



Feature:

1. Network-based all-digital audio transmission technology
2. All-digital audio transmission technology is used between conference units and between conference units and the host, with 24bit/48KHz sound quality standard and uncompressed transmission based on the network
3. Built-in high-performance DSP audio chip, which can independently adjust the microphone gain, optimize the audio signal, restore the human voice, and have excellent sound quality
4. Support dynamic allocation of unit ID, high system flexibility
5. Diverse software modules can realize multiple working modes, custom chairman/VIP, channel language selection, audio input/output, unit/host equalizer, recording, camera tracking and other functions, which is very forward-looking and can easily expand new functions in the future
6. 3.5mm stereo headphone jack can connect headphones
7. The microphone can be adjusted from -50° to 45° (horizontal)
8. The system has an automatic repair function and supports hot plugging of lines
9. Made of zinc/aluminum alloy and anodized, it is mainly used in professional occasions such as high-end conferences and lecture halls

Specification:

Frequency response	20Hz-20KHz
Maximum power consumption	1.5W
Headphone load	>16Ω
Headphone volume	10mW
Headphone output interface	Ø3.5mm stereo jack
Connection method	dedicated 6-core cable
Connector	6P-DIN+clip
Color	Base: brown Microphone rod: black
Weight	0.95kg
Dimensions (length * width * height mm)	Base: 138*98*54 Microphone rod: 200*34.5*23

Microphone parameters	
Microphone capsule	14mm diameter gold-plated capacitor capsule
Directional characteristics	supercardioid
Sensitivity	-28dB
Frequency response	20Hz-20KHz
Input impedance	2.2KΩ
Maximum sound pressure level	130dB (THD<3%)
Equivalent noise	25dB-A