

HS-A1 Getting Started Guide

Introduction

The HS-A1 is an analog handset with an armored cable and a hook switch mechanism designed to be connected to an IP7 intercom module to enable full duplex audio operations with a TalkMaster operator console or SIP Phone.

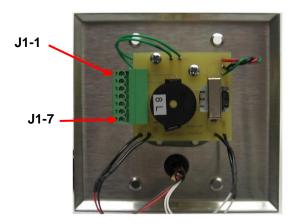
The HS-A1 can place a call by removing the handset from the switch hook. If configured for TalkMaster, it will place an Incoming Call. If the IP7 is configured to dial out to a SIP Extension, it will initiate a call to the specified phone number.

If the HS-A1 is called from TalkMaster or from a SIP phone, the internal piezo buzzer will sound to indicate there is an incoming call. The HS-1A must be connected to either an IP7-SE8 or IP7-FD IP Module.



Installation

The Handsets are designed to be flush mounted in a 2-gang box. A 3-pair cable is used to connect the Handset to an IP7-SE8. On the back of the handset is a seven position 3.81mm DIN connector. The connection details are:



Connector	Signal	IP7 Connections
J1-1	SPKR 8Ω+	Connect to J2-6 - SPKR 8Ω+ on the IP7
J1-2	SPKR 8Ω-	Connect to J2-7 - SPKR 8Ω - on the IP7
J1-3	RESERVED	
J1-4	Hook Switch	Connect to J3-5 – SENSOR on the IP7
J1-5	GND	Connect to J2-4 – GND on the IP7
J1-6	MIC -	Connect to J2-2 - MIC- on the IP7
J1-7	MIC +	Connect to J2-1 - MIC+ on the IP7

For best audio quality, install the Handset as close as possible to the IP7 to reduce the chance of picking up audio noise. Use an 18 AWG 3-pair multi-conductor with a shield. Refer to the IP7 Reference Manual for instructions on connecting the shield and reducing audio noise.

Configuration

TalkMaster FOCUS Software is a suite of Windows® based applications used to configure and manage Digital Acoustics IP Intercoms and Paging endpoints. TalkMaster FOCUS is available for download on the Digital Acoustics Website. The TalkMaster Server/Admin Console must be installed to configure IP Endpoints.

The following procedure is for configuring endpoints as TalkMaster Clients. For SIP Configuration, see the **eSIP Stand-alone Configuration Steps** topic in the Admin Console online help.

To enable Handset support for an IP7:

- Power up the IP7 and connect it to the network
- Open the TalkMaster FOCUS
 Administrator console
- Select the IP Endpoints tab
- Click the **Find All** button. This will discover any new intercoms added to the network
- Select the IP7 used for the Handset by matching the ICOM ID on back of the unit with the ICOM ID displayed in the left-hand pane of the IP Endpoints tab
- Enter the Location Name, IP
 Address information, check
 the Authorized IP Endpoint
 box and enter the Server IP and port

General eSIP ZC Configuration Options 1 Options 2 Options 3		
Туре	Client	
Location Name	Handset	
Assign IP autor	natically with DHCP	
IP Address	10.3.3.157	
Subnet Mask	255.255.255.0	
Gateway	10.3.3.1	
DNS Address	192.168.254.237	
Port	0	
✓ Authorized IP Endpoint Server IP 10.3.3.220 ▼ : 3000		

- Select the **Options1** tab
- Check the Disable Sensor Repeat
 box
- Check the Use UDP/RTP box
- Check the Supports Full Duplex
 box

General eSIP ZC		
Configuration Options 1 Options 2 Options 3		
Fail Forward IP 1 5.0.0.0 : 0		
Fail Forward IP 2 0.0.0.0 : 0		
Fail Forward IP 3 0.0.0.0 : 0		
Enable Automatic Return to primary		
Relay Mode		
Factory Defaults		
Security Set		
Sensor Activate when grounded (Door)		
Input Source Microphone		
☐ Enable ST/ADA Mode		
✓ Use UDP/RTP Fixed RTP Port 0		
Supports Full-Duplex Audio		

- Select the Options 2 tab
- Place a check next to the Handset option
- Under Audio Settings, if paired with an IP7-SE8, check the Disable AGC box and enter 16 for the Microphone Sensitivity
- Under Audio Settings, change the Default Microphone to uLaw
- If paired with an IP7-SE8, check
 Disable AGC and enter 16 for the microphone sensitivity
- If paired with an IP7-FD, press the SET button and select HS-A1 Handset from the list
- Click the **Save** button to save the changes to the Intercom

General eSIP ZC
Configuration Options 1 Options 2 Options 3
🗍 Remote Listen Disable 🗍 Show on Paging Tab
Call Button
🔽 Handset 🗌 Speaker
IP7 Tracking Enabled
Use IGMP v3 instead of IGMP v2
🔲 Multicast Background Music 📃 PTT Sends Audio
225.0.0.254 : 4000 8000 👻
Server-Peer Settings
No Beep Tone After PTT Released
Default to Listen (Direct Mode)
Server-Peer Hands Free Mode
Send Sensor State to Client
Audio Settings
✓ Disable AGC (0-63) 16
Default Microphone
Audio Profile None Set

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