Main Controller

Model: VCS-HR6330M Plus





SUMMARY

• This is a new intelligent full-function digital conference system, which perfectly introduces all digital technology and integrated network technology into the conference system, and fully combines advanced digital technology, network technology and audio technology. It can be seamlessly connected with intelligent central control system, video conference system and fire monitoring system, providing a more complete solution for efficient modern conference system engineering.

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FEATURES

- Adopting DSP/DOV digital processing and transmission technology, equipped with high-performance ARM Cortex A9 industrial grade embedded processor, 1.4GHz main frequency, 1G memory, faster running speed, supporting discussion, video tracking, sign in and voting functions
- Built in AC100V~240V international universal power supply and voltage stabilization system, ensuring safe and stable use of voltage in unstable situations
- All-metal structure, with anti-interference circuit design, effectively eliminates interference from electromagnetic signals
- The system supports conference room merging/splitting functions, suitable for various conference application venues;
- Adopts high-fidelity lossless audio transmission technology, 48KHz audio sampling frequency, and frequency response of 20Hz-20KHz;
- Adopts 4.3-inch full-view IPS capacitive touch screen, and the graphical human-machine interface is easy to operate, which improves the debugging efficiency. and supports simplified Chinese and English display menus, the menu language can be customized according to needs;
- Built in USB automatic recording module, it can record all audio content in the meeting and support 2000 hours of continuous recording;
- Adopts ID addressing mode, and can customize the unit number to ensure a unique ID number, which can effectively avoid ID duplication conflicts;
- Support factory-reset function, supports one-click reset to the factory state, and prevents parameter adjustments;
- With 4 groups of 8-channel RJ45 microphone unit interfaces, adopts closed-loop Ethernet link mode, supports bilateral power supply of microphone units, and supports hot-plugging of microphone units. Each channel supports 20 units, which can be connected to 80 units.
- With 4 channels of 8-core microphone unit interface, each channel supports 25 units, can be connected to 100 units
- The Single controller can be connected to a maximum of 256 conference units through the expansion power supply, or a conference expansion controller (multiple extension controllers are connected in series in hand), and a conference system can access up to 2000 conference units;
- Support the dual-machine hot backup function. Through the backup transfer controller, one conference host can be set as a backup controller. When the conference controller fails, the backup controller automatically takes over;
- Support the function of "hot plugging" when the line is live, supports automatic repair and quick positioning of system faults, and does not affect the operation of other equipment;
- With a variety of conference modes: FIFO (first in, first out mode), APPLY (application mode), FREE (free mode), LIMIT (restricted mode), VOICE (voice control mode)
- Speaker limit function: can limit the maximum number of units that can speak at the same time (can be set to 1-8 different numbers), the chairman unit is not limited
- Speaking time limit function: can limit the speaking time of the speaking unit (can be set to 0-1000S), there are regular shutdown and automatic shutdown, the chairman unit is not limited

- Built-in multi-channel internal communication and conference service functions, the conference host can receive conference service request information sent by the conference unit (such as tea, coffee, paper, pen, help and other services);
- Built-in electronic voting function, with data management functions such as sign-in, voting, election, scoring, etc.;
- Supports functions such as keystrokes and supplementary sign-in, and the sign-in countdown time and sign-in data can be displayed in real time;
- Supports named and anonymous voting methods, all votes can be exported and archived and support graphic printing;
- With extended simultaneous interpretation function, it can be configured with 4/8/12/16/20/32/64 different languages;
- Use high-performance DSP processor to perform 5-band equalization processing on audio, make the sound clear and thorough through settings, effectively suppress howling, eliminate interference, distortion, crosstalk and other phenomena to the greatest extent, and preset 3 default modes (full, human voice, dynamic) and 1 custom mode for users to use flexibly
- Network-based collaborative management, system host can set IP address, with TCP / IP network interface, can connect to wireless router, use computer, mobile phone, tablet and other devices in the same local area network to set the operation of the host, support Google Android / Microsoft Windows systems and other operating system
- With USB interface, can be connected to a computer to set up the controller
- With 2 control interfaces and control modes: 232 and 485 (6P Phoenix plug), and the cameras can be connected in series or in parallel:
- Built-in camera tracking function, supports PELCO-D, PELCO-P, SONY VISC and other communication protocols, and has the functions of adjusting, saving and calling preset positions;
- With video switching 232 interface (3P Phoenix interface), can connect HD video matrix
- With 4x1 HDMI HD seamless video switching interface(Optional-seamed or seamless)
- With central control code 232 interface (3P Phoenix interface), can be connected to central control system, and support SDK development kit secondary development
- With alarm signal input interface (alarm voltage interface +1
 2P Phoenix interface alarm audio input port), when the public
 address alarm system starts, it can automatically pause the
 conference and send alarm information to the participating
 units
- With 2 balanced audio output interfaces (3P Phoenix plug + XLR Canon), 5 unbalanced audio output interfaces (3P Phoenix plug + 6.35mm audio interface + 3 Lotus interfaces)
- With 2 unbalanced audio input interface (2P Phoenix plug and Lotus interface), can input external audio signals (such as background music or remote voice signals)
- With optical fiber extension interface, the sound quality is not attenuated during long-distance transmission, and two remote conference rooms can be merged into one.
- With 1 USB sound card interface and 2 sound card unbalanced output interfaces (2P Phoenix plug), it can be connected to a computer for remote conferencing.

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SUPPLIED ACCESSORIES

1 Power cable

1 USB cable

1 Installation and operating manual

SYSTEM ENVIRONMENTAL CONDITIONS

Working conditions fixed/stationary/transportable

Temperature range

Max.relative hurmidity < 95% (not condensing)

TECHNICAL SPECIFICATIONS

General

 Power supply.
 .AC110V-220V/50Hz

 Frequency response.
 .20Hz-20kHz

 S/N ratio.
 >96dBA

 T.H.D.
 <0.05%</td>

 Recording connector.
 .USB

RCA (unbal) x1 2P connector (unbal) x1

(alarm device)

Audio output 3P connector (bal) x1

2P connector (unbal) x2

XLR (bal) x1 6.35mm (unbal) x1 RCA (unbal) x3 Coaxial x1 Optical Fiber x1 Sound card (unbal) 2P connector x2

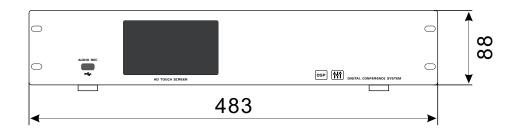
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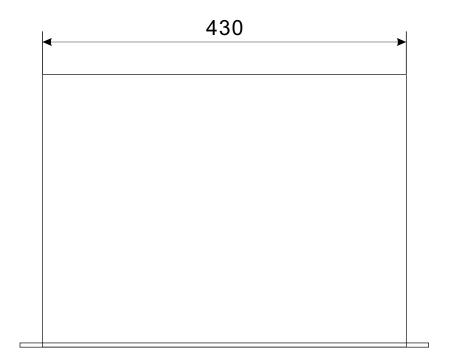
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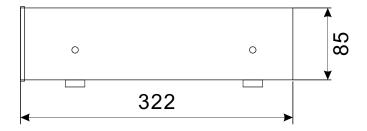
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DIMENSIONS (mm)







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