### **PUMA XD1616**

# **Digital Signal Processor**



#### **Product Introduction**

The digital signal processor is a freely designable audio processing and control system product. The front panel comes with a high-definition color screen, which can display the current working status of the equipment. using advanced DSP processing technology, the system provides a variety of audio processing modules to choose from, open software architecture, targeted to solve a variety of practical problems in the application scenarios. 128 network audio channels for network audio transmission provides a high-bandwidth, low-latency, high compatibility and low-cost solutions.

#### **Product Features**

- ➤ Highly integrated, integrating a variety of traditional Analog audio processing equipment in a Digital audio processor;
- ➤ High-performance 32-bit floating-point DSP processor, all-digital processing, fast response to feedback cancellation, echo cancellation, noise suppression and other core algorithms;
- → High-performance A/D, D/A converter, 24bit/48KHz sampling frequency, high-quality Analog
   → Digital, Digital → Analog conversion;
- ➤ 16 Analog input channels and 16 Analog output channels, very small distortion and ultra-low background noise;
- Rich interface expansion;
- > Supports Dante network transmission, which makes audio transmission more stable and faster;
- Humanization, graphical, intuitive and easy-to-operate control software interface;



> Comprehensive matrix mixing functions.

#### **Functions**

- → Advanced and Reliable Hardware: Analog Devices High Performance DSP Sharc21569 + NXP
  i.MX ARM Processor;
- ♦ Support 64\*64 Dante network audio channels (AES67 compatible)
- ♦ Open signal flow design, do what you want to do;
- ♦ Visualized user control interface design, easily meet customer needs
- ♦ Support built-in multi-channel recording & broadcasting;
- ♦ Support any signal real-time monitoring, debugging is no longer blind;
- ♦ Display IP address & device name, no more worries about not finding the device;
- ♦ Built-in all the required software, no more lost CD-ROM trouble;
- ♦ Highly optimized DSP, worry-free processing up to 3400 equalization bands
- ♦ 4 independent feedback elimination, 2 independent echo cancellation;
- ♦ Open Websocket protocol control, easy cloud platform integration, support for RS232, UDP center control;
- ♦ Independent dual control network port, easy network backup;
- ♦ Dual Gigabit Ethernet ports, network audio transmission can be redundant backup;
- ❖ Built-in graphic equalization, parametric equalization, high and low-pass filtering, all-pass filtering, FIR filtering, crossover, noise gate, limiter, multifunctional mixer, automatic mixing, adaptive echo cancellation, a variety of feedback cancellation algorithms (traps, adaptive filtering), etc.;
- ♦ Built-in USB sound card, support for recording and playback functions, playback and recording path selectable, recording storage space real-time display;
- ♦ Support audio dynamic level real-time feedback;
- ♦ Support real-time analysis of the signal spectrum of each channel.



## **Specification**

16 balanced/line inputs 16 balanced/line outputs
64 input,64 output
1
8, freely configurable inputs and outputs
4
1 USB Type A,1 USB Type B
48KHz@24bit
60dB(3dB per stop, 20 stops in total)
48V
20Hz∼20KHz,±0.2dB
≤0.003%@1kHz,4dBu
120dB
114dB
20kΩ
100Ω
≤-125dBu
100dB@1KHz,4dBu
70dB@50Hz
108dB



Maximum Input Level	18dBu
Maximum Output Level	18dBu
Background Noise	-90dBu
Operating Power Supply	AC 110V-220V,50Hz/60Hz
Power Consumption	<40W
Operating Temperature	0-40℃
Operating Humidity	10%~90% No condensation
Product Weight	2.2KG
Product Size	482x258x45mm
Package Weight	2.7KG
Package Size	590×430×110mm