10 dBu to off.

The GENERATOR TO EXTERNAL switch connects the signal generator output to the talk external output XLR.

The TALK XLR socket accepts balanced $150 \Omega$ microphone signals.

The MIC GAIN preset gives continuous adjustment of the microphone amplifier gain from +20 dB to +60 dB and operates in conjunction with a peak limiter which is factory set to +10 dBu .

The TALK TO INTERNAL switch connects the talk mic output to the console's internal talk system and at the same time dims all the local outputs by 20 dB to prevent howl round.

The TALK input routes the talk external input to the mono local monitor output.

The MONO masters switch routes the post fader mono master mix to the mono local monitor output.

The SOLO switch routes solo signals to the mono local monitor output when ever a solo is active on the console. This overrides any signals sent from the mono master but does not override talk inputs.

The mono output "b" C/O switch disconnects the mono local monitor output from the main "a" output and re-routes it to the secondary " b " output.

The phones MUTE switch mutes the headphone outputs.

The ST master switch routes the post fader stereo master mix to the stereo local monitor outputs.

The MONO master switch routes the post fader mono master mix to the stereo local monitor outputs.


The mono output MUTE switch mutes the mono local monitor output.

The PHONES level control gives continuous adjustment of the headphone level from +10 dB to off.

The EXT switch routes the stereo external input ( 2 track return etc.) to the stereo local monitor outputs.

The SOLO switch routes solo signals to the stereo local monitor outputs when ever a solo is active on the console. This overrides any signals sent from the stereo master, mono master or external input.

The stereo output "b" C/O switch disconnects the stereo local monitor outputs from the main "a" outputs and re-routes them to the secondary " b " outputs.

The PHASE switch reverses the phase of the left hand monitor signal.

When the left/right reverse is ON the left hand monitor signals are routed onto the right channel output speakers and the right hand monitor signal are routed onto the left channel output speakers.

The LEFT switch routes left hand monitor signal to both the left and right local monitor speaker outputs.

The left MUTE switch controls the mute function on the left hand side of the stereo local monitor speaker outputs.

The SOLO PFL switch sends the mono PFL solo bus signals to the headphones and local monitor outputs in place of the stereo AFL solo bus signals.

The SOLO ON / CLEAR switch and indicator has two functions; it illuminates when any solo switch is active and when pressed it clears any active solo switches.

The MONITOR fader gives continuous adjustment of all three local monitor output levels from +10 dB to off.


The MONO sum switch adds the left and right monitor signals with a 4.5 dB summing loss.

The - 20 PAD switch acts on all three local monitor outputs causing them to dim by 20 dB . This function is also activated whenever the talk system is in use to prevent howl round.

The RIGHT switch routes right hand monitor signal to both the left and right local monitor speaker outputs.

The right MUTE switch controls the mute function on the right hand side of the stereo local monitor speaker outputs.

The SOLO ADD MODE switch allows multiple channel access to the solo busses. When the solo add mode is off the action of pressing a solo switch will cancel any previously active solo. Multiple solos such as stereo left and right signals can be monitored in this mode of operation as long as the solo switches are pressed at approximately the same time. When the solo add mode is on the auto cancelling is defeated which allows multiple channel or output soloing. In this mode input solos have priority over output solos and VCA solos and will temporarily override them. When the input solo is cancelled the output solos or VCA solos will return.



The VCA switches assign the matrix outputs to VCA control from the MASTER module fader.

The INS switches connect the matrix insert return signals to the matrix mixes.


The mute SAFE switches remove the matrix mutes from snapshot automation.

The SPLIT switch changes the matrix AFL solos from mono to stereo.

The MATRIX faders give continuous adjustment of the matrix output levels from +10 dB to off.


The TALK switches connect the matrix mixes to the MONITOR module. When the TALK INTERNAL or GENERATOR INTERNAL are active on the MONITOR module the oscillator, pink noise and talk mic can be routed into the matrix mixes.

The matrix MUTE switches mute the matrix signals at all points after the insert send. The switches can be controlled from snapshot automation.

The SOLO switches send matrix signals to the PFL mono and AFL stereo busses. If a SOLO switch is pressed for a short time it will latch on or off, but, if it is held on for more than 1 second the latching is disabled and when the switch is released the channel solo will turn off. As a default the solo system is auto cancelling so each new solo cancels the last. This function is time dependant which allows several solos to be active as long as they are switched on at approximately the same time, i.e. to solo both sides of a stereo mix press both solo switches at the same time. Alternatively the SOLO ADD MODE switch on the MONITOR module can be used to defeat the auto cancelling and allow multiple channel monitoring. In this mode input channel solos have priority over the matrix solos and will temporarily override them. When the input solos are cancelled the matrix solos will be active again.

Automation


## Assignment control

When the LOCK switch is illuminated, all assignment changes are disabled. The LOCK switch will toggle state on each press. The console will automatically go into a locked state if no assignment controls are operated within 90 seconds.

The VCA, MUTE and FADER switches set the current assignment/display mode.

If the console is in VCA or MUTE mode, the ASSIGN KEYS set the changes for input VCA/automute assignment when its SET switch is operated. These can be selected in the following ways:-

To enter ASSIGNMENT mode
Depressing any single assign key that is OFF will set that assign key on.

In this mode, the assign keys define which groups an input VCA/automute is assigned to when its SET switch is operated. i.e. if VCA mode is active and assign keys one and two are on. Pressing an input SET switch will cause that channel to toggle the VCA 1 and 2 assignments of that input from on to off or visa versa.

To enter CLEAR mode set all assign keys to off.
To switch the assign keys off simply press the ones lit to off. Once in clear mode all the assign keys and the input channel sets switches will flash.

In this mode operating an input SET switch will clear all inputs VCA and /or automute assignments to off.

The AUTOMUTE GROUP MASTER
keys activate the mute circuits on any assigned input channel.
 the virtual faders. It is only active if the console is unlocked to recall level.

The A/B switch selects which system is controlling the console.
For reliability the assignment/automation systems are $100 \%$ duplicated.
The console can operate on either of the systems. All snap shots are stored on BOTH of the systems. The LEDs indicate the status of each system in the following manner.

LED green indicates which system is active
LED off indicates which system is inactive
LED red indicates that a system is damaged or not responding and that a service engineer must be called as soon as possible.

## Automation control

The snap shots stored in the automation system are numbered in ACTs and SCENEs. There is no difference between an ACT and a SCENE apart from its numbering.


The FAST SCENE keys provide the operator with ten quick entry points into the ACT/SCENE sequence. I.e. if FAST SCENE key one is associated with ACT.SCENE 10.02, pressing it will recall ACT.SCENE 10.02.

When a menu function or midi editor is not selected, the ACT/SCENE C/O key selects the act or scene number. The appropriate indication, ACT or SCENE, will be illuminated to indicate this status.

The UP/DOWN keys allow the operator to enter act/scene numbers and navigate through menus.

The LAST, NOW and NEXT keys recall snap shots to the console surface.

NEXT recalls the snap shot numerically proceeding the snap shot that was currently recalled/stored.

LAST recalls the snap shot numerically preceding the snap shot that was currently recalled/stored.

NOW recalls the snap shot that is currently indicated on the numeric display.

The SYSTEM switch gives the operator access to the system menu. Navigation of the menu is achieved by using the UP/DOWN keys to select an entry, CONFIRM to execute a function, enter a sub menu and CANCEL to exit a menu/sub menu.


The system menu comprises: -
LOCK defines the level of console operation.
These are: -
TOTL All automation and assignment functions are disabled RCAL Only recall and assignment functions are available. STOR Scene storage/editing, recall and assignment are operational SYST All functions are available.

All following levels of the menu are only available when the console is unlocked to system level.

MIDI accesses the following MIDI functions
CHIN Sets the MIDI channel for console snap shot triggering. i.e. on receipt of a note on message on this channel number the console will recall the ACT.SCENE contained in the note velocity part of the message.
SOUT Allow all snap shot data to be exported as a MIDI SYSEX dump.
SIN Allow all snap shot data to be imported from a MIDI SYSEX dump

CONS allows several consoles to operate as a single system
NONE The console will operate independently MAST The console will be the master of the system.
SLAV The console will be a slave of the system
CLER Clears all the console snap shot data.

Operating the STORE key will store the current console assignments and settings to the snap shot being displayed on the numeric display.

The MIDI key allows the operator to edit the snap shot MIDI information. On entering this mode the operator will be presented with a menu of the four MIDI messages that are stored within each snap shot, its operation is similar to the system menu.

The COPY, DELETE and INSERT keys allow the operator to edit the snap shot sequence in the following manner.

INSERT operating this key will allow the operator to insert a snap shot at the number on the numeric display. The scene that was originally at the number and all con-current scenes will be renumbered by adding one to their scene numbers.
COPY will copy the snap shot currently displayed on the numeric display to a temporary store. This can then be stored/inserted to/at a new number in the normal fashion. If whilst in COPY mode the operator depresses a FAST SCENE keys the number of the snap shot will be associated with that FAST SCENE key.

DELETE allows the operator to delete the snap shot that is currently being displayed on the numeric display.

The CHECK key allows to operator to preview snap shot on the console surface WITHOUT recalling the setting to the console surface. Whist in check mode the C/O, UP and DOWN keys can be used to step through the snap shots.

## Heritage Menu Overview Ver 1.00

(Key this symbol denotes a CONFIRM button press )

| Menu | Menu $1^{\text {st }}$ | Menu $2^{\text {nd }}$ |
| :--- | :--- | :--- |
| Button | Level | Level |

SYSTEM $\longrightarrow$ LOCK $\longrightarrow \longrightarrow$\begin{tabular}{l}
TOTL <br>
MENU

 

All automation and assignment functions are disabled. <br>
RCAL <br>
Only recall and assignment functions are available.
\end{tabular}

STOR
Scene storage, editing, recall and assignment operation available.
SYST
All Functions are available.

CLR This clears all scene information on the console to give a clean starting point for programming

CONS $\longrightarrow \begin{aligned} & \text { NONE } \\ & \text { SLAVE } \\ & \text { MAST }\end{aligned} \begin{aligned} & \text { No consoles connected } \\ & \text { Console configured as a slave console } \\ & \text { Console configured as a Master Console }\end{aligned}$




MIDI Out Sub Menus:
( $\varnothing$ denotes a number )

| OUTø | $\rightarrow$ CLR |
| :---: | :---: |
|  |  |
|  |  |
|  | PROG $\square \rightarrow$ Chøø (Midi Channel) $\longrightarrow$ Pøøø (Program Change Number) |

## Unlocking the Console:

To unlock the mixing console press the SYSTEM menu button. Using the up/down keys scroll through the menu until LOCK is displayed, press CONFIRM. Using the up/down keys scroll through the menu until the desired level of unlock is displayed on the screen, then press the confirm button.

## Locking the Console:

To lock the mixing console press the SYSTEM menu button. Using the up/down keys scroll through the menu until LOCK is displayed, press CONFIRM. Using the up/down keys scroll through the menu until TOTL is displayed on the screen, then press the CONFIRM button

The LOCK button located on the MODE SWITCHES disables the Assign keys, Mode switches and Set switches on the Centre section and input faders.

## Storing a Scene:

Setting up a scene, Assigning VCA, Mutes, Faders etc.

## Assigning VCA's:

a/ Ensure that the lock button is not illuminated on the mode switches (if it is just press the button to extinguish the LED).
b/ Press the VCA mode button so that it is illuminated. This has now selected the VCA mode on the input modules.
c/ Using the ASSIGN KEY select which master VCA you wish to assign to a particular input module ( 1-10 ). Quickly enabling a button will clear all other buttons enabled so only the one selected is illuminated, push and holding the button down for 0.25 seconds will not disable other buttons previously enabled.
d/On the input channels you wish to assign to the master VCA(s) selected press the SET Button, the relevant LED(s) on the input channel will illuminate. If the SET button is pressed quickly the VCA's selected on the assign keys will be added to those already selected on the channel. If the SET button is pressed and held for a short time then any VCA's already selected on that channel will be cleared and replaced with those selected on the assign key.

## Assigning Mutes:

a/ Ensure that the lock button is not illuminated on the mode switches if it is just press the button to extinguish the LED.
b/ Press the MUTE mode button so that it is illuminated. This has now selected the mute mode on the input modules.
c/ Using the ASSIGN KEY select which Automutes you wish to assign to a particular input module ( 1-10 ). Quickly enabling a button will clear all other buttons enabled so only the one selected is illuminated, push and holding the button down for 0.25 seconds will not disable other buttons previously enabled.
d/ On the input channels you wish to assign to the Automutes selected press the SET Button, the relevant $\operatorname{LED}(\mathrm{s})$ on the input channel will illuminate. If the SET button is pressed quickly the Automutes selected on the assign keys will be added to those already selected on the channel. If the SET button is pressed and held for a short time then any Automutes already selected on that channel will be cleared and replaced with those selected on the assign key.

## Fader Position:

a/ Ensure the Virtual Fader recall mode button is not illuminated and the faders are in normal mode, if this is enabled the new fader position will not be stored.
b/ Move the faders to the desired position.
The only other automated buttons on the console to be set are the Input Mutes, Master VCA Mutes, Group Mutes, Matrix Mutes and Master Mutes.

## Selecting a memory number and storing the memory:

a/ The numbers in the display can be altered as follows. Select either Act or Scene using the ACT/SCENE C/O switch. You will see either act or scene illuminate below the display.
b/ The digits can then be altered between 00-99 using the UP and DOWN keys. This function is looping so if you are on 00 you can go directly to 99 by scrolling down.
c/ To store a scene press STORE in the scene is new then it will just be stored and the screen will display done. If the scene already exists then the display will read over_str and you will need to press the confirm button. The screen will then read done.

## Editing Midi (Program Change):

a/ Pressing the MIDI button places you in the midi menu.
b/ Using the UP and DOWN buttons scroll through OUT01 to 04 until the required message is reached. (On the Heritage there are a maximum of 4 midi out messages that can be sent per recalled scene). Press CONFIRM.
c/ Using the UP and DOWN buttons scroll through the menu until the screen reads PROG. Press the CONFIRM button
d/ The window will read CH 00 , using the UP and DOWN keys a channel between 01 to 16 can be selected. Press CONFIRM.
e/ The window P000, using the UP and DOWN keys a program, change between 00 and 127 can be selected. Press confirm and you will be dropped back to the first level of the menu. When the desired messages have been edited press the MIDI button again to drop out of the menu.
$\mathrm{f} / \mathrm{To}$ then store the midi information with the scene the STORE button must be pressed followed by CONFIRM.

## Inserting Scenes:

a/ Once you have created the scene you wish to insert edit the act/scene number display using the ACT/SCENE, UP and DOWN buttons until the desired position is displayed.
b/ Press the INSERT switch. The screen will then display done. The original scene and any scene preceding it will then be incremented by one position.

NB The INSERT button will only be illuminated if a scene exists where you wish to place the scene, otherwise STORE may be used as normal.

## Copying Scenes:

a/ Recall the scene you wish to copy by selecting the scene number and pressing the NOW button.
b/ Press the COPY button the act/scene numbers can now be scrolled through using the ACT/SCENE,UP and DOWN buttons. Once the desired position is reached the scene is copied to that position when the CONFIRM button is pressed.

NB There is a BUG in the software so that once the COPY button has been pressed the screen is not updated. You can't see were the scene will be placed unless you mentally count the number of up or down button presses, apart from this the information copied is correct.

## A WORD ABOUT THE VIRTUAL FADER MODE:

There are four modes in which the automated faders can operate:
a/ DEFAULT MODE with the VIRTUAL FADER RECALL MODE switch extinguished. In this mode the faders act like any normal console fader. When a scene is recalled and the Mode switch is selected to FADER the LED's along side the input fader wil indicate where the fader should be. The fader should always be moved towards the LED(s). Two LED's illuminated together either mean the fader is very far away from them as it gets closer one should illuminate. If you over shoot then a lower LED will illuminate showing overshoot. When the fader is in position all LED's on that channel should be off.
b/ Virtual Fader Mode with the VIRTUAL FADER RECALL MODE switch flashing. In this mode the faders have no function and will not effect the channel post fader level. The position of the virtual fader will be indicated by the LED's by the side of the fader.
c/ Virtual Fader Mode + Trim with the VIRTUAL FADER RECALL MODE switch illuminated. In this mode the faders will have no function until they are set to 0 dB , at this point the SET LED for that channel will flash. This indicates that the Fader will now sum together with the virtual fader to give a trim of +10 dB to minus infinity. By pressing the VIRTUAL FADER RECALL MODE switch the mode is changed back to virtual mode. Once you have trimmed the channel fader level for however many scenes you wish and the level alteration is no longer required you can switch back to virtual mode. The fader may be returned infinity without effecting the level of the output. The channel will go to the stored Virtual fader level when the next scene is recalled.
d/ FADER SAFE MODE, When this button is enabled on the input channel the fader is taken from automation control and behaves as a normal fader.

The Cross fade is set to $1 / 4$ second and cannot be altered.

## NB fader positions CAN NOT be stored when the Virtual Fader Mode is enabled.

## Previewing a Scene:

To preview a scene without effecting your mix select the scene number on the display using the ACT/SCENE, UP and DOWN buttons. Once the desired number is displayed using the CHECK button the automated switch configuration and fader positions stored for that scene can be viewed without changing the actual settings. Pressing the CHECK button will drop you back into normal mode.

The screen will also display Heritage configuration and Midi data. On $1^{\text {st }}$ issue software the messages scrolled will also read ERR before the Midi data. This is not indicating any error with the scene but is concerned with Heritage external communications, which will be solved shortly. This does not effect any part of the operation of the console.

## Recalling Scenes:

There are 3 methods by which scenes can be recalled:
a/ Stepping through existing scenes using the LAST and NEXT buttons. This steps through the scenes in numerical order.
b/ Select the act/scene number using the ACT/SCENE, UP and DOWN buttons, when the correct scene number is displayed in the screen press the NOW button and the scene will be recalled.
c/ A scene can be assigned to a fast scene key (1-10). In this instance the scene is recalled by just pushing the fast scene key.
a/ Recall the scene you wish to assign to a FAST SCENE KEY.
b/ Press the COPY button, followed by the FAST SCENE KEY button you wish to assign that ACT/SCENE to. The screen will then display done.

## Deleting a Scene From A Fast Key:

a/ Press and hold down the FAST KEY you want to delete.
b/ When the YES and NO button start to flash you can now select either YES or NO to delete or cancel deletion of the FAST KEY.

## Deleting A Scene:

Recall the scene you wish to delete, Or display the scene number on the screen using the ACT/SCENE, UP and DOWN buttons. When this is done press the DELETE button. You will be asked to confirm this. Press the CONFIRM button the screen will then say done when the scene is deleted.

## Midi In Assignment:

In the version 1 software this function is not yet enabled.

## Midi Sysex Dumps:

To store the recall a memory between the console and a midi device ( such as an MDF3 midi Filer) select the SYSTEM menu, using the UP and DOWN keys select DATA and press CONFIRM. Using the UP and DOWN keys select either SAVE or LOAD and press CONFIRM. There are 2 of communication either MIDI (through the midi port rear of the console or via RS232 ( NB software will shortly be available to support this ). Select either MIDI or SERIAL using the UP and DOWN buttons and press CONFIRM. The screen will then indicate the function being carried out and notify the user when finished.



## Rear Panel Left View



Rear Panel Centre View


## Rear Panel Right View

| $0$ | $0$ | O | $0$ |  | O | $0$ | O | ${ }_{2}^{\circ}$ | $0$ | $0$ | ${ }^{\circ}$ | $0$ | $0$ | ${ }^{\circ}$ | $0$ | ${ }^{\circ}$ | $0$ | $\bigcirc$ | $\begin{aligned} & \circ \\ & \\ & \hline \end{aligned}$ | $0$ | $0$ | $0$ | O |
| :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: | :---: |
|  |  |  |  | $0_{0}^{\circ}$ | $0_{0}^{0}$ |  |  |  |  |  |  |  |  |  | $0_{0}^{0}$ |  |  |  |  |  |  |  |  |
|  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |  |

Heritage 3000 Frame Measurements


Weight<br>(out of flight case)<br>$200 \mathrm{Kg} / 440.91 \mathrm{~b}$




HERITAGE 3000 MATRIX MODULE BLOCK DIAGRAM

ISSUE A
DATE 1-5-98




1. The Heritage 3000 is a 30 buss console with an additional $27 \times 8$ output matrix. The busses are as follows:-

| 24 | stereo or mono configurable groups | $=24$ |
| ---: | :--- | ---: |
| 1 | stereo master | $=2$ |
| 1 | mono master | $=1$ |
| 1 | stereo AFL | $=2$ |
| 1 | mono PFL | $=1$ |
|  | TOTAL | $=30$ |

2. The 3000 has 10 automute sub groups and 10 VCA sub groups which include VCA sub group muting.
3. The 3000 has 52 input channels plus an additional 26 direct inputs on the group and master modules.
4. The 3000 has a total XLR input count of 96 as follows:-

52 channel mic inputs
24 group direct inputs
8 matrix bus inject inputs
3 solo bus inject inputs
2 master direct inputs
2 external inputs (2 track return)
1 master bus inject
2 talk mic input
1 talk external input
1 talk int line
5. The 3000 has a total XLR output count of 89 as follows:-

44 input channel direct outputs
24 audio group outputs
8 matrix outputs
3 master outputs
3 solo outputs
6 local outputs
1 talk external output
6. The 3000 has a total of 180 balanced $1 / 4$ inch jacks for inserts as follows:-

52 input channel insert sends
52 input channel insert returns
24 audio group insert sends
24 audio group insert returns
8 matrix insert sends
8 matrix insert returns
3 master insert sends
3 master insert returns
3 local insert sends
3 local insert returns
7. The 3000 has 58 long throw faders for mix control with fader position recall and virtual fader functions.
8. The 3000 has a total of 1043 automated switch functions as follows:-

480 input channel VCA sub group virtual assign switches
480 input channel mute sub group virtual assign switches
48 input channel mute switches
24 audio sub group mute switches
8 matrix mute switches
3 master mute switches
9. The 3000 has a total of 89 peak program meters with 20 LED segments on all outputs and 11 LED segments on input channels.

## Heritage 3000 Technical Specifications.

| Input Impedance | Mic | 2k Balanced |
| :---: | :---: | :---: |
|  | Line | 20k Balanced |
| Input Gain <br> (all faders at 0dB) | Mic | Continuously variable from +15 dB to +60 dB |
|  | Mic + Pad | Continuously variable from -10 dB to +35 dB |
|  | Line Level Inputs | 0 dB |
| Maximum Input Level | Mic | $+6 \mathrm{dBu}$ |
|  | Mic + Pad | $+31 \mathrm{dBu}$ |
|  | Line Level Inputs | $+21 \mathrm{dBu}$ |
| CMR at 100 Hz | Mic (gain at +40dB) | Typ 115dB |
|  | Mic + Pad (gain 0dB) | Typ 80dB |
| CMR at 1 kHz | Mic (gain + 40dB) | $>100 \mathrm{~dB}$ |
|  | Mic + Pad (gain 0dB) | $>60 \mathrm{~dB}$ |
|  | Line | $>50 \mathrm{~dB}$ |
| Frequency Response (20 to 20 kHz ) | Mic to Mix (gain +60 dB ) | +0 dB to -1 dB |
| Noise (20 to 20kHz) | Mic EIN ref. $150 \Omega$ (gain +60 dB ) | - 128 dBu |
| System Noise (20 to 20kHz) |  |  |
|  | Summing Noise (48 channels routed with faders down) | - 80dB |
|  | Line to Mix Noise (48 channels routed at 0 dB , pan centre) | - 75 dB |
| Distortion at 1 kHz | Mic to Mix (+ 60 dB gain, 0 dBu output) | < $0.03 \%$ |


| Crosstalk at 1 kHz | Channel to Channel | $<-90 \mathrm{~dB}$ |
| :---: | :---: | :---: |
|  | Mix to Mix | $<-90 \mathrm{~dB}$ |
|  | Channel to Mix | $<-90 \mathrm{~dB}$ |
|  | Maximum Fader attenuation | $>80 \mathrm{~dB}$ |
| Output Impedance | All Line Outputs | 50 Ohms Balanced Source to drive $>600 \Omega$ |
|  | Headphones | To drive $>8 \Omega$ |
| Maximum Output Level | All Line Outputs | $+21 \mathrm{dBu}$ |
|  | Headphones | $+21 \mathrm{dBu}$ |
| Nominal Signal Level | Mic | -60 dBu to +10 dBu |
|  | Line | 0 dBu |
|  | Headphones | $+10 \mathrm{dBu}$ |
| Equaliser | Hi pass slope | $12 \mathrm{~dB} /$ Oct |
|  | Hi pass frequency | Continuously variable <br> -3 dB point from 20 Hz to 400 Hz |
|  | Treble Gain | $\begin{aligned} & \text { Continuously variable } \\ & +15 \mathrm{~dB} \text { to }-15 \mathrm{~dB} \\ & \text { Centre detent }=0 \mathrm{~dB} \end{aligned}$ |
|  | Treble Shelving Freq. | Continuously variable <br> -3 dB point from 1 k to 20 k |
|  | Treble Bell Freq. | Continuously variable centre from 1 k to 20 k |
|  | Treble Bell Bandwidth | Continuously variable 0.1 Oct. to 2 Oct Centre detent $=0.5$ Oct |
|  | Hi Mid Gain | Continuously variable +15 dB to -15 dB Centre detent $=0 \mathrm{~dB}$ |
|  | Hi Mid Freq. | Continuously variable centre from 400 Hz to 8 k |
|  | Hi Mid Bandwidth | Continuously variable 0.1 Oct. to 2 Oct Centre detent $=0.5$ Oct |


| Lo Mid Gain | Continuously variable <br> +15 dB to -15 dB <br> Centre detent $=0 \mathrm{~dB}$ |
| :--- | :--- |
| Lo Mid Freq. | Continuously variable <br> centre from 100 Hz to 2 k |
| Lo Mid Bandwidth | Continuously variable <br> 0.1 Oct. to 2 Oct <br> Centre detent $=0.5 \mathrm{Oct}$ |
|  | Continuously variable <br> +15 dB to -15 dB <br> Centre detent $=0 \mathrm{~dB}$ |
| Bass Gain | Continuously variable <br> -3 dB point from 20Hz to 400 Hz |
| Bass Shelving Freq. | Continuously variable <br> centre from 20 Hz to 400 Hz |
| Bass Bell Freq. | Continuously variable |
| Bass Bell Bandwidth | O.1 Oct. to 2 Oct <br> Centre detent $=0.5$ Oct |

