

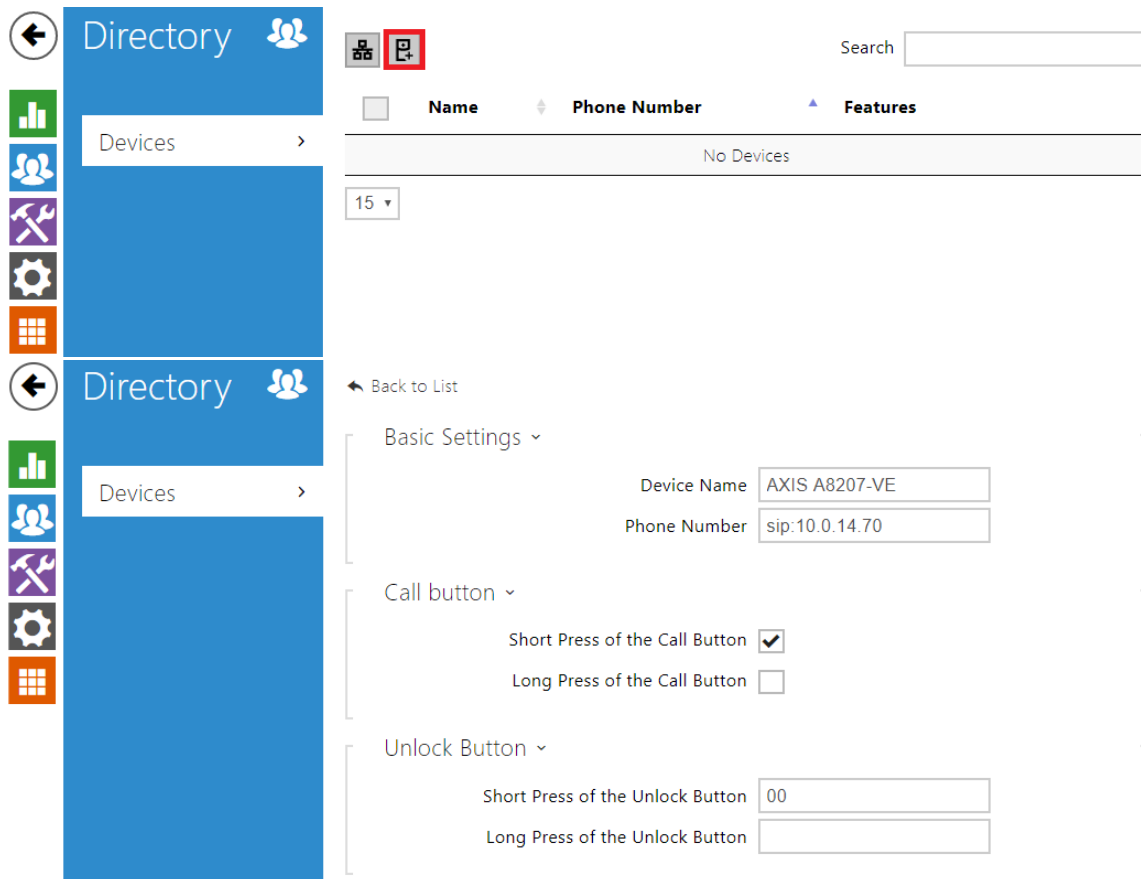
AXIS A8207-VE – Peer-to-peer call to 2N® Indoor Talk

2N® Indoor Talk is able to receive calls from the AXIS door station A8207-VE.

- Requirements**
- 2N® Indoor Talk with firmware version 2.26.0 and later.
 - Tested with AXIS A8207-VE with firmware version 18.
 - IP address of AXIS A8207-VE: 10.0.14.70. IP address of 2N® Indoor Talk: 10.0.14.31. This is only an example, please change it according to your network requirements.

Configuration of 2N® Indoor Talk

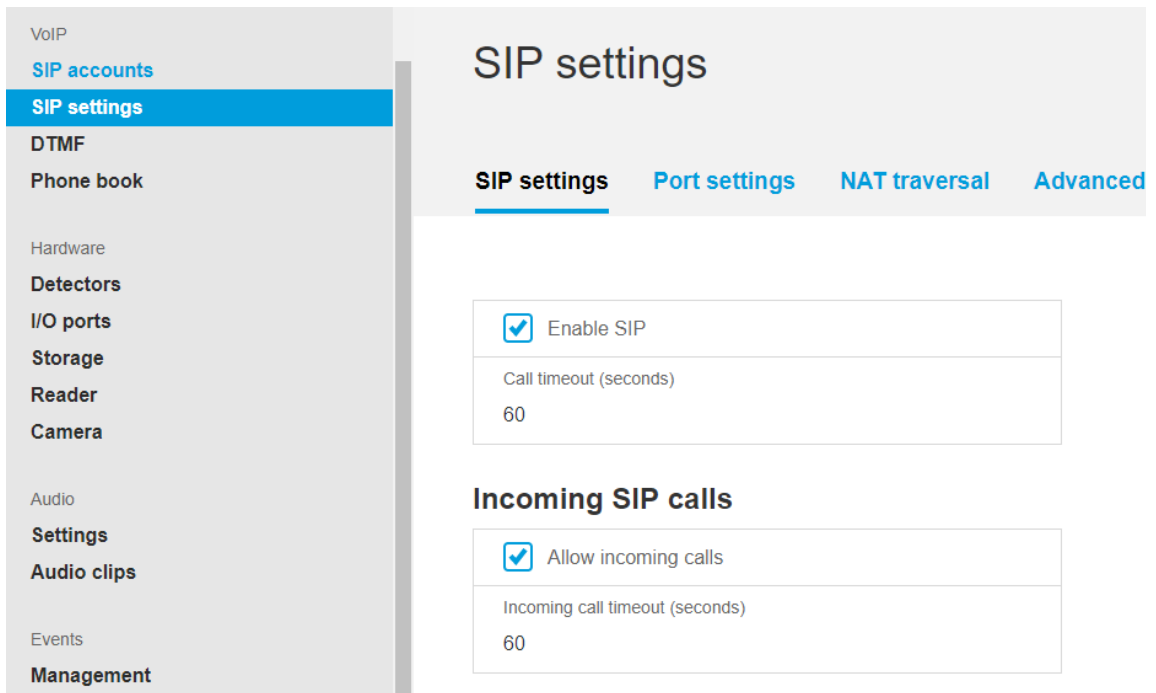
- Go to web management - Directory - Devices and add AXIS A8207-VE's SIP address and unlock DTMF code to be able to call from 2N® Indoor Talk (by the green button) to AXIS A8207-VE.



The screenshot shows the web management interface for 2N Indoor Talk. The left sidebar contains navigation icons for various functions. The main content area is titled 'Directory' and shows a 'Devices' section. A table with columns 'Name', 'Phone Number', and 'Features' is visible, but it is empty, displaying 'No Devices'. Below the table, there is a 'Back to List' link and three expandable configuration sections: 'Basic Settings', 'Call button', and 'Unlock Button'. The 'Basic Settings' section includes fields for 'Device Name' (AXIS A8207-VE) and 'Phone Number' (sip:10.0.14.70). The 'Call button' section has checkboxes for 'Short Press of the Call Button' (checked) and 'Long Press of the Call Button' (unchecked). The 'Unlock Button' section has input fields for 'Short Press of the Unlock Button' (00) and 'Long Press of the Call Button'.

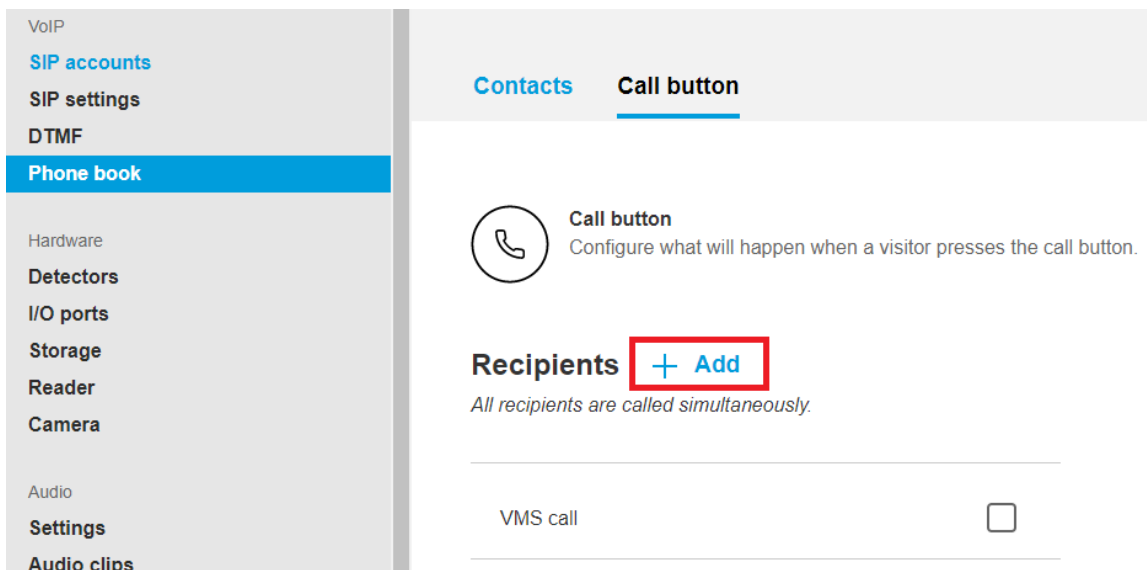
Configuration of AXIS A8207-VE

- Go to web management - VoIP - SIP Settings and enable SIP and incoming SIP calls.



The screenshot shows the 'SIP settings' page in a web management interface. On the left is a navigation menu with categories: VoIP, SIP accounts, SIP settings (highlighted), DTMF, Phone book, Hardware, Detectors, I/O ports, Storage, Reader, Camera, Audio, Settings, Audio clips, Events, and Management. The main content area is titled 'SIP settings' and has sub-tabs: SIP settings (active), Port settings, NAT traversal, and Advanced. Below the sub-tabs are two configuration sections. The first section has a checked checkbox for 'Enable SIP' and a 'Call timeout (seconds)' field with the value '60'. The second section is titled 'Incoming SIP calls' and has a checked checkbox for 'Allow incoming calls' and an 'Incoming call timeout (seconds)' field with the value '60'.

- Go to web management - VoIP - Phone book - Call button and add 2N® Indoor Talk´s SIP address as a recipient of the call button press.



The screenshot shows the 'Call button' configuration page. The left navigation menu is similar to the previous screenshot, but 'Phone book' is highlighted. The main content area has sub-tabs: Contacts and Call button (active). Below the sub-tabs is a 'Call button' section with a telephone icon and the text 'Configure what will happen when a visitor presses the call button.' Below this is a 'Recipients' section with a '+ Add' button highlighted in a red box. A note below says 'All recipients are called simultaneously.' Below the note is a table with one row: 'VMS call' and an unchecked checkbox.

Recipients + Add

All recipients are called simultaneously.

Contact

SIP address

SIP address
sip:10.0.14.31

Call from SIP account
sip:10.0.14.70:5060

Test

Add **Cancel**

- Go to web management - VoIP - DTMF and add DTMF Sequence for unlocking the door by 2N® Indoor Talk.

VoIP


- SIP accounts
- SIP settings
- DTMF**
- Phone book

Hardware

- Detectors
- I/O ports
- Storage
- Reader
- Camera

DTMF

DTMF Configuration for SIP Accounts

Peer-to-peer accounts (peer-to-peer) 

DTMF using SIP INFO (RFC2976)

DTMF using RTP (RFC2833)

Associated DTMF Sequences

Name	Sequence
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Modify DTMF Configuration ?

SIP Account: Peer-to-peer accounts

DTMF using SIP INFO (RFC2976)

DTMF using RTP (RFC2833)

DTMF Sequences

Name	Sequence
<input checked="" type="checkbox"/> Unlock	00

Add...

Modify...

Remove

- Go to web management - Events - Management - Action Rules and create a new Action Rule to open the door when DTMF sequence is received during call.

Reader

Camera

Audio

Settings

Audio clips

Events

Management

Network

Certificates

Security

Settings

Action rules
Recipients
Schedules
Recurrences
Manual trigger

Name	Trigger	Schedule	Action	Recipient
<input checked="" type="checkbox"/> Call button rule	Input Signal - Virtual Inputs	-	Make Call	New Recip ▼

Add...

Copy...

Modify...

Remove

Action Rule Setup

General

Enable rule

Name:

Condition

Trigger: Start condition only

Schedule:

Additional conditions

Wait at least before re-running the rule (max 23:59:59)

Actions

Type:

Port:

Set state: Active Inactive

Duration
 Go to opposite state when the rule is no longer active
 Go to opposite state after