IP Call Station (IP Intercom)

Model: PROV-H-610-P

PROVCOM

Vandal resistant SIP station



Main features

One key to dial emergency call for help, realize the bidirectional full-duplex voice intercom

Full duplex functionality with high volume capability for natural, convenient hands free communication Panel adopts advanced wiredrawing aluminum alloy

material, high strength damage resistant button, strong and durable

Embeddable wall or mounted in the wall, for avoid cable leakage users, to ensure that the internal network information security, provide configuration and remote configuration. Built-in speakers and high sensitive microphone, support hands-free calls and receive emergency broadcast

System integration linkage function, trigger linkage function, recording management functions

Extendable with Loudspeaker and Direct Dialing Modules Can be operated without a Server Conforms to SIP standard for seamless integration into SIP capable telephone systems



Electret microphone; two 8 Ω loudspeakers; 2.5 W Class "D" amplifier; 3-colour (RGB) call status LED; 2 relay outputs (switching contacts); 3 inputs for floating contacts; ; SNMP for call station monitoring; Integrated web server for configuration and firmware updates.



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Hardware Specifications	
Panel	High strength aluminum alloy material, Dimension 162*111*42 mm
Button	High strength damage resistant industrial buttons
Speaker	3W 8Ω
Microphone	High sensitive microphone
Network Interface	2 RJ-45(LAN & Device Monitor)
3.5 mm Interface	Two(a record, a Trigger the alarm or other notification interface)
2 pin Interface	Four (The input and output control, can answer the monitoring equipment or other security equipment)
Power	PoE or DC:12V
Weight	Net weight: 1.1 kg , gross weight: 2.0 kg
Work Environment	Temperature: 0 ~ 60 degrees of humidity: 10% ~ 90%
Audio Features	
HD voice	HD codec
Codec	G.711, G.7231 ,G.729a/b, G.722,G.726
DTMF	In-band, RFC2833 and SIP INFO
Echo cancellation	Support G.168
Software Functions	
VOIP protocol	SIP 2.0
SIP account	One SIP Account
IP Address Announcing	Long press the button for 20 seconds, automatically broadcast
Language	Multi-language
Network protocol	HTTP / FTP / TFTP / IEEE802.1Q
QoS	Voice priority tag(TOS), Dynamic jitter buffer, Voice detection, Comfortable background noise is generated(CNG), Echo elimination G.168(32ms)
Security	Voice signal encryption
Trigger buttons	 Signal Input: to the monitoring server, the corresponding info. Output: the output is optional: key control, switch input control, monitoring and control. Server Message: the default buttons to control as a priority.
Conference Calls	Support up to 12 conference calls

