



OpenVox VS-GWM501V Audio Board

Summarize

VS-GWM501V audio board is a powerful voice access equipment, convenient teleconferencing and conference room audio mixer system docking, teleconferencing reception, amplification through the telephone line transmission, to meet the needs of different industries of video teleconferencing applications, to ensure clear voice quality, powerful telephone access control capabilities. Vs-GWM501V audio board not only supports pstn access but also supports sip line registration, is the first choice for high reliability needs of conference room applications. VS-GWM501V audio board not only supports PSTN access but also supports SIP line registration, which is the first choice for meeting room applications with high reliability requirements.

Features

System Characteristics

- Strong compatibility
- Echo suppression, noise suppression
- Simultaneous output of two audio channels
- Unrivaled noise cancellation technology
- Moderate price/performance ratio, simple integration of applications
- 1U Standardised design saves deployment space
- Patented voice algorithm for clear voice quality
- Simple and convenient configuration via WEB GUI
- Multiple telephone (SIP, PSTN, GSM) access processing capability

Powerful compatibility

The equipment supports the use of different telephone lines, and can be accessed to PBX subscriber lines, telecommunication bureau subscriber lines, analogue subscriber gateway IAD and other equipment. In addition, it can also support IP-PBX, softswitch, SIP server SIP applications of major companies, and can be seamlessly connected to the majority of scheduling command system, conference video, central control system.

Venue Audio Gateway Applications

- large conference rooms
 Court room
- Movement Control Centre
- Forces, public security
- Government departmental conference rooms
- Field command conference room
- building broadcast
- Training room
- Command vehicle conference room

	VS-GWM501V Audi o Board
Port Type	AV Lotus Head
port protocol	Standard SIP、TCP/IP
sizes	160mm*100mm
weights	0.132kg
Panel Color	gray
Capacity	2 pairs Audio inputs and outputs
Voice Coding	G.711a、G.711U
Distortion	≤1%
Signal-to-noise Ratio	≥70dB (excluding telephone line noise)
Side Tone Suppression	≥45dB
Maximum Power	2W
Power Supply	12VDC
Operating Temperature Range	-20℃ ~60℃
Operating Humidity Range	8% ~ 90% non-condensing
Minimum Ventilation Space	6.4cm