## Programmable digital audio matrix processors

### DNM-88 · DNM-88D · DNM-1616 · DNM-1616D



### **FEATURES**

- DSP processing: 400 MIPS, 1,6 GFLOPS.
- Sampling rate: 48 kHz, ± 100 ppm.
- THD+N: <-94dB @4dBu.
- Input dynamic range: 110dB.
- Output dynamic range: 112dB.
- The processing modules are configurable and freely interchangeable as required.
- 8 units available for in-line operation simultaneous.
- Built-in USB sound card with one input and one output, supporting music playback, recording and soft video conferencing.
- For example: ZOOM, Teams, Tencent meeting, etc.
- Bus-based ACS, waiting time: 512 ms, convergence rate: 60 dB/S, echo cancellation amplitude: 60 dB.
- AFC (feedback suppression) for independent channels with trap-type algorithm, amplitude of sound transmission gain improvement: 10 dB.

- Noise suppression (ANS), improved ratio S/N of 18 dB.
- · Priorities (Ducker).
- Noise gain compensator (ANC).
- Automatic gain sharing mixer (AMC), Automatic gate mixer (Gate Mixer).
- 12-band parametric EQ with 5 filter options.
- Parametric, Lowshelf, Highshelf, Lowpass, Highpass.
- Customizable interface for the end user, up to 30 units with the same interface management; support for camera tracking.
- With 8 GPIO channels.
- With centralized control function, support for : RS232, RS485, UDP three kinds of control mode.







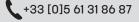
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### **TECHNICAL CHARACTERISTICS**

Model	DNM-88	DNM-88D	DNM-1616	DNM-1616D		
Processor	ADI SHARC 21489 @450 MHz SIMD	ADI SHARC 21489 @450 MHz SIMD x2	DI SHARC 21489 @450 MHz SIMD	ADI SHARC 21489 @450 MHz SIMD x2		
Sampling rate/quantification bits	48K/24bit					
Number of analog input and output channels	8x8	8x8	16x16	16x16		
Number of Dante input and output channels		8x8		16x16		
Input gain	0/3/6/9/12/15/18/21/24/27/30 33/36/39/42/45/48 dBu					
Phantom power supply	+48V/10mA max					
Frequency response (20 ~ 20kHz)	±0.3dB					
Maximum level	+18dBu					
THD+N	<-94dB @4dBu					
Input dynamic range	110dB					
Output dynamic range	112dB					
Channel isolation at 1kHz	108dB					
Input impedance (balanced connection)	5.4ΚΩ					
Output impedance (balanced connection)	600Ω					
System delay	<3ms					
Power supply	AC110~240V, 50Hz/60Hz					
Dimensions (W x D x H)	482x260x45mm					
Weight with packaging	4 KG					
COMPATIBLE DEVICES AND ACCESSORIES						
LM-W	Programmable IP/PoE wall control panel with custom editing control function, supporting up to 32 control functions:  Gain control  Mute  Presets  Matrixing  This wall control panel is ideally positioned and assigned to a zone in order to control the volume and the choice of sources remotely  Built-in or surface-mounted					
RM-8	The RM-8 is a 8 zones re Line level balanced au Power supply +12VDC RS-485 communicatio	udio output .				
W-TAB10	W-TAB10 is a TFT touch screen compatible with all DNM products. It includes a 10" capacitive touch screen (user control panel). Each unit can be configured to operate as a control panel for one or more zones. It allows to control the volume, selection of the sound source, launch Presets, communication via TCP/UDP with external devices, etc.					







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SOFTWARE						
MAJORCOM DNM	The DNM software is a complete Windows-based application that configures and controls DNM matrices. Systems can be designed to perfectly match specifications and comply with changing or evolving needs and requirements. All classes of audio processing modules for signal routing and shaping are provided.  DNM software offers a revolutionary way of thinking about DSP software. This software is designed by audio professionals for audio professionals. It uses an interface that will be very intuitive, even for the most novice integrators, while being extremely powerful in its capabilities.  Intuitive and user-friendly interface  Automatic discovery of devices  Parametric/graphic EQ, dynamics processors and automatic gain control  Acoustic echo suppression  USB sound card for playback and recording  Signal generator per channel  And much more					
Requirements for the installation of the software	A Windows PC with:  An Intel i5 or higher processor  8 GB of memory or more  1 GB of free storage space  Windows 7 or higher  Minimum resolution of 1920 x 1080  Network port (Ethernet)					
CUSTOMIZABLE GUI CONTROL (Graphical User Interface)	DNMs support a customizable graphical interface that can be built to facilitate remote control of general elements such as volume, mute, presets, matrixing, and other external control commands. Supports Android and Windows control options with or without wires.					

### **FRONT FACE**



DNM-88

USB audio for playback and recording (see manual)

Status light

Screen

Power indicator

Non-contractual design. Design may differ without notice.







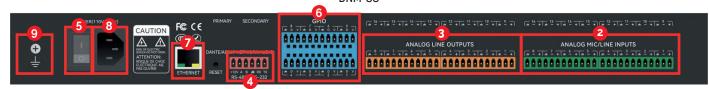


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#### DNM-88D







- Dante Primary and secondary (according to model)
- Analog audio inputs (number depending on model)
- Analog audio outputs (number depending on model)
- Serial communication: RS-232 and RS-485
- On/Off switch
- **GPIO** connector
- Ethernet connector
- 230VAC power supply
- Grounding lug





