



## GEN-6392A01

### Network Horn Speaker

GEN-6392A01 is a network horn speaker that supports DC/POE power. It can receive calls from SIP phones or VMS platforms and handle up to 20 multicast streams. The device stores 20 audio files, offers priority management, timers, and custom actions (with HTTP API trigger).

SIP

Onvif

POE

Multicast

## MAIN FEATURES

### ▲ Network Audio Transmission

GEN-6392A01 Supports audio signal transmission over IP networks, enabling real-time audio transmission over long distances. The device is compatible with standard audio communication protocols (such as SIP, ONVIF, and multicast), allowing seamless integration into existing communication systems and flexible adaptation to various audio systems and devices.

### ▲ Rich expansion interfaces

The GEN-6392A01 features a rich set of expansion interfaces, supporting 2 alarm input channels for integrating third-party devices (such as cameras, infrared curtains, smoke detectors, etc.) for linked alarm actions. It also supports 1 alarm output channel, capable of directly driving a siren, and 1 audio input channel, allowing for the expansion of a microphone panel for communication applications.

### ▲ Durability and Flexible Deployment

The GEN-6391A01 features an aluminum alloy casing and stainless steel bracket, with an overall protection rating of IP66, making it suitable for most outdoor environments. It is equipped with industrial-grade chips, supporting stable operation in temperatures ranging from -40°C to 70°C, greatly enhancing the durability of the device.

## SPECIFICATION

### Power Parameters

Power Supply	DC24V PoE (IEEE802.3af/at)
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Power Cons	≤3W
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### Audio & Interface Parameters

Power Output	15W (Use DC 24V) 15W (Use POE/POE+)
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Frequency Response	280~12.5KHz
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Sensitivity	112dB
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Max SPL	121dB
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Audio Trans	Multicast , ×20(Editable)
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Audio Codec	Opus, PCMU (G.711u), PCMA (G.711a), G.722
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Audio Input	×1, Line/Mic input(Balanced,Switchable)
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Input Sensitivity	-10dB(Line) , -20dB(Mic)
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Control Input	×2
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Control Output	×1 ,Relay output/DC12V output switchable
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Built-in Mic	×1,Electret Condenser Mic
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Audio Storage	Max 20 audio files ,80MB
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Audio File Format	WAV, MP3
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### Network Parameters

Network I/F	100BASE-TX ,RJ45 Connector
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Network Protocol	TCP/IP,UDP,HTTP,RTP,RTSP ARP,ICMP,NTP,SIP,ONVIF
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Config Method	Web Interface or DevConfig Tool
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### Mechanical & Environment Properties

Mounting Method	Wall-Mounted or Pole
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Install Env	Outdoor Locations
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Ingress Protection Grade	IP66
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Working Temp	-40°C~+70°C
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Operating Humidity	≤90% RH non-condensing
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Housing Material	Aluminum Alloy
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Color	White
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Dimensions	222*162*281mm
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Weight	1600g
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Warranty	2 Years
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### ACCESSORIES

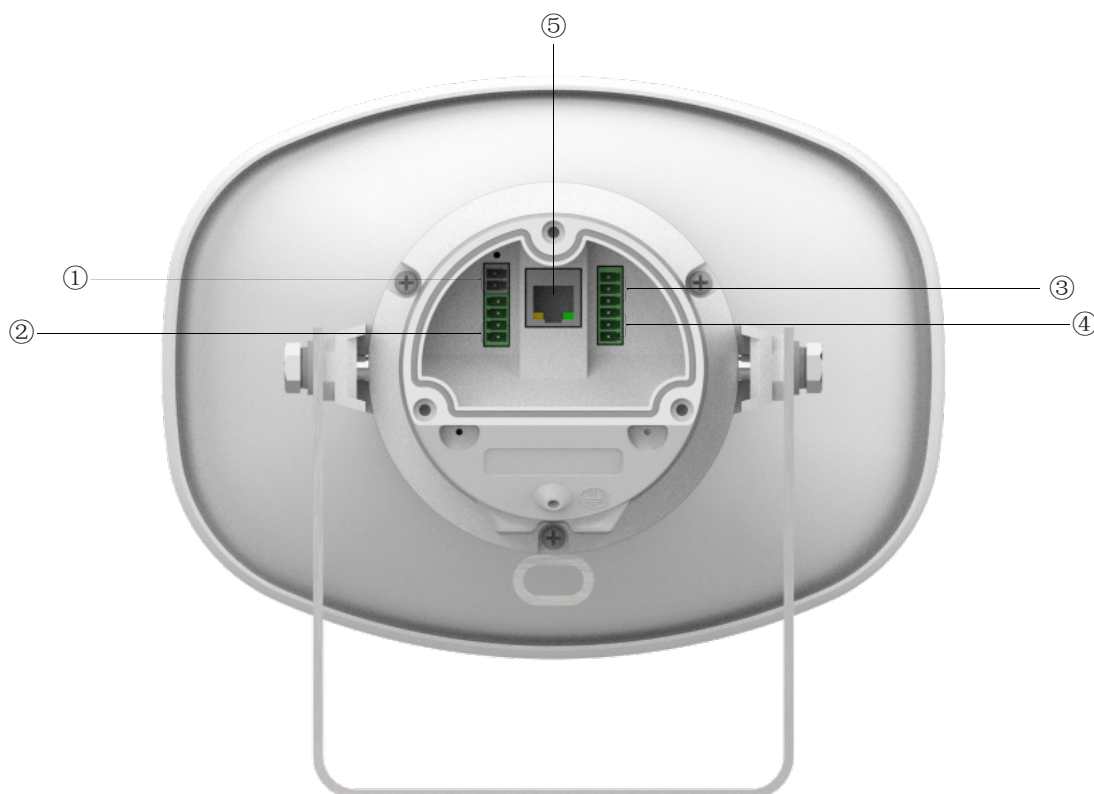
1×2pin Terminal(3.81-2P)
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1×4pin Terminal(3.5-4P)
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1×6pin Terminal(3.5-6P)
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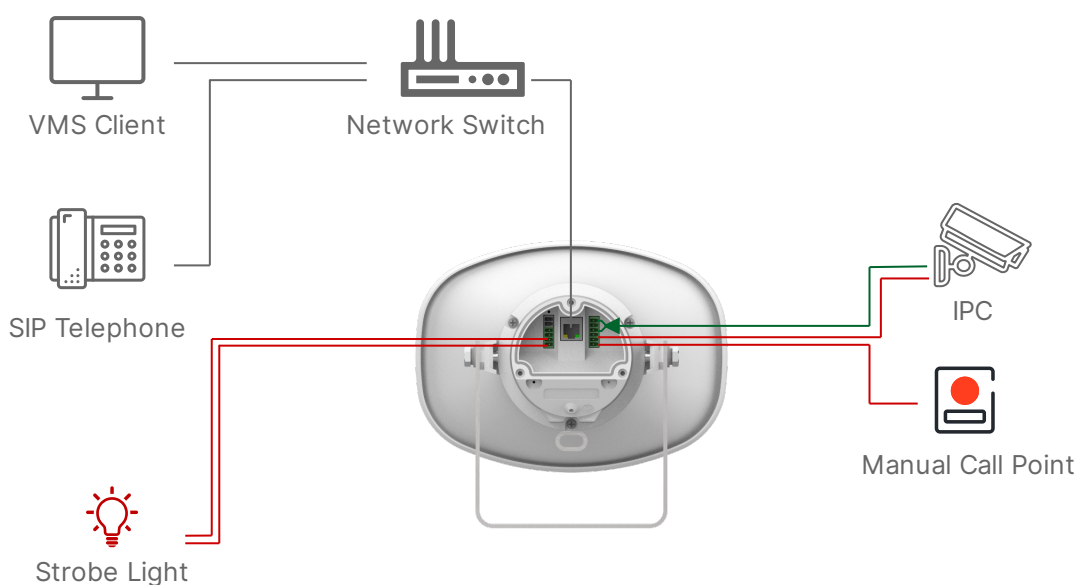
1 × Installation Manual
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## INTERFACE DESCRIPTION



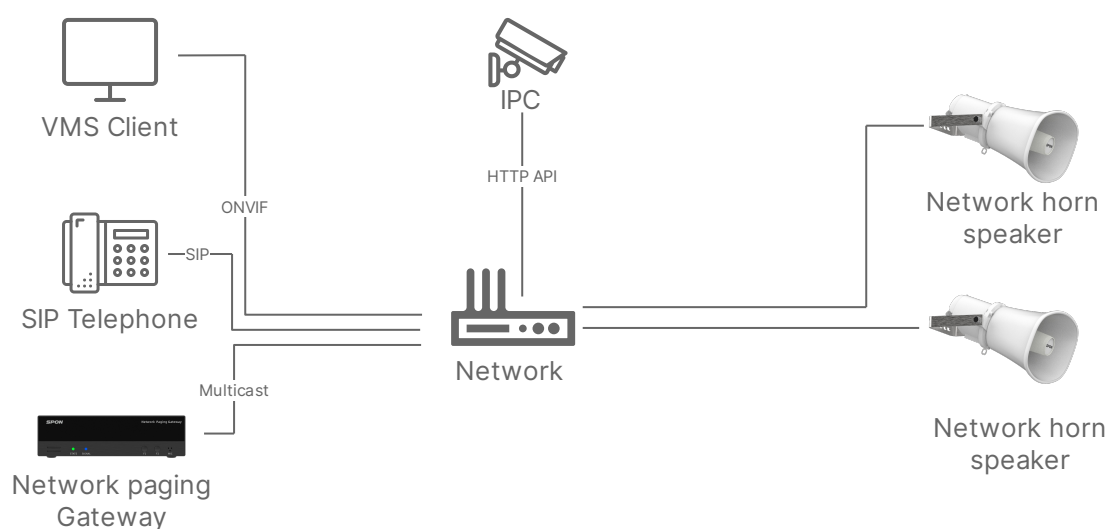
- ① Power Input
- ② Alarm Output Terminal
- ③ Line/Mic Input Terminal
- ④ Alarm Input Terminal
- ⑤ RJ45 Connector

## CONNECTION DIAGRAM



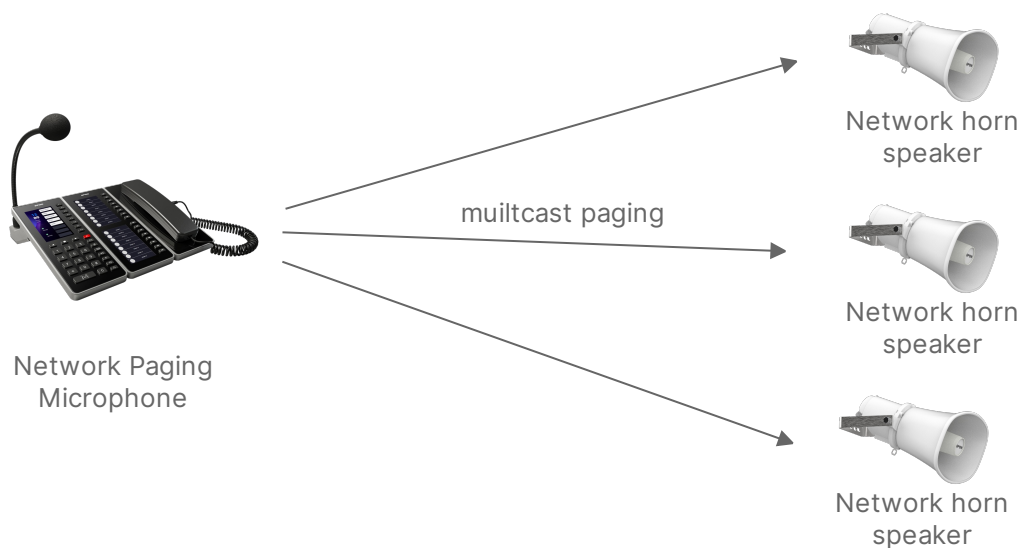
## APPLICATION

The GEN-6392A01 Network Horn Speaker adopts standard industrial audio communication protocols and supports receiving audio streams via SIP, ONVIF, and multicast. The terminal features multiple I/O control interfaces (2 alarm inputs, 1 output) and supports HTTP commands, facilitating integration with third-party systems.

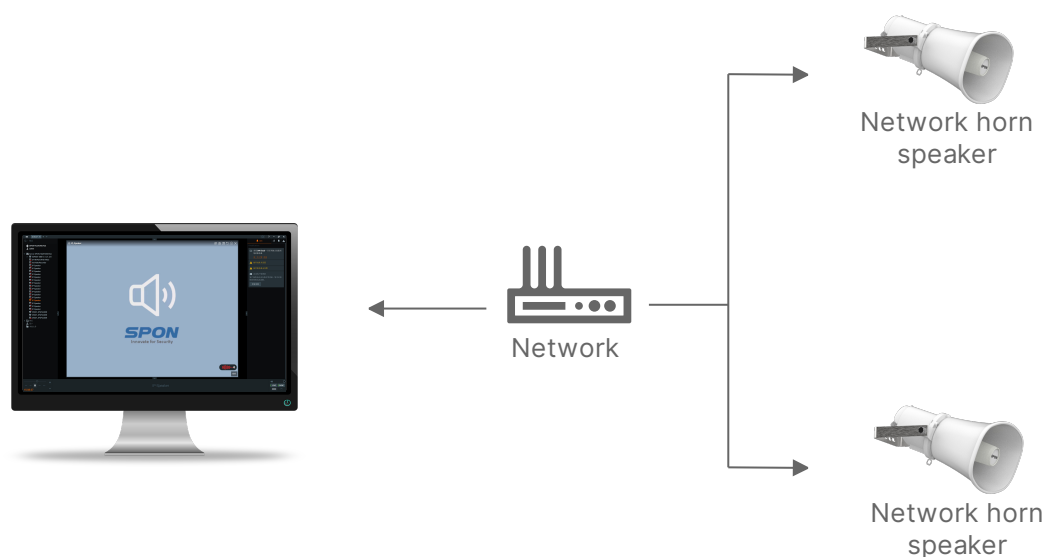


## APPLICATION

The Network Horn Speaker GEN-6392A01 support paging from SIP phone by using multicast address , the multicast address and priority can be set on its webpage .



The Network Horn Speaker GEN-6392A01 support Onvif protocol which can be registered to the 3-rd party monitoring system ,and during emergency situation ,the operator can directly make a live announcement to the speaker on the monitoring platform.



# INSTALLATION

Unit: mm

