

SIP-T300

Audio Gateway T300 is a SIP broadcast module carefully developed for the specific needs of industrial broadcast users, which uses the international standard IP/ RTP/ RTSP protocol for media streaming. The module integrates a variety of functional interfaces, covering broadcast, real-time acquisition, two-way intercom and other functions. With its excellent adaptability, the T300 audio gateway can meet the needs of a variety of complex use environments, and users can quickly deploy and configure devices. Its compact size makes the T300 audio gateway ideal for customized applications in a variety of integration projects, providing a flexible and efficient solution for broadcast communication systems.



Advantages of equipment



Small and exquisite, can be embedded in other equipment installation



Supports single-key direct call to the dispatching console



Linein Audio input interface
3.5mm standard audio interface, plug and play



Line out -Audio output interface
Extensible external active speaker interface



10W~30w mono power amplifier output
Adapt the output power according to the input voltage



100 Megabit Ethernet - Adaptive 10/100 Mbps network port
Provides fast network transmission and supports PoE power supply

Basic function

- 2 SIP lines
- Speed dial
- IP address for voice broadcast
- MP3 radio
- Supports custom DSS key interfaces
- Support mono active speaker
- Supports audio line input
- Power over Ethernet

Device function

- Default Auto answer (Auto answer)
- User-defined DSS key
- Key or remote volume control
- Network time synchronization
- Dynamic multicast function
- Action URL/ Active URL

Phone function

- Dial/ answer
- Call waiting
- Intercom function, multicast
- Speed dial
- Anonymous call

Configuration and management

- Auto-Provisioning through FTP/ TFTP/ HTTP/ HTTPS/ DHCP
- OPT66/ SIP PnP/ TR-069
- Web management interface
- Export/ import the configuration of Web-based Packet-dump
- upgrade the software
- Configuration import/export
- Syslog

Audio characteristic

- Narrow band coding: G.711A/U, G.729A/B, G.726-32K
- Wide band coding: G.722
- Duplex intercom, support AEC
- DTMF: In-band, out of band (RFC2833/SIP INFO)

Network characteristics

- 10/100 Mbps Ethernet
- IP mode: IPv4/IPv6/IPv4&IPv6
- IP Acquisition mode: static IP setting/ DHCP
- VPN: L2TP/OpenVPN
- VLAN

Related agreement

- SIP V1 (RFC2543), V2 (RFC3261) over UDP/TCP/TLS
- ORTP/RTCP/SRTP
- STUN
- DHCP
- IPV6
- L2TP
- OpenVPN/SNTP/FTP/TFTP/HTTP/ HTTPS
- TR-069

Temperature&humidity

- Working temps: -20°C~50°C
- Storage temp: -30°C~60°C
- Working humidity: 10~95%

Power

- DC input: DC12V~24V/2A
- PoE level: Class 4
- PoE power consumption: 1.85W~10.8W
- Adapter power consumption: 1.1W~8.5W

Gateway keys

- 3 physical keys including 2 sound controls and 1 reset key (soft reset)

Gateway interface

- DC interface *1: Connects to the power
- RJ45 interface *1: Connects to the Internet or LAN, support PoE power over Ethernet
- Indicator light*1: SIP status and net status
- Headphone jack*1: 3.5mm standard audio interface
- Audio output interface*1: Connect the active speaker.
- DSS key interface*1: Connection speed dial key (support with light)
- 4Pin Access control interface *2: Connect access control switches and support the control lights
- Microphone input interface *1: Connect the MIC
 - It is recommended to use electret condenser microphone.
 - Sensitivity: -38dB
 - impedance 2.2K0hm,
 - bias voltage 2.54vpp
 - Interface: Plug-in terminal blocks
- Passive speaker interface *1: Connect speakers.
 - Output power :42, PoE/10W, 12V/10W, 18V/20W, 24V/30W depends on the power .
 - Wire diameter: 18AWG or larger
 - Interface: Plug-in terminal blocks
- Audio files are broadcast offline

Physical specification

- Quantity/box: 20PCS
- Net weight/box: 6.6kg
- Gross weight/box: 7.12kg
- Installation mode: desktop/wall mounted
- Color: dark gray
- Product size: 156mm x 117mm x 33mm