

Manual Type 6.5.0.9-RF

XXX

with Essex "ASCII" Keypad

For Models -

400-A-VIO	401-A-VIO
403-A-VIO	404-A-VIO
410-A-VIO	411-A-VIO

Specifications

Communication

Holovision VIO is a SIP enabled intercom module developed specifically with the custom integrator in mind. It works with Crestron RAVA/SIP and most SIP enabled VoIP phone systems and servers.

Access Control

Essex K1-26I Keypad field configurable to ASCII, Wiegand and others.



400 Series "A" Junction Board

INSTALLATION INSTRUCTIONS

501 E. Goetz Ave. Santa Ana, Ca 92707

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HOLOVISION

ASCII Connection to Crestron or HAI 400





VIO Wiring Diagram



- 1. Plug the device into the network. Device will speak it's IP address.
- 2. Navigate to device IP on a web browser.
- 3. Under "Configuration" please enter SIP ID for the name to be displayed, and enter the extension of the Main RAVA Touchscreenunder "Input Call ID 0 Default is CALL:CRESTRON"

HOME PROFILES	CONFIGURATION M CLIENT V2.12	status 2D(08 Apr 2013)	DEFAULTS	
SIP Door Station	BASIC SETTINGS			
Basic Settings Advanced Settings	SIP PROTOCOL SETTINGS Peer to Peer	🔍 No 🖲 Yes		
Apply Cancel	SIP ID (username)	VIO		
	OUTBOUND CALL SETTINGS			
	Call on Device Inputs Input 0 Call ID			

*Note out of the box the "CALL:CRESTRON" default will ring all touchpanels in the CRESTRON call group.

4. Configure the Audio settings. Note:RAVA Operation has been designated to "Push to Talk" (PTT) from Crestron and only this operation will be supported. Although full duplex and half duplex are an option, these settingdo not perform best in all senarios. As a recommendation you can experiment for the best audio settings.

AUDIO SETTING BELOW ARE ONLY A RECOMMENDATION

RAVA-SIP	CONFIGURATION ST	atus defaults upda ⁻ 2013)
SIP Door Station	AUDIO SETTINGS	
Basic Settings	Input Source	🗢 Line 🖲 Mic
Advanced Settings	Encoding	uLaw / 8 kHz (G.711) 💌
Network	Volume	50 💌 %
SIP Protocol	volume	30 . //
Outbound Calls	Microphone Gain	30 💌 dB
Inbound Calls	A/D Amplifier Gain	0 🔻 dB
Audio	Acoustic Echo Cancellation	Off On
Control Interfaces		
Security	Talk Mode	FDX In HDX
Apply Cancel	Output Trigger Level	1000
	Trigger Level Timeout	200 ms

- Configure the relay setting under "Inbound Calls". Configure the powered relay and the dry contact closure. Choose any 2 to 4 digit combination
 Note; By configuring the request to exit feature you can configure for either relay.
- 6. Once the device has been configured you can initiate the call by pressing the button

SIP Door Station	INBOUND CALLS		
Basic Settings	Input Buffer Level	100	ms
Advanced Settings Network	Stream Timeout	0	minutes
SIP Protocol	Beep on Call Answer	Off	On
Outbound Calls	Power Door Relay Control		
Inbound Calls	Door Open Code	11	
Audio Security	Open Door Relay for	5 🔻 sec	conds
	Relay Number to Enable at Call Answer	disabled	-
Apply Cancel	Switched Door Relay Control		
	Door Open Code	B£ALL	
	Open Door Relay for	1 💌 sec	conds
	Relay Number to Enable at Call Answer	disabled	•
	Request To Exit Call ID 2		•

VIO(sip) QUICK START

- 1. Plug the device into network. Device will speak its IP.
- 2. Navigate to device IP on web browser.
- 3. Under Configuration please enter the required SIP settings

HOME	CONFIGURATION	STATUS	DEFAULTS	UPDATE	REBOOT
SIP CLIENT					
SIP Door Station	BASIC SETTINGS				
Basic Settings Advanced Settings	SIP PROTOCOL SETTINGS Peer to Peer	🖲 No 🔘 Yes			
Apply Cancel	SIP Server (PBX)	192.168.1.334			
	SIP ID (username)	VIO Front Door			
	SIP Password (secret)	•••••			
	SIP Display Name	VIO Front Door			
	OUTBOUND CALL SETTINGS				
	Call on Device Inputs				
	Input 0 Call ID	100			
	Input 1 Call ID				
	Input 2 Call ID				
	INBOUND CALLS				
	Phone pickup mode	autoanswer 💌			
	Pick/hang up time	20 💌 seconds			

4. Configure the audio settings.. Although full duplex and half duplex are an option, these settings do not perform best in all scenarios. As a recommendation you can experiment for the best audio settings. AUDIO SETTINGS BELOW ARE ONLY A RECOMENDATION

HOME	CONFIGURATION	STATUS	DEFAULTS	UPDATE	REBOOT
SIP CLIENT					
SIP Door Station	AUDIO SETTINGS				
Basic Settings	Encoding	uLaw / 8 kHz (G.	711) 💌		
Advanced Settings	Volume	50 💌 %			
SIP Protocol	Microphone Gain	21 💌 dB			
Outbound Calls	A/D Amplifier Gain	0 💌 dB			
Inbound Calls Audio	Acoustic Echo Cancellation	© Off ◉ On			
Security	Talk Mode	◎ FDX			
Apply Cancel	Output Trigger Level	1000			
	Trigger Level Timeout	200 ms			
	Al Phone Support	🔘 Yes 🔍 No			

5. Configure the relay settings under inbound calls. Configure the powered relay and also the dry contact closure; chose any 2 to 4 digit code.

Note: By configuring the request to exit feature. You can trigger either relay 1 or 2 by doing so.

6. Once the device has been configured you can initiate the call by pressing on the button.

SIP Door Station	INBOUND CALLS		
Basic Settings	Input Buffer Level	100	ms
Advanced Settings	Stream Timeout	0	minutes
SIP Protocol	Beep on Call Answer	◉ Off ©	On
Outbound Calls	Power Door Relay Control		
Inbound Calls	Door Open Code	11	
Audio Security	Open Door Relay for	5 💌 see	conds
	Relay Number to Enable at Call Answer	disabled	-
Apply Cancel	Switched Door Relay Control		
	Door Open Code	22	
	Open Door Relay for	1 💌 see	conds
	Relay Number to Enable at Call Answer	disabled ·	•
	Request To Exit Call ID 2		·



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