



Feature:

- 1.This product has the function of fully automatic detection of howling points, adopts DSP high-speed floating-point computing technology, and uses adaptive feedback control algorithm, notch filter algorithm, bass compensation algorithm, automatic mixing algorithm, etc. to process high-speed feedback of sound, restore the fidelity of sound to the greatest extent, realize fully automatic feedback elimination and sound field correction, real-time response, one-button operation, fully automatic operation mode, and effectively improve the microphone's pickup distance
- 2.Adopting adaptive professional algorithm, intelligent high-speed feedback processing, can greatly suppress howling, and can effectively prevent burning of audio equipment and speakers
- 3.One-button control of feedback suppression switch function and one-button pink noise test and calibration function, on-site debugging is extremely simple Single and convenient
- 4.One-button intelligent manual debugging, no matter the position, temperature, humidity, decoration changes in the room environment, the system installation does not need to debug the sound field
- 5.Compatible with hand-in-hand conference microphones and analog conference microphones, effectively improve the microphone gain 6-12dB, speaking distance 30-80CM
- 6.The front panel color screen displays the device status in real time: input signal power, output level, system voltage, device stability, etc
- 7.Six MIC balanced (unbalanced) signal inputs and two balanced signal outputs, each balanced input with independent gain adjustment and 48V phantom power supply
- 8.The sixth microphone supports separate feedback and bypass working modes to facilitate system signal management
- 9.Supports independent gain adjustment function for each input And 48V phantom power supply with independent switch
- 10.Two sets of RCA auxiliary input and two sets of RCA auxiliary output, compatible with more devices
- 11.Six-way MIC signal input and line RCA signal input high and low frequency adjustable

Specification:

Feedback suppression mode frequency response	120hz-15khz
Pass-through mode frequency response	20HZ-20KHZ
Sampling rate	32Khz
Transient response distortion	<0.1% (1khz)
Signal-to-noise ratio	>90db
System signal delay	<11ms
Frequency shift value	-5HZ
Balanced input impedance	68KΩ
Balanced input sensitivity	-30db-56db
Line input impedance	10KΩ
Music input impedance	10KΩ
Balanced output impedance	1KΩ
Line output impedance	2KΩ
Recording output impedance	2KΩ
Power input mode	AC220 ~ 50Hz