# Power Amplifiers MDA Series - MDA4-7000DM 

MDA 4-7000DM is a high-performance and highpower touring-grade digital power amplifier. It has a total power output of $4 \times 1300 \mathrm{~W}$ under a 4 channel $8 \Omega$ load. Built-in powerful digital processor with 4 inputs and 4 outputs. The 4 .3inch high-definition touch screen clearly displays the system status, and the sufficient power is equipped with a powerful built-in DSP to enable this power amplifier to take into account both the fixed installation and the mobile performance market.

MDA4-7000DM adopts powerful Marani DSP, including 4 input and 4 output processing channels. Both DSP and AD/DA run at 96 KHz sampling rate. The complete processing function
provides a complete crossover solution for the speaker. With signal hot backup automatic switching function, 3 levels of priority can easily back up the input source.
From input gain/delay/noise gate/EQ/compression /FIR /to output gain/delay/polarity/Xover/FIR/EQ/RMS compressor/Peak limiter, there are up to 13 PEQ types to choose, The output crossover filter includes the classic Linquez Rayleigh/Bessel/Butterworth, and the MARANI brand unique NXF (Northed X-over Filter), builtin FIR filter, and newly added MIR linear phase filter can make the phase of the crossover point easier to join while maintaining a very low delay. All the functions we provide are designed to help you better restore the sound.


## Features

1. $4 * 1300 \mathrm{~W}(<1 \%) @ 8 \Omega$ high power, independent power supply for each channel, 4 channels can output maximum power at the same time regardless of whether they are connected to fullrange speakers or sub-low speakers.
2. The built-in MARANI DSP runs at $96 k$ sampling rate, and the high frequency response can reach 40 KHz . The brand-new circuit design makes the MDA4-7000DM has the advantages of soft sound, powerful self-protection function, low noise and high efficiency.
3. Source priority automatic switching function, the machine is equipped with 4 analog input interfaces, 2 independent AES digital input interfaces, and 2 Dante network interfaces. Each input channel can be set to 3 priority levels to effectively ensure the reliability of system signal transmission during major events.
4. Added a new "zero delay" hard limiter to better protect the speaker unit.
5. The fourth generation iFIR wizard V4.0 plug-in supports automatic measurement and generation of FIR coefficients, as well as the import of FIR coefficients generated by thirdparty software. A new MIR linear phase filter is added to the output crossover filter, which can
avoid the phase distortion caused by the traditional IIR filter.
6. Built-in dynamic loudness filter, the working principle is to self-adaptively boost the ultra-low and ultra-high frequency bands according to the equal loudness curve of the human ear, and the boost ratio is determined by the magnitude of the signal amplitude, which significantly improves the overall hearing of the small and medium-sized speaker system.
7. The control network port is independent of the dual Dante network port, and the control network system and the audio network system are separated design, which makes the system structure simpler and clearer, and the error rate of the network part is lower. Even if the control network fails, it will not affect the audio network work. With simple and intuitive management software, the success rate of one-time connection is extremely high.
8. The system gain of the power amplifier can be switched by software, 2 gears are optional: $32 \mathrm{~dB} / 26 \mathrm{~dB}$. The maximum input level can be switched by software, which are +15 dBu and +21 dBu , which is convenient for different usage scenarios and easily matches different front and rear gain architectures.

## General

| Dimensions-- | $482 \times 88 \times 470(\mathrm{~mm}) 2 \mathrm{LU}$ |
| :---: | :---: |
| Weight, Net / Shipping | $14.5 \mathrm{Kg} / 16 \mathrm{Kg}$ |
| Preset number-- | 50 |

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## DSP \& Processing

| Si | white noise/pink noise, level range: $-40 \mathrm{dBu} \sim \mathrm{OdBu}$ |
| :---: | :---: |
| Input | -18 dB ~ + $12 \mathrm{~dB}, 0.1 \mathrm{~dB}$ step |
| Noise gat | Threshold range: -85dBu~-50dBu Start-up time: $1 \mathrm{~ms} \sim 1000 \mathrm{~ms}$; Release time: $1 \mathrm{~ms} \sim 1000 \mathrm{~ms}$ |
|  | Gain range: 0 dB -10dB |
| Parametric Equaliz | Each input channel can have up to 12 optional types of PEQ, and each output channel have up to 8 optional types of PEQ |
| Type | Bell, 1 st order/2th order high shelf, variable $Q$ high Shelf filter, 1 st order/2th order low Shelf filter, Variable Q low Shelf filter, 1 st order/2nd order low pass filter, Variable Q low pass filter, 1 st order/2 order high pass filter, Variable $Q$ high-pass filter, band-pass filter, notch filter, 1st-order all-pass filter, 2nd-order all-pass filter with variable Q <br> The center frequency is adjustable within the frequency range of $20 \mathrm{~Hz} \sim 20 \mathrm{kHz}$ with a step accuracy of 1 Hz |
|  | The $Q$ value range of the Bell filter is: $0.4 \sim 128$, with a step of 0.01 , The $Q$ value range of the F/high-pass/low-pass filter is: $0.1 \sim 5.1$, the step is 0.01 and the $Q$ value range of the band-pass/notch filter is: $4 \sim 104$, the step is 1 . |
| Equalizer gain ra | $-15 \mathrm{~dB} \sim+15 \mathrm{~dB}$ |
| IIR crossover filt | Butterworth slope: 6/12/18/24/36/48dB per octave, bay Searle slope: $12 / 24 \mathrm{~dB}$ per octave, Linquez•Rayleigh slope: $12 / 24 / 36 / 48 \mathrm{~dB}$ per octave, NXF horn filter slope: 40/45/50/50/ 55/60/65/70/75dB per octave |
| MIR linear phase f | Butterworth slope: 6/12/18/24/36/48dB per octave, Bessel Slope: $12 / 24 \mathrm{~dB}$ per octave, Linquez•Rayleigh slope: $12 / 24 / 36 / 48 \mathrm{~dB}$ per octave, NXF horn filter slope is 40/45/50/50/55/60/65/70/75dB per octave |
| FIR Crossover filter | filter type; high-pass/low-pass/band-pass/external import, Taps range: 256 <br> 512, slope Range 21~120dB per octave, time window type: Rect / Sinc / Keiser / Hanning / Hamming / Blackman /Blackman-Harris/ Blackman-Nuttal / Nuttal/ Keiser-Bessel/Sine. |
| RMS compressor | Threshold range: $-15 \mathrm{dBu} \sim+12 \mathrm{dBu}$; compression ratio range: 2~32: 1; soft and hard Inflection point: $0 \sim 100 \%$ Start-up time: $0.1 \mathrm{~ms} \sim 1000 \mathrm{~ms}$; Release time: $100 \mathrm{~ms} \sim 15000 \mathrm{~ms}$ Gain compensation: $-12 \mathrm{~dB} \sim+12 \mathrm{~dB}$ |
| Peak limite | Threshold range: $-15 \mathrm{dBu} \sim+12 \mathrm{dBu}$ Start-up time: $1 \mathrm{~ms} \sim 1000 \mathrm{~ms}$; Release time: $100 \mathrm{~ms} \sim 5000 \mathrm{~ms}$ |
|  | The adjustable delay time of each input channel + output channel is 452 ms , and the step accuracy is 10.4 us; |
|  | Each input channel and output channel can choose to import a FIR filter with 512 taps. |

