

# US-USM-931-H / 933-H

Vandal resistant wallmount substation



## The Intercom solution for security-critical applications

Tamper/Weather resistant security substation connects to Commend servers via Cat 5 cable. It is used for either stanchion or security wall housing mounts, when both attractive appearance and extreme durability are important.

### US-USM931H

- SIP/IOIP Substation
- Full open duplex communications
- Connects using Cat. 5 cable
- Rugged 11 gauge stainless steel faceplate
- True tamper resistant construction
- Rugged call button
- Weather resistant
- Sensitive electret microphone
- Mounts in most major stanchions and security wall housings
- Mic Open LED
- Full Supervision (speaker, microphone, station electronics, cable, network)
- Red Piezo button (US-USM931H)
- Red Mushroom button (US-USM933H)

### US-USM931H-2T

- SIP/IOIP Substation
- Full open duplex communications
- Connects using Cat 5 cable
- Rugged 11 gauge stainless steel faceplate
- True tamper resistant construction
- 2 Rugged call buttons
- Weather resistant
- Sensitive electret microphone
- Mounts in most major stanchions and security wall housings
- Mic Open LED
- Full Supervision (speaker, microphone, station electronics, cable, network)
- Red Piezo Button
- Second Black Piezo button

# US-USM-931-H / 933-H

## Technical specifications

### Technical Data US-USM-931-H / 933-H with ET-908H Intercom Module

|                                      |  |
|--------------------------------------|--|
| <b>Operating temperature range:</b>  | -40 °C to +70 °C (-40 °F to +158 °F) <sup>1)</sup>   |
| <b>Storage temperature range:</b>    | -40 °C to +70 °C (-40 °F to +158 °F) <sup>1)</sup>   |
| <b>Relative Humidity:</b>            | up to 95% not condensing   |
| <b>Microphone input:</b>             | possibility for connection of an electret condenser microphone or a dynamic microphone<br>electret condenser microphone: nominal -43 dB/Pa (feeding voltage: 2.5 V at 3.3 kΩ)  |
| <b>Loudspeaker output:</b>           | 10 W at 4 Ω/6 W at 8 Ω<br>max. 6.3 V <sub>eff</sub> (volume level "11")  |
| <b>Ext. microphone, loudspeaker:</b> | possibility for connection of e.g. a headset/handset<br>EP output: max. 880 mV <sub>eff</sub> (volume level "11") ,<br>R <sub>i</sub> = 200 Ω<br>EM input, nominal level: 14 mV on 3.3 kΩ (feeding voltage 2.5 V)  |
| <b>Line input:</b>                   | for feed-in of audio (e.g. music, radio conference)<br>nominal level 0 dBu (0.775 V) at 10 kΩ  |
| <b>Amplifier:</b>                    | Built-in amplifier class "D" with 2.5 W  |
| <b>External LED:</b>                 | possibility for connection of an RGB-LED   |
| <b>Inputs:</b>                       | 3 inputs for floating contacts<br>(IoIP: detection of 5 input states)  |
| <b>Outputs:</b>                      | 2 relay outputs (switch-over contacts)<br>max. 60 W (DC)/37.5 VA (AC)<br>max. 2 A<br>max. 60 VDC/30 VAC<br>expected life: min. 5 x 10 <sup>4</sup> (2 A), 10 <sup>5</sup> (1 A)  |
| <b>Audio bandwidth:</b>              | IoIP: 16 kHz<br>SIP: 7 kHz   |
| <b>Frequency range:</b>              | 50 - 16,000 Hz<br>(depending on connected loudspeaker)   |
| <b>Connection:</b>                   | spring clamp terminals<br>(conductor cross-section: 0.2 – 1.5 mm <sup>2</sup> )<br>expansion plug, e.g. for EB2E2AHE<br>shielded RJ45 modular jacks<br><br>– IP Uplink/Downlink:   |
| <b>Device class:</b>                 | ES1, PS2 as per IEC/EN 62368-1<br>UL 62368-1 and CAN/CSA C22.2 No. 62368-1-14, Audio/video, information and communication technology   |
| <b>Cabling:</b>                      | Cat. 5 cable;  |
| <b>Power supply:</b>                 | 24 VDC (15 – 28 VDC), max. 1 A or PoE  |
| <b>PoE (power over ethernet):</b>    | following IEEE 802.3af<br>power consumption of the terminal device:<br>36 – 57 V, 15.4 W, Class 0 (0.44 to 12.95 W)  |
| <b>Protocols (IoIP):</b>             | IoIP-Protocol based on UDP/IP  |
| <b>Protocols (SIP):</b>              | IPv6, IPv4, TCP, UDP, HTTP (RFC 2617, RFC 3310),<br>RTP (RFC 3550), RTCP, DHCP, SDP (RFC 2327),<br>SIP (RFC 3261), SNMPv2, STUN, TFTP, URI (RFC 2396),<br>DTMF Decoding (RFC 2876, RFC 2833),<br>SIP User Agent (UDP RFC 3261),<br>SIP Refer Method (RFC 3515) |
| <b>Codecs (SIP):</b>                 | G.711 a-Law, G.711 μ-Law, G.722  |
| <b>Data rate:</b>                    | 2 x 10/100 MBit/s (Full/Half Duplex) auto MDIX   |
| <b>Measurements:</b>                 | 9.56 x 11.81 see drawing page 4  |
| <b>Weight:</b>                       | about 220 g (0.5 lbs)  |

<sup>1)</sup> Temperature range for MIC 480: -20 °C to +70 °C (-4 °F to +158 °F)

### Product variations US-USM-931-H

|                          |  |
|--------------------------|--|
| <b>US-USM-931H-11</b>    | Red piezo call button, with Emergency in red   |
| <b>US-USM-931H-12</b>    | Red piezo call button, with Assistance in black  |
| <b>US-USM-931H-13</b>    | Red piezo call button, with Help in black  |
| <b>US-USM-931H-14</b>    | Black piezo call button, with Assistance in blue   |
| <b>US-USM-931H-15</b>    | Black piezo call button, with Assistance in black  |
| <b>US-USM-931H-2T-11</b> | 1 red piezo call button, with Help in red, and 1 black piezo call button with Call in black                |
| <b>US-USM-931H-2T-12</b> | 1 red piezo call button, with 911 in red, and 1 black piezo call button with Help in black                 |
| <b>US-USM-931H-2T-13</b> | 1 red piezo call button, with Emergency in red, and 1 black piezo call button with Information in black    |
| <b>US-USM-931H-2T-14</b> | 1 red piezo call button, with Emergency in red, and 1 black piezo call button with Assistance in black     |
| <b>US-USM-931H-2T-15</b> | 1 red piezo call button, with 911 in red, and 1 black piezo call button with Security Information in black |
| <b>US-USM-931H-ST-1</b>  | 1 red piezo call button, with Emergency in red and square self testing call button,                        |

Other buttons and text available. Please contact Commend for a custom quote.

### Product variations US-USM-933-H

|                         |  |
|-------------------------|--|
| <b>US-USM-933H-11</b>   | Red mushroom call button, with Emergency in red                                    |
| <b>US-USM-933H-12</b>   | Red mushroom call button, with Assistance in black                                 |
| <b>US-USM-933H-13</b>   | Red mushroom call button, with Help in black                                       |
| <b>US-USM-933H-14</b>   | Red mushroom call button, with Assistance in blue                                  |
| <b>US-USM-933HX3LED</b> | Red mushroom button and 3 LEDs with Call Placed, Call Received and Help on the Way |

### Benefits

- Communication via data networks- no additional cabling required
- Crystal clear 16 kHz speech quality for optimum intelligibility
- Integration in existing housings and panels
- Connection possibilities for-
  - Microphone and loudspeaker
  - Multiple tamper resistant call buttons
  - IP-devices (IP-camera, parking terminals, etc.)
  - LED for indication of conversations
- Inputs and outputs for control and indication functions e.g. control of barriers
- Local power supply or POE (intercom module only)
- In case of a changing public IP-address, dynamic registration is possible
- Supports DSP-functionlities such as Open Duplex®, Audio Monitoring, Loudspeaker-/Microphone Surveillance, etc.
- Option boards offer additional powerful features
- Line input for feed-in of audio
- Configurable Ethernet parameters
- Easy administration via intercom server configuration

## Requirements to the network for use as SIP device

### Ports

- The configuration via the web interface is done via TCP port 80 (cannot be configured).
- The communication from the SIP device to the SIP server is done via the following ports (both are configurable):
  - SIP: UDP port 5060
  - RTP: UDP port 16384 (incoming)

## Requirements to the network for use as IoIP device

### IP addresses and ports

- For the USM 931H, the DHCP functionality is available. If DHCP is not used, the USM 931H must have a fixed IP address.
- In case of a changing public IP address, dynamic registration of the USM 931H is possible.
- Communication from the program IP Station Config is done via port 16399 (cannot be configured).
- Communication from the USM 931H to the Intercom Server (UDP protocol) is done via port 16400 (configurable).

### QoS requirements

- One-way delay max. 100 ms
- Delay jitter max. 50 ms
- 0% packet loss for perfect audio quality

### Bandwidth

For further information on bandwidth, see guideline "IoIP Technology"

## Compatibility SIP PBX

Generally, the SIP device can be used with any SIP server.

The following server types have been tested explicitly by Commend and therefore a proper functionality can be confirmed:

| Manufacturer <sup>1)</sup> | Type   | Version                               |
|----------------------------|--|---------------------------------------|
| Cisco                      | Cisco Call Manager<br>Cisco Unified Communication Manager        | Versions 5, 6, 7, 8, 9                |
| Digium                     | Asterisk   | Versions 1.2, 1.4, 1.6                |
| Avaya (former: Nortel)     | CS1000   | Version 6                             |
| Avaya                      | Avaya Aura™ (Avaya Communication Manager, Avaya Session Manager) | Release 6.1                           |
| Innovaphone                | Virtual Appliance IPVA   | Version 9 final                       |
| Alcatel                    | OmnipCX Enterprise (OXE)   | Release 9                             |
| Siemens                    | Hipath 4000<br>Hipath 3000 + HG 1500                             | Version 5                             |
| 3CX                        | 3CX for Windows  | 3CX PhoneSystem<br>Versions 9, 10, 11 |
| Starface                   | Starface free  | Versions 4.x, 5.x                     |
| Aastra (former: Ericsson)  | MX-ONE   | Version 4.1 SP 1                      |
| Kamailio                   | Kamailio (OpenSER)   | Version 3.3.0                         |
| FreeSWITCH                 | FreeSWITCH   | Version 1.1 Beta1                     |
| ELMEG                      | elmeg ICT880   | Version 7.67D                         |
| 2N®                        | 2N® Netstar IP   | Version 3.1.0.96                      |
| AVM                        | Fritz!Box Fon 7170<br>Fritz!Box Fon 7270                         | Version 29.04.87<br>Version 54.05.05  |
| Sipgate                    | sipgate.de   | tested in Dec 2010                    |
| Vodafone Arcor             | vodafone.de  | tested in Jan 2011                    |
| blue SIP                   | blueSIP.net  | tested in May 2011                    |
| Mitel                      | 3300ICP  | 12.0.0.49                             |

<sup>1)</sup> The listed products and company names are brand names or registered trademarks of their respective owners.

## System requirements

### IoIP

#### Intercom Server

- GE 800 (min. PRO 800 6.3) with G8-IP (min. version 6.6A) or
- GE 300 (min. PRO 800 6.3) with G3-IP (min. version 6.6A) or
- IS 300/G8-IP-32 (min. PRO 800 6.3, min. version 6.6A) or
- S3/S6/VirtuoSIS (min. version 7.1)

#### Configuration software

- CCT 800 (min. version 7.1)
- IP Station Config (included in setup of CCT 800)

### SIP

- Compatible SIP server or
- S3/S6/VirtuoSIS (min. version 7.1) or
- GE 800 with G8-VOIPSERV or
- Serverless operation

#### Device firmware

- IoIP-Device (min. version 7.2)
- SIP Series (min. version 3.8.1, build 61)

### ATTENTION:

Downgrading to firmware version SIP Series 3.9 build 24 or lower is not supported.

### Line length in LAN

The maximum line length of Cat. 5 cabling in a LAN is 100 m (328 ft)  
- e.g. from switch to Intercom station.

# US-USM-931-H / 933-H

## Installation instructions

### Precautions / Mounting Info

- When opening the stations ESD precautions must be observed.
- The stations may only be opened by authorized service engineers.
- Do not expose the station to extreme temperature (below -40 °C or above +70 °C / -40 °F to +158 °F)
- For stations to be mounted in outdoor areas, the screws must be closed with a sealing compound.

**Note:**

On two-button models, buttons connect with 0 and T.

**All connected circuits shall fulfill the following requirements:**

- Safety Extra Low Voltage (SELV) and Limited Power Source (LPS) according to IEC/EN 60950-1 or
- ES1, PS2 circuits and Annex Q (Limited Power Source) according to IEC/EN/UL 62368-1
- UL 62368-1 and CAN/CSA C22.2 No. 62368-1-14, Audio/video, information and communication technology

### Compatible Stanchions

|                   |   |
|-------------------|---|
| <b>US-CSH-01</b>  | Wall mounted stanchion                    |
| <b>US-CSH-02</b>  | Wall mounted blue light stanchion         |
| <b>US-CSH-02A</b> | Compact wall mounted blue light stanchion |
| <b>US-CSH-03B</b> | 9" stanchion                              |
| <b>US-CSH-04</b>  | 7" stanchion                              |
| <b>US-CSH-06F</b> | Flush mount housing                       |
| <b>US-CSH-06S</b> | Surface mount housing                     |

### Extent of supply

- Intercom terminal
- Security Screws (10-24)

