

T L Audio

User Manual

FAT TRACK Tube Production Suite

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INTRODUCTION

Congratulations on your purchase of the TL Audio Fat Track Tube Production Suite the complete solution for tracking, summing and monitoring.

Designed to be the centrepiece of any home, project or professional studio, this unit combines some of the best-loved TL Audio features into one compact and professional unit, adding warmth and sonic pleasure to all recordings and mixes. Drawing on our larger and highly acclaimed M series consoles; the Fat Track offers you that unrivalled big desk sound in a more compact, intuitive and affordable package.

The Fat Track can capture your recordings with two premium TL Audio tube preamps as used on our M Series consoles contained within two fully featured channel strips with 3 band swept musical EQ taken from our M1 console. Use the Fat Track for summing and mix-down also, with 4 stereo and two mono returns. You will immediately become aware of the richness and clarity of sound you gain when summing your audio analogue via the Fat Track. You can even bring your entire mix back via the 2 mono channels to add some EQ or final warmth, patch in a compressor to the master inserts and suddenly you have a complete mastering solution too!

The monitor section also comes packed with features, including, separate return solo and selection buttons, effects return, alternative loudspeaker mode, loudspeaker mute, and 2 headphone outputs with individual rotary control. Headphone output 1 (producer) also follows your selections in the master section, whilst headphone output 2 (artist), stays fixed on the stereo bus to enable you to create a separate cue mix for your client.

The multitude of inputs and outputs available within easy reach on the top panel makes for a flexible and intuitive workstation where quick cable changes are made easy. Space for fitting a TL Audio DO-8 ADAT interface is also provided allowing easy digital connectivity.

Designed for the producer, musician, singer/songwriter, engineer or general enthusiast, this product caters for every requirement, is easy to use and produces fantastic, professional results.

PRECAUTIONS

The TL Audio FAT TRACK requires very little installation, but like all electrical equipment, care must be taken to ensure reliable, safe operation. The following points should always be observed:

- All mains wiring should be installed and checked by a qualified electrician,
- Ensure the mains operating voltage stated on the rear panel is correct before connecting to the mains supply,
- Never operate the unit with any cover removed,
- Do not expose to rain or moisture, as this may present an electric shock hazard,
- Replace the fuse with the correct type and rating only.

Warning: This equipment must be earthed.

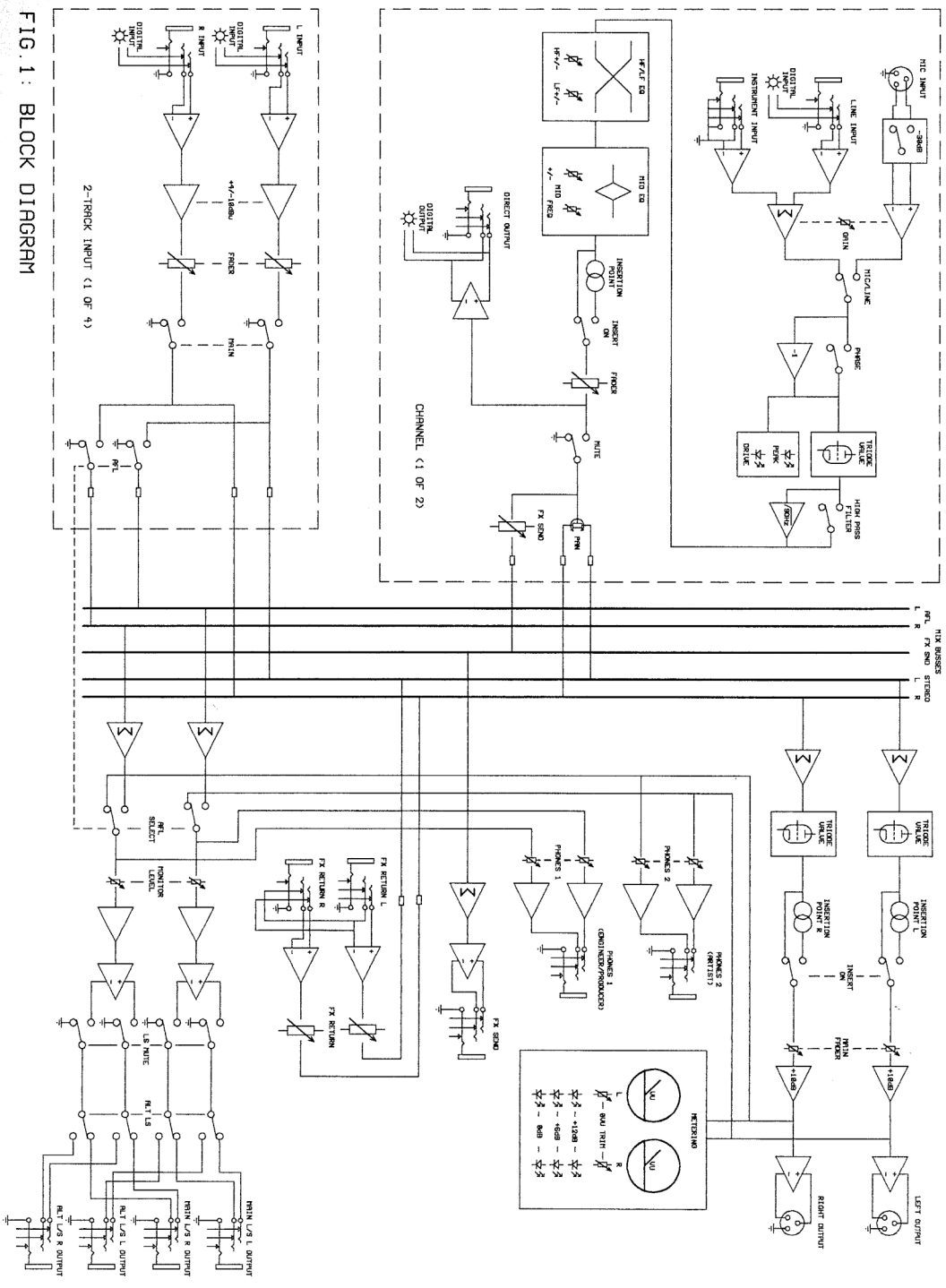


FIG. 1: BLOCK DIAGRAM

FIGURE 1 – BLOCK DIAGRAM

INSTALLATION

AC Mains Supply.

The unit is fitted with an internationally approved 3 pin IEC connector. A mating socket with power cord and mains plug is supplied. All mains wiring should be performed by a qualified electrician with all power switched off, and the earth connection must be used.

Please note that when using this unit with alternate supply voltages (in foreign countries, for example) the internal wiring of the unit may be set for 115V (accepting voltages in the range 110V to 120V, 60Hz AC) or to 230V (for voltages in the range 220V to 240V, 50Hz AC). If the wiring needs to be changed, the unit must be taken to an approved service centre.

Alternate supply voltages may also require a different power cable or mains plug. If in doubt, consult local electrical regulations.

Warning: attempted operation on the wrong voltage setting, or with an incorrect fuse, will invalidate the warranty.

Channel Microphone Inputs.

Each mono input channel has a female, 3 pin XLR connector, for connection to a microphone. The microphone will normally be balanced, but unbalanced connections may also be made as follows:

Balanced inputs:

- Pin 1 = Ground (screen).
- Pin 2 = Signal Phase (“+” or “hot”).
- Pin 3 = Signal Non-Phase (“-” or “cold”).

Unbalanced inputs:

- Pin 1 = Ground (screen)
- Pin 2 = Signal Phase (“+” or “hot”).
- Pin 3 = Ground (screen or connect screen to pin 1 and connect pins 1 & 3 together).

When using unbalanced signals, the signal ground may be obtained by linking pins 1 and 3 in the mating XLR connector. Do not leave pin 3 open circuit, as this may result

in a loss of level and/or increased noise. Good quality screened cable should be used, particularly for microphone or low level sources, to prevent hum or noise pickup.

Channel Line Inputs.

The line inputs are connected via a stereo TRS jack connectors. These may be wired to accept balanced or unbalanced connection as follows:

Balanced inputs:

- Sleeve = Ground (screen).
- Tip = Signal Phase (“+” or “hot”).
- Ring = Signal Non-Phase (“-” or “cold”).

Unbalanced inputs:

- Sleeve = Ground (screen)
- Tip = Signal Phase (“+” or “hot”).
- Ring = Ground (or connect screen and ring together)

Note: When using unbalanced connection on 3 pin plugs, the ring must be linked to the sleeve to avoid possible reduction in level and/or increase in noise. (alternatively a 2 way mono TRS jack may be used)

Instrument Inputs.

A 2 way (mono) jack plug is required, which should be wired as follows:

- Tip = Signal Phase (“+” or “hot”).
- Screen = Ground.

Insertion Points.

The insertion points on the input channels and stereo mix are balanced, with separate send and return sockets. They are “half normalised” i.e. a plug inserted into the send socket (to derive an additional output) will not break the signal path through the channel, but a plug inserted into the return socket will override the signal through the channel.

Mating plugs may be wired for balanced or unbalanced connection as follows:

Balanced sends and returns, (using stereo TRS Jack connectors) :

- Sleeve = Ground (screen).
- Tip = Signal Phase (“+” or “hot”).
- Ring = Signal Non-Phase (“-” or “cold”).

Unbalanced send and returns: (using stereo TRS Jack connectors)

- Sleeve = Ground (screen)
- Tip = Signal Phase (“+” or “hot”).
- Ring = Ground (screen)

(Alternatively a mono TRS Jack may be used for unbalanced signals wired hot to tip, screen to sleeve). Balanced connection is always preferable in an audio system, where the equipment being connected supports it. It is possible to mix balanced and unbalanced connections, for example using unbalanced sends and balanced returns, with no loss in level.

Direct Outputs, L/S Outputs and FX Send.

These outputs all use balanced TRS jack connectors, which may also be wired for balanced or unbalanced connection as follows:

Balanced outputs:

- Sleeve = Ground (screen).
- Tip = Signal Phase (“+” or “hot”).
- Ring = Signal Non-Phase (“-” or “cold”).

Unbalanced outputs:

- Sleeve = Ground (screen)
- Tip = Signal Phase (“+” or “hot”).
- Ring = Ground (screen)

Unbalanced Outputs

A 2 pin (mono) jack plug is required, which should be wired as follows:

- Tip = Signal Phase (“+” or “hot”).
- Screen = Ground.

Phones and Tape Outputs.

These outputs are unbalanced, stereo signals, on TRS jack connectors. The mating connectors should be wired as follows:

- Sleeve = Ground (screen)
- Tip = Left channel
- Ring = Right channel.

Balanced XLR Outputs.

The outputs are via balanced, 3 pin male XLR connectors. The mating connectors should be wired as follows:

- Pin 1 = Ground (screen),
- Pin 2 = Signal Phase (“+” or “hot”),
- Pin 3 = Signal Non-Phase (“-” or “cold”).

If an unbalanced output is required, pins 1 and 3 should both be connected to ground.

Ventilation.

The unit generates a moderate amount of heat internally, which should be allowed to dissipate by convection through the grills in the front panel, which must not be obstructed. Do not locate the unit where it will be subject to external heating, for example, in the hot air flow from a power amplifier or on a radiator, or in direct sunlight. Ensure that air is able to circulate freely around and under the unit. The Fat Tracks power supply and power transformer has been mounted to the back of the unit to dissipate the heat created by these parts away from the main console and circuitry and ensuring that the unit stays cooler and maintains longevity.

The FAT TRACK has been designed for operation free standing on a tabletop.

OPERATION.

Input Selection

You must ensure that the correct input connector, mic, instrument or line, is being used from the required source. You can select the type of input by pressing the 'LINE' switch at the top of each channel strip in or out. In the "up" position the mic input is selected; depressing the 'LINE' switch selects the line input or instrument input as the source. Note that you can have mic and a line / instrument input connected at the same time but you will only be able to select a single input at a time. (NB do not have anything connected to the line input when you have an instrument plugged into the instrument input unless you want to combine 2 signals on the channel. Not recommended)

Channel Gain Control

The gain control should be set to obtain the best signal to noise ratio, whilst preserving adequate headroom. (i.e. as high as possible without causing signal distortion) Any changes in gain level should be gradual to avoid sudden overload or severe distortion. Extra care should be taken with higher level inputs, and high gain settings should be used sparingly to avoid an increase in the noise floor and the introduction of distortion.

Line Input

A typical line input will require less gain than a microphone signal, as it is a 'hotter' source. You should have the gain set to a minimum when connecting a line input, then gradually increase it to achieve the required level.

Instrument Input

A typical instrument level input will be lower in level than a line level signal but you should still exercise caution with the gain, starting at a minimum setting and slowly increasing until desired setting is reached. When connecting an instrument like guitar, you can connect straight into the Fat Track without the need for an additional DI unit or amplification stage, the Fat Track's input stage acts as both your DI and preamp.

Microphone Input

When connecting a microphone to the Fat Track, start with the gain set to a minimum value and gradually increase until the sound source is just tickling 0dB on the meters when the source sound is at average output level. You can then choose how much you would like to drive the valves to produce the required sound. If Phantom power is

needed, activate it before setting the gain by depressing the button marked '48V' on the channel strip. Although the Fat Track provides 48V of phantom power, microphones requiring less than 48V can still be connected without any difficulties.

Phantom Power.

+48V phantom power is available at the mic socket, selected by a front switch. Phantom power should only be used in conjunction with suitable condenser microphones. (Phantom power will not damaged other types balanced microphones apart from ribbon mics, if you have no unbalance or ribbon mics it is usually OK to leave the phantom power on subject to the caution below)

CAUTION: Operation of the phantom power switch, or plugging a microphone in or out with phantom power applied, may cause a click or thump in your loudspeakers. To prevent this happening, ensure that the system gain is set to minimum (e.g. on your mixing console fader or power amplifier), before operating the switch or plugging in a microphone.

Phase Reverse

The phase reverse switch is used to invert the phase of the input signal. It is active on both mic and line inputs. This function could be required when processing a signal that is out of phase with other signals in a mix, in which case the resultant phase error typically appears as a loss of low frequency content, due to cancellation of out of phase components. Phase reverse is commonly used when recording the bottom of a snare drum (if also using a mic on top of the snare), the back of a guitar cab (if also recording signal from the front of the cab), and when performing the stereo recording technique known as an MS pair. If 2 sources of the same audio signal are mixed together in mono and one is out of phase with the other (e.g. pins 2 and 3 are wired incorrectly on one of the inputs) then cancellation will occur. To demonstrate this plug a stereo mix into the 2 channels pan 1 left and 2 right and play back – now select the phase reverse switch on one of the channels, no cancellation yet but you should detect a loss of bass and a loss in the stereo imaging, you may also feel a slight pressure in the ear but you will hear the whole of the stereo recording. Now put both pans to the centre and the whole centre of the mix will disappear, you will only hear those parts of the signal which were panned fully left or fully right in the mix, the part of the stereo mix which was common to left and right, the singer for instance, and bass guitar and bass drum etc. will all have disappeared. Out of phase signals can be used to great effect sometimes but NEVER if you are mixing for vinyl. It is not possible to cut a vinyl master if the final mix contains out of phase signal.

High Pass Filter

This low cut filter provides 12dB per octave of gain reduction with the -3dB point being at 90Hz. Like the phase reverse switch, the high pass filter is active on mic, instrument and line inputs, and is ideal for removing low frequency rumble. The filter can be useful in restricting 'popping' on vocals or even low frequencies caused by

contact with microphone stands or microphone cables. Popping is an undesirable thump that is caused by close-miking certain spoken or sung letters, namely 'P' or 'B'. These particular letters cause a sudden expulsion of air that can result in an audible thump. As this thump has a lot of low frequency content the high pass filter can help to reduce the problem, as can using a pop filter (a device usually made out of nylon material similar to stockings) suspended in front of the microphone. The low cut filter is easily bypassed for quick A/B comparison.

30dB Pad

Occasionally when using sensitive condenser microphones the source signal may be too loud for the input preamp. In this situation, to avoid any overloading or distortion of the mic preamp stage, the 30dB pad can be used to reduce the input gain to a more manageable level. The 30dB pad only applies to the microphone input.

Equalisation

Before switching the EQ into circuit, it is advisable to set the cut/boost controls to their centre, or flat, position. The EQ is brought into circuit with the 'EQ ON' push switch, located at the top of the channel strip. Each channel has three bands of equalisation:

- 1/. shelving wide band low frequency (LF) (i.e. it extends from 80Hz to the extreme low frequency limit of the equaliser's response.),
- 2/. peaking sweepable mid (i.e. it boosts or cuts a section of the audio spectrum around its selected centre frequency only)
- 3/. shelving wide band high frequency from 12 kHz to the extreme high frequency limit of the equaliser' response (HF).

The LF shelf operates at a frequency of 80Hz, while the HF shelf is set at 12kHz. The EQ slopes have a second order 12dB/octave response, and an associated gain control on both bands provides up to 15dB of cut or boost on each selected frequency, controlling the full range of frequencies below the LF corner frequency and above the HF corner frequency. The mid band has a fixed Q of 0.7 allowing +/- 15dB cut or boost over a moderately broad band of frequencies. The centre frequency is selected via the dedicated variable frequency control, and the required amount of cut or boost is applied using the associated gain control. The range of frequencies selectable is 150Hz to 7kHz.

When listening to a mix, it is generally frequencies between around 40Hz and 12-14kHz that the ear is most sensitive to and a distinct lack of any one frequency in between these figures can result in a mix that may lack 'fullness'. Generally, a pleasant mix will contain a good balance of frequencies from across the spectrum. Each part of the frequency spectrum does however contribute its own unique characteristic. For instance, if there is too much low frequency content below around 300Hz, the mix can sound 'boomy' or 'muddy'. Too little presence from these frequencies however and the mix can sound 'thin' and lacking 'power' or 'energy'.

Too much energy in the low mids (around 300Hz – 2kHz) can cause a really ‘middy’ sound which lacks drive and excitement. The mid ranges are very important, as it is within these ranges that your vocal and lead instruments will be most prominent so it is essential that you get a good balance. Finally, towards the top of the listening spectrum, frequencies here are generally adding sparkle and a sense of space to your overall mix. Lack of energy from as low as 2kHz can cause the mix to sound dull whilst too much top and your mix will sound overly bright. But take care with the top end as most of the hiss associated with recording is within the range of the high shelving (HF), to reduce excessive noise it is better to record with too much top and wind it off when mixing than vice versa.

You should frequently A/B the sound you are Equalising by pressing and depressing the EQ ON button; it is astonishing how quickly the ear can adjust to changes. By making frequent comparisons you can make sure you do not stray from the required sound. It is also worth spending some time getting used to the EQ on the Fat Track - it sounds great and you'll be amazed how versatile and musical it can be.

Insertion Points

The insert points are configured to be post the input amplifier with its valve stage, but pre EQ and fader. Typical applications include the connection of a compressor (for example the TL Audio Ivory 5021) into the channel, or perhaps a gate (usually for percussive sounds) or a dedicated outboard equaliser (e.g. the TL Audio Ivory 5013) for extensive EQ correction or effect.

The inserted device can easily be switched on and off for A/B comparison and flexibility via the ‘INSERT ON’ button at the bottom right of each channel strip.

As the insert points are balanced, you could also use the insert send as an additional direct output. This would give the benefit of a short path audio circuit, but remember that the insert send is pre EQ and pre Fader.

Effects Send

Effects sends are available on each channel to provide you with the ability to send signal to an effects processor (typically a reverb).

The effects send rotary knob controls the amount of signal that is sent from the channel to the effects unit. Playing with the amount of signal you send and the level you set the Effects return knob at will greatly differ the severity of the chosen effect.

Direct Output

Direct outputs are provided from both channels to connect directly to your computer soundcard or recording device via balanced analogue jacks. The direct output is post Insert, EQ, and Fader but pre Mute.

Pan Control

The pan control positions the image within the stereo field, from fully left in the anticlockwise position, through centre at the dented position to fully right in the clockwise position. The gain law is -3dB at the centre.

Mute Switch

The channel may be muted or switched on without affecting the level set on the channel fader.

Channel Fader

A rotary fader is located at the bottom of each of the input channels, and sets the level of the channel's signal being fed in to the stereo mix bus or to the direct output. The fader provides up to 10dB of gain at its highest point.

Tube Stage Drive and Peak LEDs.

The yellow LEDs indicate the drive level to the tube, or valve, stage. The Drive LED is a variable intensity indicator, starting to glow when the tube harmonic distortion is around 1%, and being fully illuminated at approximately 5% harmonic distortion. The distortion generated by the tube stage is predominately second harmonic, which is responsible for the characteristic valve “warmth”. The red Peak LED operates as a conventional warning that clipping is about to occur. The operating level of both input and post fader stages is monitored, and the LED illuminates at a threshold of +19dBu, when there is less than 7dB of headroom remaining to the direct output. Normal operation would be to set the input gain so that the Drive LED is illuminating regularly, with occasional illumination of the Peak LED occurring on loud transients.

Stereo Rotary Return Channels

4 stereo return channels are provided for routing multiple outputs in via your computer soundcard / DAW into the Fat Track for summing and/or mix down. The rotary level control enables you to control the level of the input signal or create a basic mix of the levels coming in. For work that has already been mixed but you would like to just sum the audio through the Fat Track for optimum sound and fantastic results, it is quite normal to have these rotary return levels set to their maximum (10). Please note you can use the mono channels as input returns too, via the line input sockets on each channel; this gives you a maximum of 10 return paths (4 stereo and 2 mono). All stereo return sockets are selectable between +4 and -10dB.

Main Select Buttons

The 'MAIN' switch located in the centre of the unit just above each corresponding rotary stereo return knob (A,B,C,D) activates the return and routes it into the stereo bus. To activate the main switch it needs to be pressed to its downward position. The 'MAIN' switch can be used as an 'ON' switch to route the stereo signal to the stereo bus, or as a 'MUTE' switch when it is de-selected.

Main Fader

The main rotary fader controls the main output level and also drives the final tube dual triode ECC83 tube stage on the master output.

Effects Return

The Effects Return Rotary knob is located above the Main fader and the headphone levels, and controls the level of the signal coming back from your effects processor to be fed into the stereo mix. Playing with the amount of signal you send and the level you set the Effects return knob at will greatly differ the severity of the chosen effect. As it is a stereo return that feeds the stereo bus you could also use the return as an additional stereo input when summing or mixing.

Master Insertion Points

Balanced inserts are offered in the master section too. This enables you too connect a piece of hardware like a stereo compressor or limiter for an easy mastering solution.

The inserted device can easily be switched on and off for A/B comparison and flexibility via the 'INSERT ON' button just above the main rotary fader control.

Phones Level 1 - Producer

This rotary control controls the level sent to headphone output 1 (producer). It is fed from the monitor section and will follow what is selected within that section. This means that by default it will monitor the stereo bus/ main mix, unless you select one, or any number of the stereo return (A-D) buttons and it will then switch to monitor what has been soloed.

Phones Level 2 – Artist

Phones level 2's (artist) rotary control, controls the level of the signal fed to headphone output 2. This output always feeds from the stereo bus / main mix even when returns are in solo mode. This means headphone output 2 is ideal for the artist or client to listen to for a constant, uninterrupted listening experience.

Monitor Level

This ergonomically satisfying knob controls the volume of the monitor speakers connected to the Fat Track.

Alternative Loudspeaker

By selecting the 'ALT LS' switch in the master section it allows you to switch your monitoring to an additional set of monitor speakers . This is great for A/B comparisons and for switching between near-field and mid-field monitors. A great feature for mixing and mastering.

Loudspeaker Mute

This switch mutes the monitor and alternate LS outputs. This is particularly useful when using headphones, enabling the main monitors to be muted without needing to turn off their amplifiers. It is also useful for temporarily muting the monitors without needing to alter any fader levels, for instance during a telephone call.

It is good practice to mute the outputs when switching the Fat Track on or off, to avoid any loud "thumps" in the loudspeakers.

Monitor Source

Located in the master section on the far right of the console, these 4 buttons labelled A, B, C, D allow you to solo the corresponding stereo return, you can select multiple returns to solo at any one time, in any combination, and once all buttons are deselected, you will automatically switch back to monitoring the 'main' stereo bus.

Output VU Meters and Peak LEDs.

The two VU meters monitor the level on the stereo output. The meters are factory calibrated for 0VU = +4dBu on the balanced XLR outputs (-10dBu on the unbalanced outputs). The calibration may be changed by inserting a small screwdriver through the holes labelled "0VU" in the front panel underneath the Peak LEDs. There is a 3 segment LED meter adjacent to each VU meter. The LEDs indicate peak levels of 0dB, +6dB and +12dB. The LED calibration is fixed, and not affected by changes to the VU meter calibration.

Digital Output Options

The optional DO8 card can be fitted to the back of the Fat Track, providing an 8-channel ADAT lightpipe I/O interface. This professional protocol makes for easy connection when conducting computer-based recording or when connecting to other digital equipment. It provides you with digital access to channel direct outputs, master stereo output, 4 stereo returns and both channel line inputs, a total of 10 inputs and 4 outputs.

A button is located at the top left hand side of the console amongst the I/O sockets that enables you to configure the digital signal. When the button is in its up position, digital inputs 7&8 are routed to stereo return 'D', when the button is in its downward position, digital inputs 7&8 are routed to channels 1&2. This gives you access to all 10 channels with a push of a button.

A Final Word

We are confident that you will enjoy this new addition to your set up and may soon wonder how you use to manage with out it!

If you do have any questions regarding this product or operation, contact us info@tlaudio.co.uk or +44 (0) 1462 492 090.

TIPS ON OPERATION

Recording a Vocal to your DAW via the Fat Track

To set up the Fat Track for a basic recording of a single vocal, you can use the following instructions as a simple guide to get started. It is up to you how you wish to use and operate the equipment, and the following is meant as a guide only, highlighting one of many possible ways in which to record.

1. Connect your microphone cable to the 3-pin XLR connector at the top of one of the channel strips, and check the Mic / Line switch is set to the 'MIC' position. If you are using a condenser (capacitor) microphone you should also press the 48V phantom power switch - please ensure that you have the gain and master volume down when you activate this switch as turning on the Phantom Power can cause a loud pop which could potentially damage your monitor speakers. Make sure you are using a balanced microphone cable.
2. If you are recording in a room where the acoustics are not ideal or there is background noise / electrical hum, you may find it useful to activate the 90Hz High Pass Filter, this will get rid of some of the unwanted noise from your vocal recording
3. Test the input signal level by asking your vocalist to warm up while you gradually step up the gain until you are getting an adequate level coming through, you can monitor this via the analogue VU meters and by being aware of the LED indicators. If you are recording straight to your DAW via the direct output on the channel, you should also check your signal input level on your chosen music software. Please note that on the Fat Track you can push the channels and have the levels 'tickling the red' as it is a fully analogue device, however, on your software interface you should never let it clip in the red as it would result in a square wave digital clip. Once you think you are happy with the level, ask your vocalist to sing the loudest and quietest part of the song and make any necessary adjustments. The most important thing is to avoid clipping, quiet parts add to the dynamic range of a vocal and can easily be evened out via compression at the mix stage. If you are experienced in recording you may find it better to use a compressor while recording as this will control the dynamic range of the signal and make distortion at the DAW less likely.
4. To listen to the stereo mix you are tracking to, bring back the main output of the DAW / soundcard to one of the stereo returns (A-D) and then set headphone levels to a comfortable volume for your vocalist and yourself. Obviously there are other ways you can configure your set up and monitoring due to the flexible I/O and routing / monitoring available on the Fat Track.
5. Now you are ready to go, so simply press record in your DAW and let the vocalist do their work.

Summing via the Fat Track

One of the big audible benefits you will be aware of with the Fat Track is its ability to sum your sounds together in analogue to produce a more detailed, open, rich, warm and dynamic sound.

Summing is basically another term for what is essentially mixing, but in its most basic form. Summing, as the name would suggest is the process of adding several different signals together into one combined signal.

In software, the summing is done digitally using algorithms to mathematically add the individual parts including all note, pitch, volume, control and effects information. This large calculation can mean that the result is not always that accurate or that some information may become lost, confused or blurred during this process.

Analogue summing has long been proven as a more reliable and pleasant sounding way of adding signals together and the Fat Track lets you do this very easily.

The following is a basic guide as to how you can use the Fat Track to sum your audio. In this example we presume you are summing from a DAW.

1. Once you have finished your mix in your software of choice, start to group your mix via type of instrument (i.e. 'vocals', 'percussion', 'rhythm', 'lead' etc), grouping the individual channels to their assigned bus within your software. This creates mixed stereo stems of all the components of your track.
2. For mono signals that you may not want to group (like lead vocal or bass guitar) these can be kept in mono and either fed out from the channel or a mono bus output.
3. You can then connect up to 10 (4 stereo and 2 mono) channels to the Fat Track depending on the number of outputs you have from your soundcard / audio interface. If you have ADAT connections on your soundcard, you could also use the TL Audio DO-8 interface with the FAT Track to enable you to access the outputs over ADAT (see details in operation).
4. Use the level controls for the stereo returns to make any adjustments in level between the different groups of instruments.
5. If you want to add any dynamics, like compression to your mix, you can insert them via the balanced inserts provided on the master section.
6. You can then take the output from the Fat Track back into your soundcards inputs to record the Fat Tracks analogue sum of your mix back into your software (beware to make sure you do not have the summed signal going back in to your software, being fed back into the Fat Track – otherwise a feedback loop may occur).

Warming up a stereo mix via the Fat Track

Sometimes you may just want to inject some life into, or warm up your final stereo mix. It is very easy to do this with the Fat Track, giving you a great sounding back end to your DAW or general recording set up. Having a quality analogue back end to your system can put soul and life back into your recordings, taking away that clinical, dull sound.

You do not have to make it sound ‘coloured’ though, run your mix through without any added EQ to simply open up your sound giving more clarity and definition. Or you can use the musical, British EQ section to add some warmth, sparkle and depth to your mix.

If you are not sure where to start with EQing a stereo mix, it may be useful to start with the ‘smile curve’ EQ boost. This principal is used in many Consumer HiFi and Pro Audio Exciter effects to enhance the sound. The curve consists of a small boost in the low, a small boost in the high, and either a small cut or a flat response for the mid.

By all means play around with the EQ on the Fat Track to create your desired sound, but remember only small changes should be made to a finished mix otherwise your result may become imbalanced. Less is more! The EQ section on the Fat Track is taken from our M series consoles and is very smooth and musical; this means that you can push your settings a little further whilst still maintaining a pleasant sound.

1. Connect your stereo mix from your source (DAW / Master Tape) via the line inputs on each channel
2. Make sure you have the channels switched to ‘LINE’ and also have the pan controls turned ‘LEFT’ and ‘RIGHT’ for each side of your mix.
3. You may then add any EQ cuts or boosts that you require, it is sensible to frequently A/B your EQ changes by switching the ‘EQ ON’ button in and out for comparison. Your ear can very quickly get used to EQ changes so it is always wise to keep referring to your original.
4. There is the option of using the Fat Track as a complete mastering device too, by patching in a stereo compressor / limiter to the balanced inserts on the master bus. Again you can A/B your changes from the Fat Track directly, by using the ‘INSERT ON’ button to switch the dynamic processor in and out.
5. When you are happy with your sound, you can then simply take the stereo output (either main balanced, main unbalanced, or tape out) from your Fat Track to your main recording device.

SPECIFICATIONS

2 Input Channels (figures quoted to Direct Outputs with Fader at maximum):

Line Input:	Balanced, via 3 pin TRS jack. Nominal Level +4dBu, Gain range +/-20dB. Frequency Response 10Hz to 40KHz, +/-0.4dB. Maximum Input Level +22dBu. Noise -83dBu unweighted, 22Hz to 22KHz.
Mic Input:	Balanced, via 3 pin XLR. Gain range +16dB to +60dB (-14dB to +30dB with pad). Input Noise (EIN) -127dBu, 22Hz to 22KHz @ 60dB gain, Frequency Response 10Hz to 40KHz, +/-0.5dB, @ 40dB gain. Maximum Input Level +6dBu (+36dBu with pad). Switchable 48V Phantom Power.
Instrument Input:	Unbalanced 0.25" jack. Nominal input -16dBu, Gain range +/-20dB. Maximum Input Level +8dBu. Frequency Response 10Hz to 40KHz, +/-0.5dB. Noise -83dBu unweighted, 22Hz to 22KHz.
Direct Output:	Post EQ and Fader, Pre Mute Switch. Balanced, via 3 pin TRS jack. Nominal Level +4dBu. Maximum Level +22dBu.
Mono Inputs:	Phase Reverse Switch. Switchable High Pass Filter, -3dB @ 90Hz, 2nd order.
Equaliser:	"High" +/-15dB @ 10KHz. "Mid" +/-15dB, swept 150Hz to 7KHz, Q =0.7. "Low" +/-15dB @ 80Hz.
Tube Stage:	Pre EQ.
Drive LED:	Variable intensity, from 0dBu to +12dBu, indicating increasing Tube drive. Distortion is predominately 2nd Harmonic, increasing with Drive LED from 0.5% to 3% (typical).
Peak LED:	Illuminates at +19dBu. Multiple point monitoring at Input stage and post EQ.
Insertion Point:	Post EQ, Pre Fader.

Semi-balanced Send, balanced Return, on separate 3 pin TRS jacks.
Bypass switch.
Nominal Level -2.7dBu.

FX Send: Post EQ, Fader and Mute switch.
Variable send level.

Routing: To Main L+R stereo mix, via pan pot (-3dB at centre) and mute switch.

Four 2-Track Returns:

Fader: Rotary, 0dB maximum gain.

Inputs: Balanced, via 3 pin TRS jacks.
Nominal Level individually switchable +4dBu/-10dBu.
Maximum Input Level +22dBu.

Routing: To Main L+R stereo mix, via Fader and routing switch.

Main Outputs:

Fader: Rotary, +10dB maximum gain.

Tube Stage: Post Mix Amps.

XLR Outputs: Balanced, nominal level +4dBu.
Maximum level +26dBu.
Noise, all faders @ 0dB, -80dBu, 22Hz to 22KHz.
Noise, Main Fader @ minimum, -95dBu, 22Hz to 22KHz.

Jack Outputs: Unbalanced, nominal level -10dBu

Tape Output: Combined TRS jack, nominal level -10dBu.

Insertion Point: Pre Fader.
Semi-balanced Send, balanced Return, on separate 3 pin TRS jacks.
Bypass switch.
Nominal Level -2.7dBu.

FX Send:

Output: Mono, semi-balanced, via 3 pin TRS jack.
Nominal level -6dBu.
Noise -87dBu, channel sends @ maximum.
Noise -100dBu, channel sends @ minimum.

FX Returns:

Inputs: Stereo, Right Input normalled to Left Input for Mono compatibility.
Balanced, via 3 pin TRS jacks.
Nominal level +4dBu.
Maximum Input Level +22dBu.

Monitoring:

Source: Normally follows Main Output, overridden by 2-Track AFL selection.
Level controlled by large diameter knob.
Main / Alt LS switch.
LS Mute switch.

LS Outputs: Main and Alt LS, balanced on 3 pin TRS jacks.
Nominal Level +4dBu.
Maximum Level +26dBu.
Noise -80dBu, all faders @ 0dB.
Noise -100dBu, Monitor Fader @ minimum or LS muted.

Phones Outputs: 2 stereo outputs, with separate level controls.
Phones 1 follow Monitor selection, for Engineer/Producer.
Phones 2 follow Main Output at all times, for Artist.
3 pin TRS jacks.
Minimum Impedance 32 ohms.

Digital I/O: Optional DO-8/Firewire cards available, to fit slot on rear panel.

Dimensions: 381mm W x 330mm D x 150mm H. (15" x 13" x 6" approx).

Power Requirements:

Voltage: 230V 50Hz or 115V 60Hz, internally wired.

Consumption: 37VA, without digital I/O.
41VA, with DO-8 option fitted.

These specifications are typical figures, subject to normal production tolerances. They do not represent guaranteed limits for any particular piece of equipment. Specifications are subject to change without notice.

SERVICE

Should the unit require service, it must be taken or posted to an authorised dealer with a description of the fault. Please retain the original packing for possible future use, and ensure the unit is suitably protected during transit. The manufacturer cannot accept responsibility for damage caused during transportation.

This equipment is supported by a limited warranty for a period of one year from the date of purchase. During this period, any faults due to defective materials or workmanship will be repaired free of charge. The warranty excludes damage caused by deliberate or accidental misuse, tampering, operation on the incorrect mains voltage, or without the correct type and value of fuse fitted. It is the user's responsibility to ensure fitness for purpose in any particular application. The warranty is limited to the original purchase price of the equipment, and excludes any consequential damage or loss.

Please record the following details, and retain proof of purchase date:

Serial Number.....

Date purchased.....

Dealer.....

TL Audio Limited, Letchworth, England.

Tel: 01462 490600. International +44 1462 490600

Fax: 01462 490700. International +44 1462 490700

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