



Professional Audio Video Manufacturer

Audio Product & Audio-Link Solutions

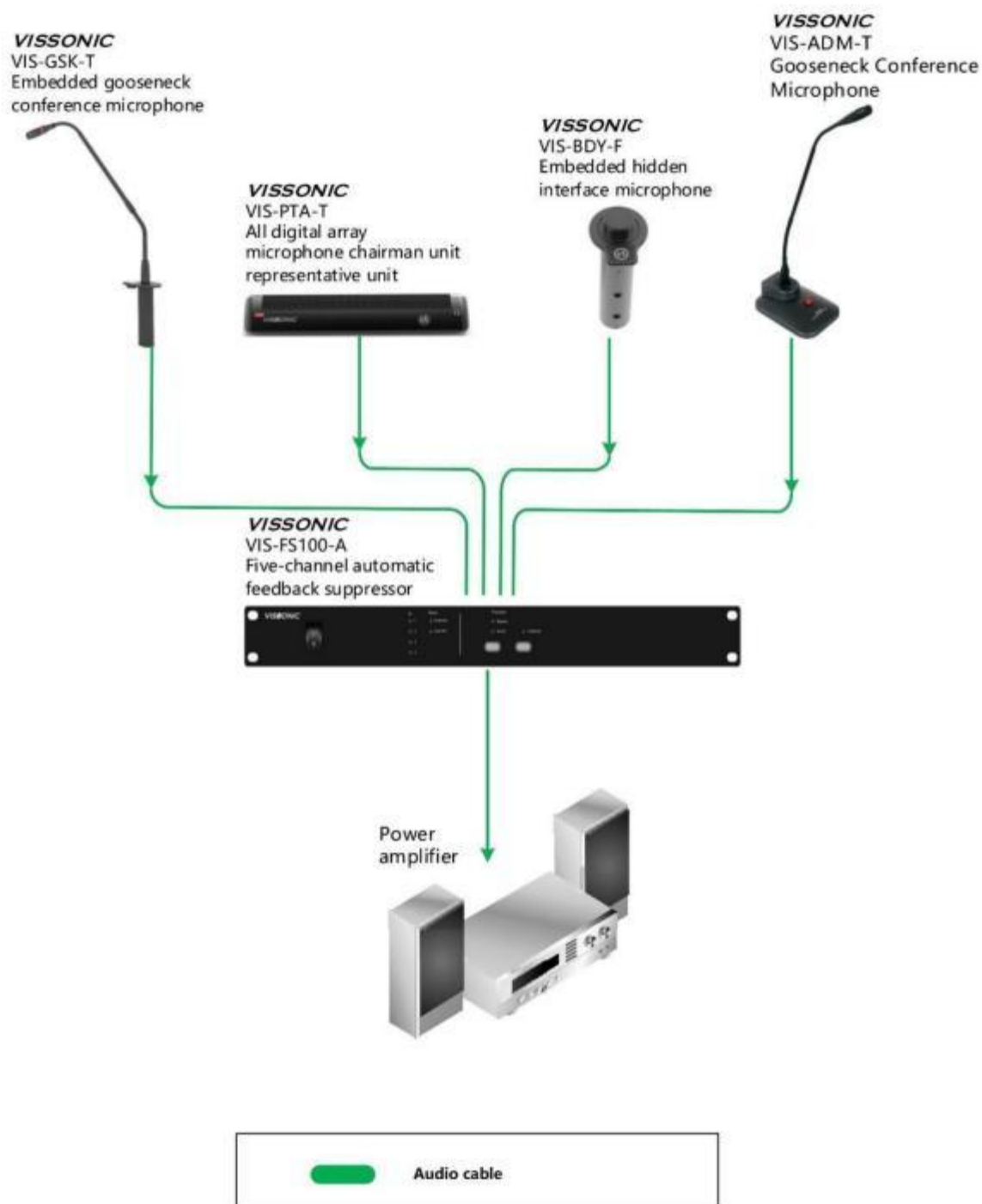
PRODUCT DATASHEET



VISSONIC ELECTRONICS LTD.

Think Solutions

Application diagram



AUDIO LINK Products

VIS-AIB

Microphone interface box



The VIS-AIB microphone interface box is designed to support VISSONIC conference microphone remotely installed under the table or armrest solution. With the phoenix connectors on both sides of the box, it can connect Dynamic mic/Boundary Condenser mic input, buttons and speakers on plate to control the conference unit etc. By the RJ45 port it can easily connect into existing VISSONIC conference microphone family to integrated into one part seamlessly.

It's an ideal solution for customized operation plate on the table and hide the main part of microphone, or any analog microphone like wired handheld type into the same system.

Features

- Compact metal cover design to mount under table or armrest of chair.
- VIS-AIB Microphone interface box is suitable for customized installation solutions. It allows to add a variety of functions to meet meeting needs. In addition, two separate audio inputs are provided, each with or without phantom power microphone, such as capacitor type or dynamic type.
- It also provides microphone switch button, priority button, approval button and corresponding LED light Phoenix interface, which is used to customize embedded button panel for gooseneck microphone.
- Adjustable sensitivity, compatible with different inputs
- RJ45 port to connect into VISSONIC networked system seamlessly
- It provides two phoenix stereo jacks for connecting to headphones, and two speaker outputs for connecting to speaker panels (VIS-SPK-F) or custom speakers.

Technical specs

Voltage supply 48V DC

Static power consumption 1 W

Frequency Response 20Hz ~ 20KHz

Page 3

Noise ratio (S / N)> 80dB

Total harmonic distortion <0.05%

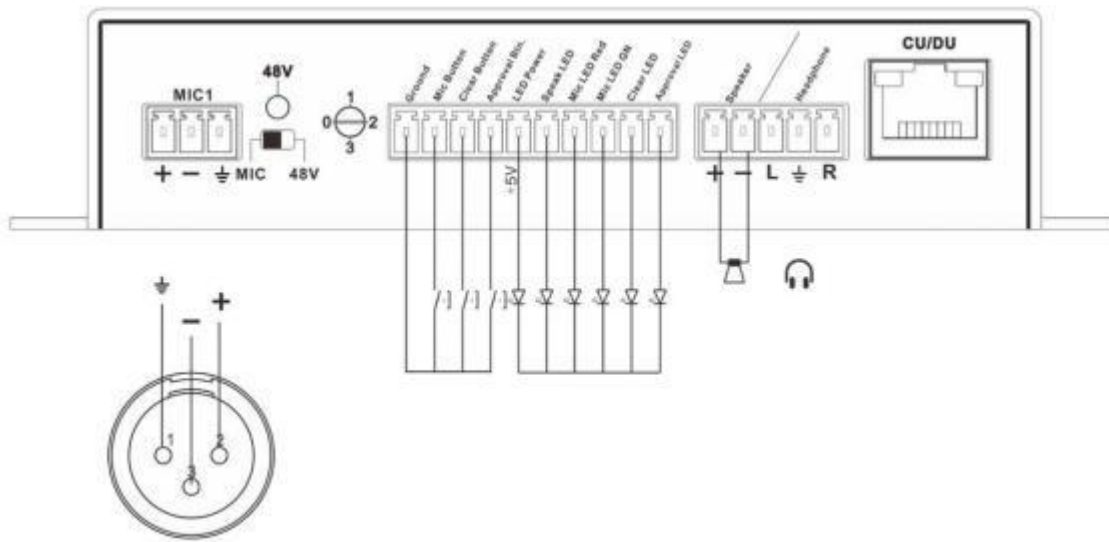
Channel Crosstalk> 80dB

Dimensions (mm) 120L x 100W x30H with two ears for screw fix

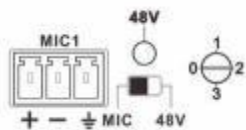
Color: Black

Weight: 0.5KG

Wiring Connections



Connectors



2xRJ45 port

2xBoundary condenser mic phoenix connector

2xSwitch for MIC/48V option

2xGain adjust knob



2x Group phoenix connectors including: Ground

Mic button for microphone ON/OFF

Clear button for chairman priority operation

Approval Button for chairman priority operation

LED Power: +5 V, max. 40 mA

Page 4

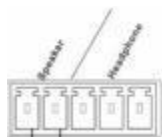
Speak LED indicate the gooseneck light color

Mic LED red indicate the microphone is ON of the conference unit

Mic LED green indicate the microphone is in request-to-speak mode of the conference unit

Clear LED shows the chairman is in priory mode to mute or close the delegate

Approval LED shows the chairman operating approval to speak



2 x 5 pin phoenix connector for speaker and headphone

Headphone Output independence: 16 Ω

Speaker output independence: 8 Ω

Frequency response: 20~ 15000hz

VIS-DCP2000-R

Full Digital Networked DSP Conference Processor for Array mic



Features

- The unique AUDIO-LINK digital ring network technology achieve full-digital signal transmission and processing.
- Single CAT5e cable transmits 64 channels of audio and various information.
- All microphone units support audio, ID independent output. Working with voice recognition system to achieve voice role separation.
- High-fidelity sound quality, lossless audio transmission technology, 48K audio sampling frequency, 20Hz ~ 20KHz response frequency.
- A single conference controller can realize the independent control and merging of 4 conference rooms, can handle the cascade of multiple conference controllers to realize the merging and separation of larger conference rooms (including conference management, etc.).
- "Hand-in-Hand-Loop-Network" connection makes the system always working perfectly no matter changing discussion unit or having any malfunctioned units.
- The sensitivity of the unit microphone and 8 segment equalizer EQ adjusted independently.
- The main controller comes with self-learning AFC automatic feedback algorithm, combined with array microphones, short-stem microphones to achieve ultra-long pickup distance.
- Support AGC (Automatic Gain Control)/AFC (Adaptive Feedback Cancellation)/ANC (Acoustics Noise Cancellation) Sound output by separated zones function, each zone is automatically adjusted according to the volume of the position of the microphone, achieve longer distances pickup without whistle.
- Echo cancellation, remote two-way real-time call with another conference controller or another third-party video conferencing terminal.
- USB recording, controller monitor, advanced noise cancellation technology to realize clearer recording.
- Display with date, time management, and support speech time, countdown speech.
- Optional DANTE output module with other DANTE equipment used.
- Can drive 2*120W POE speaker through extension main unit, support up to 16 POE speakers.

Functions

- Compliance with IEC 60914, GBT 15381-94.
- Connect with VIS-EXM extension main unit, it can expand to 5200 discussion units.
- "Hand-in-Hand-Loop-Network" connection and the system assigned to each unit independently ID to avoid ID

conflict.

- Both wired/wireless discussion units can connect to the controller, fast and easily to adding delegate of meetings.
- Supports up to 8 wireless microphones to be turned on at the same time, adaptive avoiding interference technology to achieve stronger anti-interference ability, compliance with IEEE 802. 11n standard Wi-Fi both for 2.4GHz and 5GHz.
- 2 channel audio input, XLR or RCA type, for local audio input or remote video-conferencing input.
- Maximum 8-channel audio output, XLR, RCA or Phoenix type, can be connected to simultaneous interpretation output or partitioned output.
- Add the audio input box, the system can be connected to other condenser microphones or dynamic microphones, more choices for users.
- Built-in multi-way internal communication function.
- TCP/IP connection between controller and PC
- RS232 central control, realize microphone switch, priority, volume and other controls.
- RS232 or RS485 to send the PELCO/VISCA protocol to realize camera auto-tracking function.
- Hot plug-and-play, and auto-recovery function.
- Wide range of voltage from 100V~240VAC
- Without using PC software, our conference controller still owns following settings:
 - Limit on the number of speaking persons: By Setting the number of active units at the same time (Number is 1/2/4/6/ units), supports up to open 32 wired microphones in full-on mode.
 - Discussion modes:
"OPEN" Free for all, limited by the active units setting by controller, "VOICE" Can activate MIC by detecting sound without press button, "OVERRIDE" First in First out, "APPLY" To apply then to speak, "PTT" Press to speak (Optional), "ALL" Full open mode (Optional).
 - Voting function: Approve/Abstain/Object
 - Chairman unit has priority to close all other units or mute them temporarily
 - Chairman unit can approval the application for speaking from delegate
 - Simultaneous language interpretation
- With PC software to add more ways of voting:
 - Approve/Object
 - Approve/Abstain/Object
 - Level: 1/2/3/4/5
 - Ranking: --/-/0/+/++
- By using RFID card to identify, can realize voting with names.

Control & Indicators

LCD display to show all the operations and results

4-Way button for menu operate

One-button press to start recording

Recording light indicator

Master volume control knob

Interfaces

Front panel USB recording port
3.5 mm headphone for monitoring
1 x RCA audio input
1 x XLR audio input
1 x RCA audio output
1 x XLR audio output
6 x Phoenix connector audio output
2 x RJ45 digital audio connection
2 x RJ45 for chairman unit, delegate unit, unit translators and other connection
1 x RJ45 for wireless AP or router
1 x DP9 male port for external control
1 x DP9 female port for camera
1 x three-pin phoenix connector for camera control keyboard
1 x RJ45 for computer connection or network switcher
1 x grounding screw
1 x 100 ~ 240V DC Power Interface

Technical Parameters

Power 100V / 240V AC
Static power consumption 12 W
Maximum power consumption 150 W
Frequency Response 20Hz ~ 20KHz
Noise ratio (S / N) > 80dB
Total harmonic distortion <0.05%
Channel Crosstalk > 80dB
Dimensions (mm) 483L x 260W x 43.6H
Dark black color

VIS-PTA-T

Analog Array Microphone Unit



The SONICON series array microphones designed by VISSONIC adopt the ARRA-Tech™ array microphone technology independently developed. The sound pickup distance is not less than 80CM. The corridor-type directional characteristics increase the acoustic transmission gain. The use of hidden microphones makes the conference room cleaner. The array microphone maintains a clear and stable pickup function, increasing the speaker's range of activity.

Features

- Unique modern design with touch button.
- Using the self-developed ARRA-Tech™ array microphone technology,
- The pickup distance is 60-80CM. The establishment of a personal pick-up sound area, whether the speaker is standing, backward or sideways, provide freely speaking experience.
- With no goose neck design, speakers and attendees can communicate without blocking the view.
- The hidden adjustable gain knob can quickly set up the whole system according to the field sound amplification conditions.
- 2m audio cable with XLR male connector for extending to the VISSONIC dB-NET series DSP audio matrix or any mixer system.

Technical Specification

| | |
|---------------------------------------|----------------------------|
| Microphone | Microphone Array |
| Polar pattern | Corridor |
| Control | Touch button for on/off |
| Frequency response | 30 - 18,000 Hz |
| Optimal distance of the speaker | 60 - 80 cm |
| Open circuit voltage at 1 kHz | approx. 200 mV/Pa |
| Output level with selector | 0dB, -5dB, - 10dB or -20dB |
| Nominal impedance | 200 Ω |
| Load impedance | > 1 kΩ |
| S/N ratio rel. to 1 Pa | 60 dB |
| T.H.D | <0.1% |

Voltage supply Phantom powering 48 V \pm 4 V to 6.8 k Ω
supply impedance 24 V \pm 4 V to 1.2 k Ω supply impedance
Power consumption 11 mA*
Connection 3 pins XLR

Item Model for Order

VIS-PTA-T..... Array Microphone with XLR connector
VIS-FS100-A..... 5 channel automatic feedback suppressor with Audiolink port
VIS-DSP8/ 12/ 16.....DSP audio processor

CLEACON VIS-BDY-F

Flush mount boundary Microphone



VIS-BDY-F is a built-in microphone that makes the microphone hide on the desktop when not in use. This is achieved by simply pressing down on the top of the microphone until it locks. Then press the microphone again, and the microphone will rise back to its original position. This microphone can be used in the CLEACON Full-digital network conference system through the VIS-AIB analog connection box, or it can be used in the analog system, such as VIS-FS100-A automatic feedback suppressor, mixer, audio processor, etc.

Features

- Low profile on any desktop or surface
- The microphone is equipped with a reed switch, which is designed to switch at the beginning and end of each session when the microphone is no longer used. When the microphone is in the up position, the switch contact is closed (ON), and when it is retracted, the switch will be OFF.
- Capacitive touch button can be optionally installed to easily set the microphone off or on for sound pickup.
- Cardioid pattern.
- Built-in microphone preamp power supply module, no need to use external power supply module
- The sturdy all-metal structure design provides two layers of steel mesh cover protection
- The microphone works from DC 11V to 52V phantom power.
- Output with 3-pin XLR or 3.5 Phoenix connector
- Built-in ring-shaped LED status indicator
- Built-in RF filter
- Equipped with shock-proof isolators to reduce vibration noise on the installation surface

Specification

Mic capsule electrostatic capacitive type

Directivity Pattern Cardioid

Frequency response 40- 19,000 Hz

Turn-on sensitivity -40 dB (10 mV) at 1V at 1 Pa
Impedance 200 ohm
High maximum SPL 142 dB SPL, 1 kHz at 3% T.H.D.
Dynamic range (typical) 113 dB, 1 kHz at highest sound pressure level
S / N..... 65 dB, 1 kHz at 1 Pa
Phantom power DC 11-52V, power consumption 5.7 mA typical
Switch Touch control On / Mute
Weight 65 g
Overall length 84.8 mm, diameter 38.0 mm, maximum width 48.0 mm
Output terminal Built-in 3-pin XLR
Standard configuration 1 pair of shock-proof isolator

Item Modelfor Order

VIS-BDY-F ... Flush mount boundary Microphone
VIS-AIB Analog microphone connection box
VIS-DCP2000-D Full Digital Network DSP Conference System Controller
VIS-FS100-A Fully automatic feedback suppressor



VIS-GSK-F

Flush Mount Microphone Base



VIS-GSK-F is a flush mount gooseneck microphone base that can be connected to all range of VISSONIC M series microphones, the microphone can be removed from the base when it's not in use. The overall base is small and exquisite, with a touch button to switch the microphone on and off. It can be connected to the VIS-AIB analog microphone connection box and used in the CLEACON conference system, or it can connect to an analog system, such as the VIS-FS100-A automatic feedback suppressor, mixer, audio processor etc.

Features

- Built-in microphone socket, suitable for VIS-M series 6-pin gooseneck microphone
- It is used for inconspicuous installation on the desktop with compact design.
- Capacitive touch switch button can be optionally installed to easily set the microphone off or on for sound pickup. The built-in switch can be used for control from remote equipment.
- Touch button comes with LED status light, operates with phantom power, provides white / red two colors
- Operation with 24-48V phantom power
- Output with 3-pin XLR or 3.5 Phoenix connector
- After inserting the microphone in place, the microphone comes with fasten lock, and the microphone can only be removed after unlock it
- Built-in RF filter
- Equipped with rubber isolators to reduce vibration noise on the installation surface

Specification

Output impedance: 360 Ω

control

High level when in down position: (DC + 5V);

Low level when in up position: (DC 0V)

LED indicator

High electric level DC + 5V): red /

Low electric level (DC 0V): white

Phantom power

DC: 24-48V,

Power Consumption: 4 mA typical

Switch

Touch control

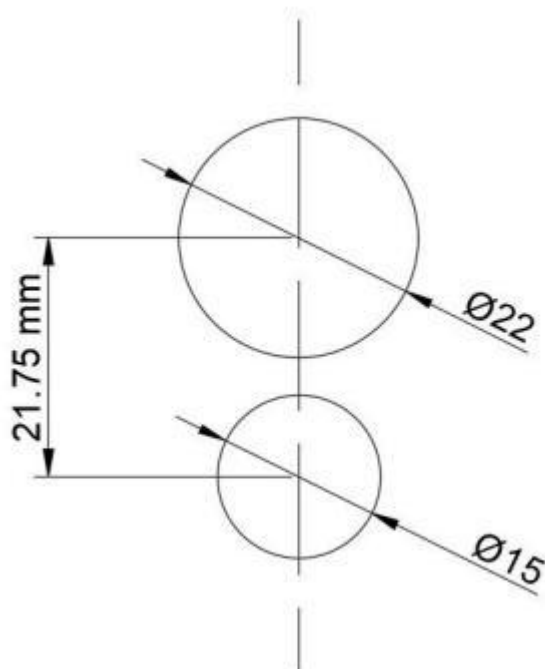
Weight: 65 g

Dimensions: length 128.2 mm, maximum width 55.3 mm

Accessories

One set of metal fixing ring for standard installation, one pair of shockproof isolators

Drill hole dimension



Blueprint

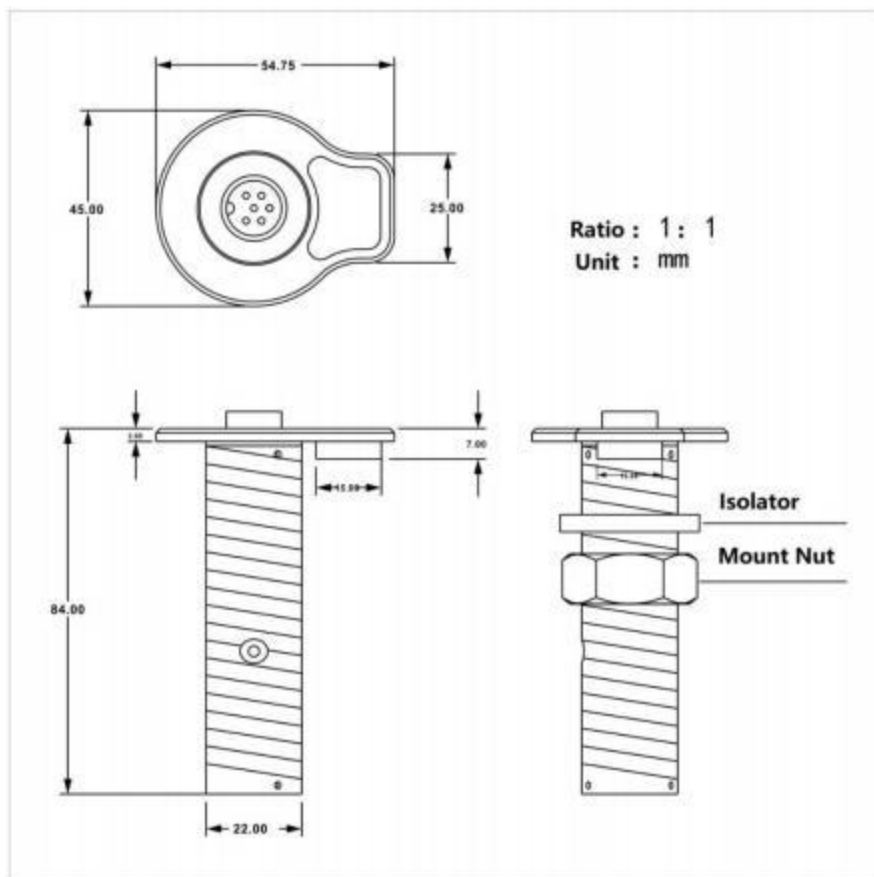
Dimension:

Diameter of the microphone body: 22mm

Touch button diameter: 15mm

The distance between the main body and the button center: 21.75mm

Product Dimension



Item Modelfor Order

- VIS-GSK-F Flush mount Gooseneck Microphone Base
- VIS-MDA Rectangular metal microphone
- VIS-M220 220mm microphone (black) with anti-wind foam cover
- VIS-M330 330mm microphone (black) with anti-wind foam cover
- VIS-M410 410mm microphone (black) with anti-wind foam cover
- VIS-M485 485mm microphone (black) with anti-wind foam cover
- VIS-M600 600mm microphone (black) with anti-wind foam cover
- VIS-AIB Analog microphone connection box
- VIS-DCP2000-D All Digital Network DSP Conference System controller
- VIS-FS100-A Fully automatic feedback suppressor

CLEACON VIS-CIB

Full digital interface box

**Features**

- The full-digital interface box is a conference product specially designed to meet the special needs of the conference. The exterior is mini and lightweight, and the all-metal thin shell design can cope with any environment and can be easily installed under the table and table. Indirect seams or armrests can be connected to the VISSONIC hand microphone for conference environments without a conference table. Or when there is a need to customize a dedicated conference panel or a panel with special features, VIS-CIB can meet the diverse needs of customers.
- Built-in DSP digital sound processor, 48kHz High-quality sound sampling frequency for AGC function of the picked-up audio for devices such as gooseneck microphones on connected panels or handheld microphones. At the same time, it also has ANC function to filter out the noisy background audio in the conference space. Through the AFC function, the feedback problem can be processed effectively and quickly, avoiding the whistling of the microphone and affecting the quality of the conference. And can transmit high-definition lossless digital audio through digital audio processing technology
- All-in-one interface box with chairman and delegate switching function, only need to connect with the conference controller, and quickly and easily switch the device function settings directly from the controller menu, without using the computer, through the web program or It is a software setting.
- Ability to use the Cat5e cable with all VIS-DCP2000 full-digital conference controller and conference device are connected to each other and support hot-swap function. Users can join or remove the conference device at any time during the conference according to the needs of the conference, and will not affect any microphone in the conference
- With digital ID configuration function, users can customize the representative ID address of the conference device before the conference by extending the external customized panel according to the needs of the conference, which can be used for sound group distribution and external output, or the conference microphone is used for the automatic tracking of the speaker image.
- When using an extended external custom panel, the VIS-CIB provides the following features:
 - The plug-in gooseneck microphone dual-color display function can display the speech or wait for the speech status.
 - For the panel of the chairman function: Provide the chairman function switch to turn off the delegate

function.

- When it is a panel of the function: Provide speaking switch.
- Provides speaker output function, which can be used to selectively set the conference system loudspeakers on the panel.
- Supports the speaking timing function, which automatically turns off the microphone when the speaking time is cut off.
- The conference microphone function can be activated by voice control through the voice of the speaker.

Specification:

Maximum power consumption.....2.0W
Frequency response.....20~20kHz
Total Harmonic Distortion (THD).....<0. 1%
Dynamic range.....>94dB
Signal to noise ratio (S/N)>96dB
Rated microphone input.....>70dB SPL
Maximum microphone input.....> 100dB SPL
Speaker rated output >70dB SPL at 0.5m
Sensitivity-46 dBV/Pa

Connect:

RJ-45 female seat 2
D-Sub 15 Pin 1
Color Black
Weight 300g
Size (mm) 100L x 60W x 24 H

CLEACON VIS-HM100

Handheld Microphone



The VIS-HM100 is a handheld conference microphone for the CLEACON system. The microphone is designed for the CLEACON series of conference units VIS-CIB. Applications for this microphone include a pressroom where the speaker needs to stand up, an auditorium, etc., or where the microphone must be hidden when not in use. The microphone features a low-noise condenser core design for clear speech and reduced feedback through a cardioid directional mode. The mic core is acoustically and mechanically isolated to prevent the microphone from picking up mechanical noise from the table. There is a large illuminated ring on the microphone that indicates when the microphone is turned on. The switch is used to turn the microphone on/off or to add a unit to the request queue. The microphone is supplied with a professional 6-pin XLR plug for use with spiral or straight cable.

Features

- LED ring indicator
- Heart-shaped pickup
- Rear electret condenser pickup
- Double rubber suspension pickup
- Highly close SPL level
- Low noise pickup
- Dual function switch
- Two-color LED

Specification

Pickup head..... Back-electret condenser
Frequency response..... 50Hz~ 18kHz
Power consumption..... 1.5 to 9 VDC
Dynamic range..... >94dB
Signal to noise ratio (S / N).....<60 dBA
Maximum microphone input..... > 138 dB SPL
THD..... <0. 1%
Speaker rated output.....> 70dB SPL at 0.5m
Sensitivity..... -68 dB±3dB /μBar
Impedance.....2 kOhms @ 1 kHz
connection:
Mechanical connector 6 - pin XLR male
Color Black
Weight 0.2kg
Dimensions (height x diameter) 160 x 30mm

How to Order

VIS-HM100 Handheld microphone with speak request button and light ring
VIS-HMC-02 2m 6-pin XLR-f to 15HD-sub-m spiral cable
VIS-HMC-03 3m 6-pin XLR-f to 15HD-sub-m spiral cable

VIS-FS100-A

Automatic feedback suppressor



VIS-FS100-A feedback suppressor uses a powerful signal processor and a fully automatic adaptive algorithm, which is easy to use. In addition to supporting 4-line / microphone input, it is also equipped with an AUDIOLINK digital audio cascade interface and supports 5 channel automatic mixer.

Features

- Fully automatic high-speed howling suppression algorithm, no need to debug, ready to use
- Eliminate feedback
- Automatically adapt to the acoustic environment
- Up to 12DB additional gain to effectively prevent feedback
- Equalized line or microphone input to provide phantom power
- With AUDIOLINK digital cascade input and output interface
- Four microphone inputs and digital interface AUDIOLINK inputs with automatic mixer

Function

- The adaptive filter can switch between "Fast" mode and "Precise" mode. Fast mode is suitable for situations where the microphone position changes frequently. Adaptive filters converge at a slower rate, suppressing more reverberation components. Depending on the acoustic environment and the selected operating mode, an additional gain of up to 12DB can be obtained before acoustic feedback occurs.
- The feedback suppressor also has a built-in automatic mixer to achieve four microphone inputs. In many cases, such as on a podium, pulpit, or conference table, the use of the above two microphones can better capture the voice of a moving speaker, which often increases the risk of acoustic feedback. In view of this problem, the automatic mixer in the feedback suppressor will automatically reduce the gain of the microphone with a weak signal input, while increasing the gain of another microphone with a strong signal input. In this way, it can track the speaker on the move, producing the best sound clarity. At the same time, since the total amount of gain remains constant, a maximum volume is maintained without generating acoustic feedback.
- When the feedback suppressor is configured with two AUDIOLINK cascade interfaces, it is connected to the CLEACON series conference controller through the AUDIOLINK interface, and the conference controller outputs audio to the feedback suppressor for mixing output, and can be cascaded to the next feedback suppressor or Other devices with AUDIOLINK interface.

- Even if the feedback suppressor is switched to the "bypass" mode, the automatic mixer function still works.

Controls and indicators

- Switch
- Bypass button with yellow LED
- Activate ACTIVE button with yellow LED
- Calibration button. Start quick calibration cycle
- 4-channel microphone / line input status indicator
- Overload (at 0DBFS), red LED
- Enable Auto Mix, yellow LED
- Calibration, yellow LED

Specification

Electrical characteristics

Power supply voltage230VAC / 115VAC, $\pm 10\%$, 50 / 60Hz

Maximum power consumption 15W

Maximum starting power 1.5A (230VAC) / 3A (115VAC)

Sampling rate..... 32KHZ

Frequency response..... 125HZ- 15KHZ

< 1% at 1KHZ distortion

Gain0dB line in, 12dB / 0dB / - 12dB (Low / Med / High) microphone input

Gain adjustment range..... Mute ~ 20dB

S / N ratio.....> 90DB

Signal delay< 11 ms

Decoupler frequency variation..... up to 5HZ shielding noise

Line / microphone input

Maximum input level..... 6dBv

Output resistance..... < 100 ohm

CMRR.....> 25DB (50HZ-20KHZ)

24V phantom power (microphone only, switchable)

10dBv maximum output level

conventional

Relative temperature <95%

Weight 3 kg

Operating temperature range from 10 °C to 55 °C

Storage temperature range 40 °C to 70 °C

Relative humidity <95%

Item Model for Order

VIS-FS100-A Five-channel Fully Automatic Feedback Suppressor

VIS-FS100 Four-channel full-automatic feedback suppressor

VIS-DSP8/DSP12/DSP16

dB-NET series Digital DSP Audio Matrix



Features

- 8, 12, 16 CH balanced microphone input, using the phoenix plug
- 8, 12, 16 CH balanced microphone output, using the phoenix plug
- A variety of audio formats in front panel can extend the USB multimedia storage for recording.
- Can control by PC, mobile phone, tablet, web, key panel, Control system and so on
- Main algorithms, such as built-in signal generator, auto mixing, feedback suppressor, echo cancellation, noise cancellation(optional) etc.
- Input per channel: preamplifier, signal generator, expander, compressor, 5 parameters of equilibrium.
- Output per channel: 31 segment graphic equalization, delay, divider, limiter
- Full functional matrix mixing, with the built-in patented component matrix adjustment function.
- Built-in automatic video tracking function, easy to implement video conferencing, support scene presupposition function, power automatic protection memory function
- There are more than dozens of professional audio processing modules, such as: 5-band full parametric equalizer, 31-band graphic equalizer, high-precision compression and limiter, high-sensitive expansion and automatic gain, crossover, automatic mixing Amplifier, delayer, matrix mixer, component matrix adjuster, noise threshold, mute module, signal generator and signal indicator level meter, etc.
- The built-in media control code generator does not require the self application formula to edit the code.
- The equipment provides a universal programmable I/O port and sets firefighting functions.
- RS-232 bidirectional serial control interface; it can control other external devices such as video matrix, video camera and other RS-232 devices, or receive third party RS-232 control
- The phantom power supply (each input) +48 VDC10mA.

Technicalparameter

Performance parameter

Number of analog channels 8 channels input + 8 channels output (DSP8), 12 channels input + 12 channels output (DSP12), 16 channels input + 16 channels Output (DSP16)

Core algorithm automatic mixing, feedback cancellation, echo cancellation

GPIOinput and output share 8

RS232/RS485 1

RJ45 control interface 1
USB interface .. 1
Simulated maximum gain 51 dB
Sampling rate48K
Frequency response (20~20KHz)±0.2dB
A/D dynamic range (A-weighting) ... 114dB
Digital/analog dynamic range (A-weighting) ... 120dB
Input to output dynamic range..... 108dB
Total harmonic distortion + noise.....<0.002% @1KHz ,4dBu
Noise floor (A-weighting)90dBu
Delayed storage.....2s
Analog input to analog output system delay.....3ms
Input impedance (balanced).....20KΩ
Output impedance (balanced) .. 100Ω
Maximum input level+24dBu, balanced
Maximum output level+24dBu, balanced
Equivalent input noise EIN (20-20kHz, A weighting)≤- 131dBU
Phantom power (per input)48V
Input common mode rejection, 60Hz...80dB
Channel isolation, 1kHz..... 100dB
Power consumption.....<40W
Working power...AC110V-220V, 50Hz/60Hz

Mechanical parameters

Installation: 19 "standard cabinet installation
Dimensions (W x D x H)483*260*44.5(mm)
Working temperature.....-20°C~60°C
Relative humidity.....5%~95%
Storage and transportation temperature.....-40°C~80°C
Weight.....3KG

VIS-SPK12/8/6/4/2-AL Series

Full frequency linear POE column speaker



The SPK series full-frequency linear POE sound column adopts a 3-inch full-frequency unit, a wooden speaker, and a built-in POE amplifier, which has a high degree of sound reduction and DSP processing. The speaker only needs to connect a network cable to realize audio, power supply, DSP configuration and real-time signal control.

Features

- Built-in DSP processing, support gain, mute, equalization, pressure limit, noise gate software adjustment
- AUDIOLINK digital transmission protocol, ultra-low latency, wireless loss and interference
- A network cable transmits audio signals, control signals, and POE power
- Overcurrent protection, high temperature protection, overload protection

Specification

| Model | VIS-SPK12-AL | VIS-SPK8-AL | VIS-SPK6-AL | VIS-SPK4-AL | VIS-SPK2-AL |
|----------------------------|-----------------------------|----------------------------|----------------------------|----------------------------|----------------------------|
| System Components | 12 × 3 "full frequency unit | 8 × 3 "full frequency unit | 8 × 3 "full frequency unit | 8 × 3 "full frequency unit | 8 × 3 "full frequency unit |
| Frequency response (± 3dB) | 120Hz- 18kHz | 120Hz- 18kHz | 130Hz-20kHz | 135Hz-20kHz | 140Hz-20kHz |

| | | | | | |
|------------------------------|----------------------------|----------------------------|----------------------------|----------------------------|----------------------------|
| Sensitivity | 97dB | 96dB / w / m | 95dB / w / m | 93dB / w / m | 91dB / w / m |
| Maximum sound pressure level | 121dB | 120dB | 119dB | 117dB | 113dB |
| power | 120W | 120W | 120W | 120W | 80W |
| Coverage angle | H100 ° × V15 ° | H100 ° × V18 ° | H100 ° × V20 ° | H100 ° × V30 ° | H100 ° × V40 ° |
| interface | 1 way RJ45 | 1 way RJ45 | 1 way RJ45 | 1 way RJ45 | 1 way RJ45 |
| Material | Wooden | Wooden | Wooden | Wooden | Wooden |
| Cooling method | Natural convection cooling | Natural convection cooling | Natural convection cooling | Natural convection cooling | Natural convection cooling |
| size | 124×155×1060 m m | 120×150×720 mm | 120×150×540m m | 120×150×370m m | 120×150×210m m |
| net weight | 12kg | 8.5kg | 6.5kg | 4kg | 2.5kg |

VIS-SPK101-AL/VIS-SPK801-AL/VIS-SPK602-AL

POE active speaker



Overview

The cabinet of speaker is made of multi-layer high-density birch plywood and built-in POE amplifier, which has a high degree of sound reduction and DSP processing. The concave and convex groove process is used to make the cabinet more solid. At the same time, the use of high-strength reinforcement nails causes the resonance of the cabinet. The energy distortion is minimized. The cabinet is designed to be moisture-proof and dust-proof to ensure stable operation in dusty and humid environments. The speaker only needs to connect a network cable to realize audio, power supply, DSP configuration and real-time signal control.

It's suitable for high-end performances, conferences, evenings and concert system.

Features

- Built-in DSP processing, support gain, mute, equalization, pressure limit, noise gate software adjustment
- AUDIOLINK digital transmission protocol, ultra-low latency, wireless loss and interference
- A network cable transmits audio signals, control signals, and POE power
- Overcurrent protection, high temperature protection, overload protection

Specification

| Model | VIS-SPK101-AL | VIS-SPK801-AL | VIS-SPK602-AL |
|--------------------|--|---------------------------------------|---|
| System Components | 1 × 10" + 1 × 1.33"/50mm + 34mm(VC) | 1 × 8" + 1 × 1.33"/50mm + 34mm(VC) | 2 × 6.5" + 1 × 1.33"/36mm + 34mm(VC) |
| Frequency response | 60Hz-20kHz | 65Hz-20kHz | 70Hz-20kHz |

| | | | |
|------------------------------|----------------------------|----------------------------|----------------------------|
| (± 3dB) | | | |
| Nominal impedance | 8 Ohm | 8 Ohm | 4 Ohm |
| Sensitivity | 97dB/w/m | 93dB / w / m | 95dB / w / m |
| Maximum sound pressure level | 128dB | 120dB | 122dB |
| power | 180W (AES), 720W (PEAK) | 150W (AES), 600W (PEAK) | 190W (AES), 760W (PEAK) |
| Coverage angle | H90 ° × V60 ° | H90°×V60° | H90°×V90° |
| interface | 1 way RJ45 | 1 way RJ45 | 1 way RJ45 |
| Material | Wooden | Wooden | Wooden |
| Cooling method | Natural convection cooling | Natural convection cooling | Natural convection cooling |
| size | 265×320×500mm | 261×282×448mm | 215×230×542mm |
| net weight | 13.2kg | 9.7kg | 11kg |

VIS-X8D-AL

POE ceiling speaker

***Overview***

This POE ceiling speaker is installed on the ceiling. It is built-in POE amplifier, which has a high degree of sound reduction and DSP processing. The speaker only needs to connect a network cable to realize audio, power supply, DSP configuration and real-time signal control. Ceiling speakers are widely used at Office buildings, hotels, and shopping malls are public places that use ceiling speakers.

It's suitable for high-end performances, conferences, evenings and concert system.

Parameters

Frequency response ($\pm 3\text{dB}$): 60Hz-20kHz

Unit: 1×8" coaxial

Nominal impedance: 8Ω

Power: 160W

Sensitivity: 89dB/w/m

Maximum sound pressure level : 127dB

Interface: 1 way RJ45

Product size: $\Phi 305 \times 215\text{mm}$

Hole size: $\Phi 270\text{mm}$

Net weight: 4.5kg

Material: galvanized iron drum

VIS-HL002/ VIS-HL003/ HL005/ HL010/ HL020/ HL050/HL100

Professional CAT5 E conference extension cables

***Features***

- TIA/EIA 568B CAT5E
- ISO/IEC 11801 ClassD
- CELENEC EN50173, EN50167, EN50169
- UL C(UL)&ETL Verified
- Pre-test by VISSONIC

Functions

- Extension cable for main unit to CU/DU unit
- Carry 48DC power and signals
- Male to Male cable
- Size from 2m to 100m

Interfaces

2x male RJ45

Technical Parameters

Standard: TIA/EIA 568B CAT5E

Connector: Shielded RJ45

Shielding: F/UTP

Resistance: 100ohm \pm 15%, 1MHz to 100MHz

Delay skew: 25ns/ 100m max

Gauge: 24AWG solid bare copper

Flame retardant rating: CM, IEC332/ 1

Storage Temperature: -20 C~80 C

Operation temperature: -20 C~60 C

Item Model for Order

| | |
|----------------|---|
| VIS-HL002..... | 2m black shielded Cat5e U/FTP cable (shielded RJ45) |
| VIS-HL003..... | 3m black shielded Cat5e U/FTP cable (shielded RJ45) |
| VIS-HL005..... | 5m black shielded Cat5e U/FTP cable (shielded RJ45) |
| VIS-HL010..... | 10m black shielded Cat5e U/FTP cable (shielded RJ45) |
| VIS-HL020..... | 20m black shielded Cat5e U/FTP cable (shielded RJ45) |
| VIS-HL050..... | 50m black shielded Cat5e U/FTP cable (shielded RJ45) |
| VIS-HL100..... | 100m black shielded Cat5e U/FTP cable (shielded RJ45) |

About VISSONIC Electronics Limited

Our mission is to develop and manufacture the most comprehensive and innovative audio visual products for our clients. We provide the best performance/price ratio products because it could give you satisfaction just from the time you use them, we believe the good design with cutting edge technology on products will provide value to all our partners and end users. Listen to your demands, we fulfill it.



4/F, Building 6, No. 50 Nanxiang 1st Road, Huangpu District, Guangzhou, China

• Tel: +86-020-82515140 • E-mail: info@vissonic.com

@2022 VISSONIC Electronics Ltd. all rights reserved.