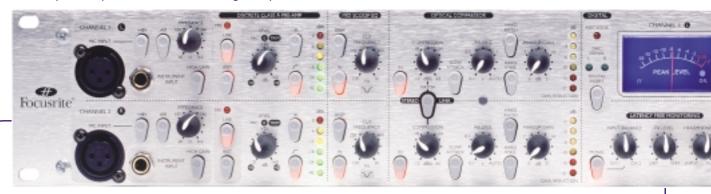


TWINTRAK PRO

Lose the Mixer

Designed principally as an affordable, high-specification dual mono / stereo tracking device, TT Pro, (as it's affectionately known,) also boasts comprehensive stereo latency-free monitoring for direct and delay-free mix control, and unmatched digital connectivity, including a D-A as standard, all within a sturdy 2U 19" rack-mount chassis. The award-winning Class A pre amps are capable of capturing every nuance from any source, whilst the optical compressor, with its custom fast-acting optos, comprehensive controls and true stereo linking, provides extremely smooth yet effective control of the signal's dynamics.



- 2 Class A mic pre's Dual mono/stereo compressor Stereo latency free monitoring
- Dual 'scoop' eq D-A Converter as standard Twin output meters

Combining all these facilities, TT Pro provides everything you need to track, monitor and mix, leaving your mixer gathering dust.

- Two Class A Focusrite mic pre-amps,
- Wire-wound, inductor-induced 'Air' and controllable mic input impedance
- Dual-mono compressor, featuring custom fast-acting optos and true stereo linking
- Comprehensive stereo latency-free monitoring with enhanced routing and control
- High quality 24bit D-A, fitted as standard
- Optional high quality 24 bit/96kHz A-D, including word clock
- Two mid-scoop EQ's
- Post pre-amp inserts and compressor side-chain inserts
- Custom Focusrite Ocean Blue peak-reading meters.
- Front panel direct instrument inputs

Two Class A Pre-amps

TT Pro boasts two award-winning pre-amps, but with some exclusive new improvements; 'Air' utilises a unique wire-wound inductor circuit that imparts a boost in the high-end frequencies, enabling the pre-amp to recreate the open and airy qualities of more expensive transformer-based pre-amps.

TT Pro also sports variable impedance control, allowing you to perfectly match the pre-amp with any microphone, (and so maximise level,) or to use different settings creatively, to interactively shape the sound of your chosen microphone.

The Pre-amps demonstrate the same widebandwidth philosophy featured in original Focusrite ISA units from years past, ensuring low noise and distortion, whilst displaying the signature sound for which Focusrite has become famous. Also in this section are Phantom Power, Phase reverse and a High Pass/Low Cut filter for controlling pops, rumble and proximity effect, enabling TT Pro to handle any mono or stereo audio source with ease.

Direct Instrument Input

Also included on the front panel are two instrument inputs, allowing quick and easy connection without the need for a separate DI box. A High Gain switch provides the option to boost the incoming signal by 20dB, allowing you to accurately set the input level for both passive and active instruments.

Mid Scoop EQ

The Mid-scoop EQ is a quick and effective problem-solver, with two different depths of cut and an adjustable frequency control. Ranging from 120Hz to 2K, it's ideal for vocal problem solving or for creating great miked-up cabinet ambience.

Optical Compressor

In pride of place is TT Pro's powerful dual mono/stereo optical compressor. It features Focusrite's custom fast-acting optos and operates in Class A mode, allowing for smoother, lower distortion performance than off-the-shelf VCA-derived circuits. With the addition of both Hard/Soft knee switches and side-chain inserts, every compression need is provided for.

True Stereo Linking

The link feature allows TT Pro to act as a true stereo compressor, enabling complete stereo control of every parameter. When switched to 'link', channel one becomes the master and controls both compressors simultaneously. Unlinking allows you to operate the unit as a dual mono device, processing two separate channels independently.

A Hardware Plug-in.

TT Pro is the first Platinum to feature a D-A converter as standard, alongside its optional A-D converter. The D-A converter enables the user to feed mono or stereo digital signals into TT Pro. These signals can be fed either into the stereo Latency-free Monitoring section, or via line inputs, (post- the pre-amp stage) for further processing as a 'hardware plug-in'.

The signal can then return to the digital audio workstation via Focusrite's optional stereo 24bit/96kHz A-D converter. The converter is 24 bit and 128 x over-sampled, handling sample rates of up to 96kHz, with S/PDIF output and a BNC word clock connector. An ADC lock LED on the front panel shows when Word Clock sync has been established, whilst two DAC input LED's show when signal is flowing through the D-A converter.

Together, these facilities make TT Pro an ideal tracking device, the perfect analogue interface for any digital audio workstation, and equally the world's first 'hardware plug-in'.



"This really is one serious piece of kit that's been well thought out to make sense in the real world of desktop recording."

Paul White, Sound on Sound



Stereo Latency-Free Monitoring

This section offers the capacity to monitor your source signals within the mix, directly at source, avoiding having to resort to monitoring via your DAW or digital desk, both of which may be prone to latency issues. A headphone or monitor mix can be created for either the artist or the engineer, with immediate and tactile control over levels. The opportunity to feed in and control an external effect (e.g. reverb) for the source signals is also available. With all the controls for latency-free monitoring available direct from the front fascia, TwinTrak Pro spells an end to all your latency nightmares.

TWINTRAK PRO SPECIFICATIONS

Inputs			
MIC			
Gain	OdB to 60dB continuously		
_	variable		
Frequency Response	OdB at 10Hz and -2dB		
	down at 200KHz.		
Mic EIN	-128dB (measured at 60d		
	of gain with 150 Ohm		
	terminating impedance		
	and 20Hz/22KHz bandpa		
	filter)		
THD + N	0.0004% (measured with		
	+16dBu input signal and		
	with a 20Hz/22KHz		
	bandpass filter.)		
LINE	40 ID 40 ID		
Gain	-10dB to +10dB		
_	Continuously variable		
Frequency Response	0.1dB down at 10Hz and		
	-3dB down at 200kHz.		
THD + N	0.0006% (measured		
	with OdBfs (+22dBu)		
su ·	input)		
Noise	-94dBu (measured		
	with a 20Hz/22kHz		
INSTRUMENT	bandpass filter.)		
	OdB to 40dB		
Gain	Continuously variable		
	0.5dB down at 10Hz and		
Frequency Response	-1dR down at 200kHz.		
TUD - N	- 10B down at 200kHz. 0.006% measured with		
THD + N			
	-10dBu input signal and		
	with a 20Hz/22kHz bandpass filter.		
Noise	•		
Noise	-90dBu (measured with a 20Hz/22kHz		
	•		
	bandpass filter.)		
High Pass Filter			
Roll off	12 dB per octave		
	2 pole filter		
Cut off frequency	-3 dB at 120 Hz		
Channel Insert and Compr	essor Sidechain Insert		
Connectors	Unbalanced input and		
	output on TRS		
	socket as follows:		
	Tip=Send (Output)		
	Ring=Return (Input)		

Inputs		Optical Compressor		
MIC		Threshold range	-12dBfs (10dBu) to	
Gain	OdB to 60dB continuously		-42 dBfs (-20 dBu)	
	variable	Compressor Ratio		
Frequency Response	OdB at 10Hz and -2dB	Hard ratio switch out	2.5:1	
	down at 200KHz.	Hard ratio switch in	6:1	
Mic EIN	-128dB (measured at 60dB	Attack time		
	of gain with 150 Ohm	Slow Attack switch out	0.5ms	
	terminating impedance	Slow Attack switch in	5ms	
	and 20Hz/22KHz bandpass	Release time	100 ms to 1 s and then	
	filter)	noiouou unio	auto release mode when	
THD + N	0.0004% (measured with a		the release knob is turned	
I TID T N	+16dBu input signal and		fully clockwise.	
		M-1		
	with a 20Hz/22KHz	Makeup gain	0 to +21dB	
	bandpass filter.)	Mid Scoop EQ		
LINE		EQ shape	Peak	
Gain	-10dB to +10dB	Centre Frequency	Variable between 120 Hz	
	Continuously variable	,	and 2 kHz	
Frequency Response	0.1dB down at 10Hz and	Cut and O		
	-3dB down at 200kHz.	Deep switch out	Cut=-6dB; Q=1.5	
THD + N	0.0006% (measured	Deep switch in	Cut=-12dB; Q=3	
	with OdBfs (+22dBu)	Deeh awiten in	Cut-12ub, u-3	
	input)	FX Send and Headphones Mix		
Noise	-94dBu (measured	Connector	Unbalanced output, signal level	
	with a 20Hz/22kHz		equivalent to -6dB below	
	bandpass filter.)		normal operating level	
NSTRUMENT	• /	Noise	-94dBu, measured with a	
Gain	OdB to 40dB		20Hz/22kHz bandpass filter.	
ouiii	Continuously variable 0.5dB down at 10Hz and	Maximum output signal		
requency Response		level	+20 dBu	
roquency ricoponico	-1dB down at 200kHz.			
THD + N	0.006% measured with	FX Return and Monitor In	lonitor Input	
ווי ד עחו	-10dBu input signal and	Connector	Balanced input (TRS socket)	
	• •	Operating level	+4 dBu	
	with a 20Hz/22kHz	Monitor Output		
	bandpass filter.	•	D-I (VID)	
Noise	-90dBu (measured	Connector	Balanced output (XLR)	
	with a 20Hz/22kHz	Operating level	+4 dBu	
	bandpass filter.)	Noise	-98 dBu, measured with a	
ligh Pass Filter			20Hz/22kHz bandpass filter	
Roll off	12 dB per octave		and monitor level knob at	
ion on	2 pole filter		maximum output level	
Cut off frequency	-3 dB at 120 Hz	DAC Performance		
		Playback sample frequency 32 kHz to 96 kHz.		
	Channel Insert and Compressor Sidechain Insert		Maximum bit depth 24 bit	
Channel Insert and Compre	Unbalanced input and	Maximum analogue outpu		
•	Olivalaliceu liiput allu	ıvıaxımının amaloyde outpl	ıı	
•	output on TRS	laval	+21 JD.,	
•	•	level	+21 dBu	
•	output on TRS socket as follows:	level Dynamic Range	111 dB measured with	
Channel Insert and Compre Connectors	output on TRS socket as follows: Tip=Send (Output)			
•	output on TRS socket as follows:		111 dB measured with	

Weight

5 Kg

