

NE 24.24M

NETWORK-ENABLED MATRIX PROCESSING WITH PROTEA[™] DSP

Whether you are designing or installing a system for corporate boardrooms, restaurants, courtrooms, houses of worship, left/center/right high output speaker systems in performance spaces, auditoriums or conference centers, our widely-popular **Protea™ ne24.24M Matrix Processor** will more than satisfy your requirements for any zoned system. When your install requires input/output matrixing with signal processing it doesn't get much easier than programming your channels using *Protea™* ne Software on your PC.

The ne24.24M uses modular expansion cards to provide up to 24-channels of audio matrixing and processing. The base unit offers a 4-input/4-output configuration. Each input and output expansion card has an individual DSP processor allowing you to expand the total input or output 4 channels of DSP processing at a time.

These cards are easily installed in the field without the need to reprogram the device.

Matrixing allows you to route any input to any output and control individual levels once they have been assigned. Fixed path architecture and extensive processing power per channel will reduce the amount of time it takes to set up your system. An optional GPO Logic Card allows the ne24.24M to trigger projection screens, curtains or lights. The logic card is installed in place of a 4-input or 4-output card and occupies one of the four expansion slots.

ne24.24M Features:

- 10/100 Ethernet & RS-232 computer interface standard
- Extensive DSP available
- Easy and intuitive user interface
- Mic/line inputs
- 24-bit A/D–D/A audio resolution
- Up to 24-channels of audio processing
- 4x4 base unit configuration
- Expand inputs or outputs 4-channels per module
- Modules easily field installable
- Euroblock connectors for audio, preset recall, DC remote level control and data in/out
- 31 preset locations
- Remote controls for level, preset recall and programmable functions
- Third-party control-friendly
- Input and output metering viewable in dBu
- Multi-level security
- Safety/Compliance: cTUVus, FCC, CE, RoHS



Specifications Note: OdBu = 0.775 VRMS Input Active Balanced, 18k Ohms Input Gain Range -50dB - +12dB, Selectable Polarity Output Active Servo Balanced, 112 Ohms Input/Output Level +20dBu (Max) **Output Gain Range** -50dB – +12dB, Selectable Polarity 20Hz-20kHz. ±0.25 dB **Frequency Response** THD <0.01% @ 1kHz, +20 dBu Dynamic Range >110dB (20Hz-20kHz) Unweighted Output Noise <-90 dBu Unweighted Environmental 40-120 deg. F, (4-49 deg, C) noncondensing Rear Panel Remote level control, Data In/Out ports, Controls Preset Recall, Logic Inputs, On/Off switch 10/100 Ethernet port. RS-232. Connections Euroblock In/Out Power Cord 3-Prong, Detachable Weight, Dimensions & Power 19"L x 3.5"H x 8.5"D (483mm x 89mm x 216mm) Dimensions Unit Weight 8.9lbs (4.04kg) Shipping Weight 12lbs (6kg) 90 - 240VAC, 50/60Hz, 40W Power Requirements

Internal Modules		
4-Channel, Input Module		
4-Channel, Output Module		
GPO Logic Output Option Module		
External Remotes		
2-Channel Level Control		
Preset Recall and Level Control		
Four-Position Preset Recall Switch		
Programmable Button Controller		
8-Channel Fader Remote		
Programmable Network Button Controller		
8-Channel Network Fader Remote		
16-Channel Network Fader Remote		
Remote Application for Apple® iPad®		





Protea™

DIGITAL SIGNAL PROCESSING FOR THE NE24.24M

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Protea is compatible with Microsoft® Windows 8, 7 (Vista/XP) 32 & 64 bit systems.

Audio professionals find our *Protea*[™] *DSP* to be very intuitive and easy to navigate—and you will too. No need to attend a one-week training class away from home to learn our software. Common sense layout of controls and features, on-line help, or a visit to the Technical Support page on our website provides answers to all of your questions.

Protēa™ DSP Specifications		
All DSP functions can be linked to 1 of 16 link groups		
Compressor		
Threshold	-20dBu to +20dBu	
Ratio	1.2:1-∞	
Attack	0.2 to 50ms	
Release	5ms/dB to 1000ms/dB	
Detector	Peak/Average	
Attenuation Bus	1 available	
Metering	In, Out, Attenuation, Graphical	
Autoleveler Controls		
Target Level	-40dBu to +20dBu	
Action	Gentle, normal, aggressive, user defined	
Maximum Gain	0dB to +27dB	
Metering	Attenuation	
Ratio	1.2:1 to 10:1	
Threshold Below Target	-30dB to 0dB	
Gain Increase/Decrease Rate	5ms/dB to 1000ms/dB	
Hold Time	0-6 sec	
Ducking: High/Low Priority, Trigger, Filibuster, Ducked Program		
Trigger Threshold	-80dBu to +20 dBu	
Ducking Release	5ms/dB to 1000ms/dB	
Ducking Depth	0dB to -30dB, -∞	
Enable Ducking at Matrix Mixer	Yes	
Metering	Input	

Gate		
Threshold	-80dBu to +20dBu	
Range	off, 100dB to 0dB	
Attack	0.2ms/dB to 50 ms/dB	
Release	5ms/dB to 1000ms/dB	
Metering	Gate LED, Graphical	
Gain		
Gain	-50dB to +12dB, off, polarity invert	
Remote Level Control	8 available, 0dB to -∞	
Remote RD8C Gain	Enable per channel, 0dB to -∞	
WR-5 (neWR-5) Remote Gain	0 to -50dB, Mute	
EQ: Parametric 15 Band		
Frequency	20-20kHz	
Level	-30dB to +15dB	
Q Value	0.016 to 3.995 Octave	
EQ: Hi/Low Shelf 6/12 dB/oct		
Frequency	20Hz–20kHz	
Level	-15dB to +15dB	
EQ: All Pass		
Frequency	20Hz–20kHz	
EQ: Variable Q HP/LP		
Frequency	20Hz–20kHz	
Q Value	3.047 to 0.267	
EQ: Notch/Bandpass		
Frequency	20Hz–20kHz	
Q Value	92.436 to 0.267	
Crossover: 2 Way, 3 Way, 4 Way Crossover & High Pass/Low Pass Filters		
Bessel & Butterworth Filters	12/18/24/48 dB/oct	
Linkwitz-Riley Filter	12/24/48 dB/oct	
Frequency	Off, 20Hz–20KHz	

Delay: @ 48kHz Sampling Rate (Input Time, Distance & Temperature)		
Speaker Delay	0–21ms	
Delay	0–682ms	
Delay: @ 96kHz Sampling Rate (Input Time, Distance & Temperature)		
Speaker Delay	0–10.6ms	
Delay	0–341ms	
Audio Metering Tool		
Range	-60dBu to +20dBu	
Increments	1dB	
Peak Hold Indicator	Yes	
Signal Generator Tool: Pink noise, White noise, Sine wave		
Signal Level	Off, -50dBu to +20dBu	
Sine Wave Frequency	20Hz–12kHz	
Matrix Mixer		
Gain (0.5dB increments)	Off, -50 to +12dB	
Mute	Per channel	
Enable Ducking at Mixer	Yes	
Ducking LED	Per channel if enabled	
Processors		
Input A/D, Output D/A	24-bit	
DSP Processors	24-bit signal, 48-bit filters, 56-bit accumulator	
Sample Rate	48kHz	
Propagation Delay @ 48kHz:	1.46ms	



NE 24.24M

ARCHITECT & ENGINEERING SPECS

ne24.24M

The digital signal processor base unit shall consist of four inputs and four outputs and shall use modular expansion cards to provide up to twenty-four channels of input / output audio matrixing and processing. Each expansion card shall have an individual DSP processor allowing for expansion of the base unit's total inputs or outputs four channels at a time. Expansion cards shall be factory installed or easily installed in the field without the need to reprogram. The processor shall use fixed path architecture to reduce set-up time. The processor control and programming shall be accomplished using a PC platform through a standard Ethernet connection. An RS-232 jack shall be provided for control and monitoring by a third-party controller. Multi-level security and no front panel controls shall insure tamper-resistant operation. Input channel processing blocks shall include a Mic/Line Preamp with 48V Phantom Power, Gain, Pink Noise Generator, Delay, fifteen EQ Filters, Gate, Autoleveler and Ducker. Output channel processing blocks shall include a Cross-Point Mixer, HPF/LPF, Delay, fifteen EQ Filters, Gain, and Limiter. The cross point mixer shall allow any input to be routed to any output at any level and mute any input at any output without affecting the true input configuration. Rear panel Euroblock connectors shall include eight preset recall contact closures plus eight remote potentiometer level controls. The DSP processor shall mount in a standard 19" rack using 2 spaces (3.5" high).

The digital signal processor shall be an Ashly DSP Matrix Mixer model ne24.24M



