

Conference Controller

Full digital conference system audio transmission embedded software

TS-0300M embedded V4.41

TS-0300M /custom/



Feature:

- Uncompressed audio transmission with a sampling rate of 48K. Category 5e shielded cables are used to ensure reliable transmission of conference information over long distances while providing perfect sound quality. The device uses segmented compression mixing processing technology and clock synchronization transmission technology, and the delay from conference unit pickup to host output is 5ms. Built-in DSP processor, with 16-channel audio matrix, howling suppression, 10-band EQ adjustment, volume dB value adjustment, and delay adjustment functions.

- 4. Supports two recording methods: host USB flash drive and client software; with conference microphones and recording boxes, it can record the speech audio of a single conference unit and the mixed speech audio of all conference units

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 5. The device has 1 USB interface; the rear panel has 2 RS-232 interfaces, 1 RS-485 interface, and 4 RJ45 communication interfaces; it has 1 RCA input, 1 XLR input, and 2 Phoenix terminal input interfaces; 1 RCA output, 1 XLR output, and 16

 Phoenix terminal output interfaces; 1 DIP switch, 1 grounding column.

 6. It has 16 audio output channels, which can be expanded to 272 audio output channels. The audio output channels can be configured as wired role separation output mode, wireless role separation output mode, and simultaneous transmission output mode; each audio output channel can independently adjust audio parameters, including 30-level volume adjustment, 10-band equalizer adjustment, and 100-level delay adjustment functions.

 7. It has 3 backup mechanisms; it supports the host dual-machine hot standby function, which can set one device as the host and the other as the slave. When the host falls, it can automatically switch to the slave to realize the dual backup function; it supports the ring dual-link function to ensure that the meeting can continue normally when one of the network cables is disconnected or the unit has problems; it supports the T-link backup function. Even if multiple conference units fail in the link, other
- conference units are not affected, ensuring the normal progress of the meeting.

 8. Supports the use with conference microphone processor. The host and microphone processor transmit audio through the network cable connection. It can simultaneously transmit the audio signals of 16 wired conference units and 8 wireless ference units, and provide feedback suppression, intelligent mixing and automatic gain audio adjustment processing functions.

 16-channel role separation output mode allows wired or wireless units to output independently according to ID numbers, supporting up to 128 channels

 Independent audio output of wireless units or wireless units in a supports independent audio output of wireless units or wireless u

- expanded through external devices The host has a 16-channel audio group output interface; using conference partition phase control technology, it can be split into 16 independent conference systems for use, or it can be used as a large conference system, realizing multiple
- 12. In ho host thas a 16-channel audio group output interface; using conference partition phase control technology, it can be split into 16 independent conference systems for use, or it can be used as a large conference system, realizing multiple ways of conference room merging/splitting.

 13. It has C/S and B/S control architecture, including client, WEB, local full-color touch screen, and Android phone/tablet control mode; through the client and WEB, you can adjust the audio matrix parameters (including EQ, volume, delay, conference unit sensitivity, etc.), 16-channel output mode switching, switch conference unit, switch between Chinese, English, Russian and French, and control role separation host function; use the local full-color touch screen to adjust the conference mode, wired/wireless conference unit microphone number, ID, host/slave setting, switch between Chinese, English, Russian and French, display brightness/output volume adjustment, display remaining days, enter registration code for host registration function; use Android phone/tablet to control the conference unit switch, start sign-in, vote, receive conference service information, and one-click close the wireless conference unit function, no PC operation required.

 14. It has C/S and B/S architecture management software, and 8 operating systems that can run on both client and WEB software, including Windows7/10/11, Kylin Desktop Operating System (Zhaoxin Edition), Kylin Desktop Operating System (Zhaoxin E
- (Feitian Edition), macOS System, Tongxin UOS, and Ubuntu Desktop Operating System.
- 15. The device has the function of being controlled by Android mobile phone APP software and tablet APP software. The software can control the microphone switch, sign in, vote, receive conference service information, and turn off the wireless microphone with one click, etc., without PC operation.
- It supports 4096 wired conference units and 300 wireless conference units and 300 wireless conference units to access and manage at the same time; it supports 4396 conference units to participate in the meeting agenda (sign in, vote, service, etc.) and speech control at the
- It supports with effection of customizing the identity of conference units speakers, and the number of wired conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of conference units speaking at the same time; it supports on the units of the un

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- Support AB channel scanning, monitor the use of wireless channels on site, support automatic or manual configuration of the best channel, and support automatic or manual configuration of the best channel, and support automatic or manual configuration of the best channel, and support automatic or manual configuration. The conference host has the function of displaying the number of days registered, so you can know the remaining days after registration at any time; support entering the registration code on the touch device screen for host registration. The system is deeply adapted to the voice transcription systems. The system is elegable and the same time, support entering the registration of under the registration of the destruction of the system in the same time, and the two can be used in parallel; cross-domain audio synchronization technology is adopted, and the audio of wired and wireless conference units at the same time, and the two can be used in parallel; cross-domain audio synchronization technology is adopted, and the audio of wired and wireless conference units is the mixed and output. 33. 34
- The frost panel has 5 status indicators, which can display the use status of wired and wireless conference units are normal communication; 1 of them is the communication indicator of wireless conference speaking units, which flash when the wireless can be quickly
- The WEB management terminal has the function of switching personalized theme styles, which can switch 4 styles, including simple theme, government theme, fashion theme, and vitality theme. Different themes provide different UI interface background colors
- It has the function of hot standby of host dual machines, which can set the host or slave function. When the host fails, it can automatically switch to the slave operation to realize the dual backup function
- 17. It has the function of hot standby of host dual machines, which can set the host or slave function. When the host stalls, it can automatically switch to the slave operation to realize the dual backup function.

 18. It has the function of operation and maintenance management platform, which can remote upgrade the firmware through the web end; it has log management function, which can automatically collect and store system logs; for example, real-time monitoring of equipment operation status and equipment failure information, including insufficient memory, fire alarm prompts, ID duplication, etc.

 18. The conference host software is integrated into the audio integrated management platform to achieve unified management of audio equipment. The platform can scan the online status of digital conference hosts, audio processors, mixers, suppressors, and amplifiers. Multiple online devices of the same product can also be scanned, and the device hardware name, hardware IP address, online and offline status information will be displayed; it has the function of uploading configuration information to the cloud or saving it locally for backup with one click, and restoring configuration information with one click.

Specification:

Model	1 S-0300M
Microphone capacity	Wired microphone ≤4096; wireless microphone ≤300
Simultaneous interpretation channel	63+1 channel
Frequency response	80~16KHz
SNR	>78dB(A)
Dynamic range	>80dB
THD	<0.05%
Main power	100-120VAC/200-240VACbyswitch
Audio input	LINEIN1: 775mVrms balanced; 2 input Phoenix terminal: 775mVrms balanced; LINEIN2: 775mVrms unbalanced
Audio output	LINEOUT1:1Vrms balanced; 16 multi-function output Phoenix terminal: 1Vrms balanced; LINEOUT2:1Vrms unbalanced
Output load	>1ΚΩ
EXTENSION port	1 for connect conference system extension equipment
DANTE/NC port	1 for connect to external devices with DANTE protocol
WIFI network port	1 for connect to wireless AP
PC network port	1 for connect to the computer
DELEGATES output interface	4 for connect conference speaking units
RS-232 interface	2 channels, 1 channel for camera tracking, 1 channel for docking external equipment
RS-485 interface	1 for camera tracking
Static power	30W
Output power consumption	320W
Wired microphone connection method	Special cable (6 cores)
Touch screen control	4.3 inch full color touch screen
Colour	black
Net weight	5.6Kg
Dimension (LxWxH)	484x03x88mm
Installation method	19 inch standard cabinet
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