



Configuration Guide for 2N IP intercoms



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Revision History

Version	Date	Author	Change
1.0	18.10.2022	Anna Bezverkhnia	Config Guide version 1.0 created
1.1	16.11.2022	Jakub Kuba	Revisions of version 1.0
1.2	4.12.2022	Jakub Kuba	Function description and revision
1.3	8.12.2022	Jakub Kuba	Update based on Zoom feedback

Supported 2N IP Intercom Versions

Manufacturer	Device	Firmware Version
2N TELEKOMUNIKACE a.s.	2N IP Intercom range	2.35+

2N IP Intercoms share the same SIP Stack, ensuring compatibility across the product range in FW version 2.35 and higher. Supported models:

- 2N® IP Style
- 2N® IP Verso
- 2N® IP Solo
- 2N® IP Force
- 2N® IP Safety
- 2N® IP Base
- 2N® IP Uni
- 2N® IP Audio Kit

Supported Features

Features supported during calls 2N Intercoms with Zoom

- Audio calls from 2N Intercom to Zoom (desktop, mobile)
- Calls from Zoom (desktop, mobile) to 2N Intercom
- DTMF Switch control (Zoom desktop, mobile)
- Call transfer (Zoom desktop, mobile)
- Call conference (Zoom desktop, mobile)
- Call recording (Zoom desktop)

1.

Overview

This document provides instructions on how to configure and add your 2N IP intercoms to the Zoom Web Portal.

2N IP intercom can register to Zoom Phone as a third-party SIP Endpoint and provide Telephony features and capability such as calling and other phone system/PBX functions.

2N IP intercoms can be connected via Zoom Phone subscription, which is a Zoom licensed feature.

2. Configuration Steps - Zoom Web Portal

This section provides instructions on how to configure SIP Endpoint in Zoom Web Portal (does not include steps for configuring Vendor Endpoint specifically).

Prerequisites:

- A subscription to Zoom Phone is required to provision a 2N device. To obtain a subscription please visit [Zoom Plans & Pricing](#) and contact Zoom sales.
- Zoom approval is required for provisioning of 2N devices, account administrator should contact Zoom representative.

To register a SIP Endpoint to Zoom Phone you can use either a User or a Common Area to which you will assign a new device.

In order to set up a call between multiple devices, it is best to set up users and common areas respectively, depending on intended use of the Zoom Phone service in combination with 2N Intercoms.

2.1 User Creation

When using Zoom account, you are starting with your user account as the only user linked to the account. However, you can invite multiple users to be joined under your plan and allow them to receive calls via Zoom from 2N devices. To create a new user, simply follow these steps:

1. **Sign in** to the Zoom Web Portal
2. Click **User management > Users**
3. Click the **+Add Users** button

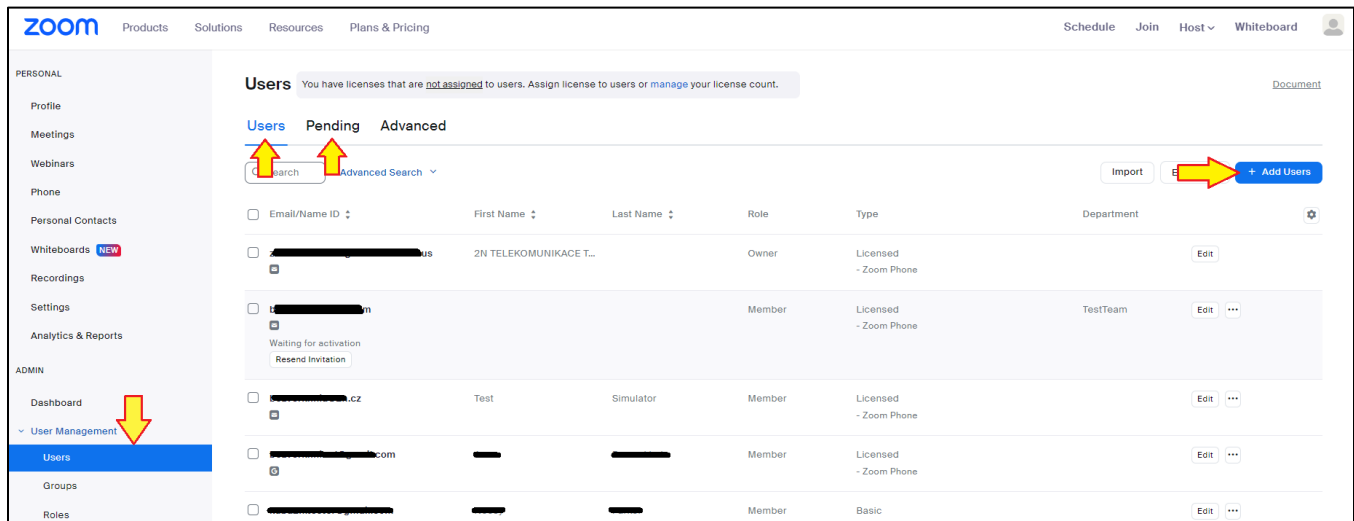


Figure 1 - Add a new user to your Zoom account

4. Click **Add** and enter the following information
 - a. **Add users with their e-mail addresses**
 - i. List the e-mail address of a user you wish to add to your account
 - ii. Additional information about the user can be added now or edited later

Add Users

Add users with their email addresses
You can add users of all types to your account. If you enter the email address of account owners, all users on their accounts will be added to this account.

Use comma to separate multiple email addresses.
Enter e-mail

User Type [?] Basic Licensed On-Prem [?]

Meeting (4 available) Large Meeting (500 participants) (10 available) Zoom Webinars (500 attendees) (10 available)

Department
Manager
Job Title
Location

Add Cancel

Figure 2 - Enter e-mail(s) of invited user(s) and optionally fill in their additional details

5. Click **Add**
6. Invited users will be listed in the **Pending** tab until they accept the invitation

2.2 Common area creation

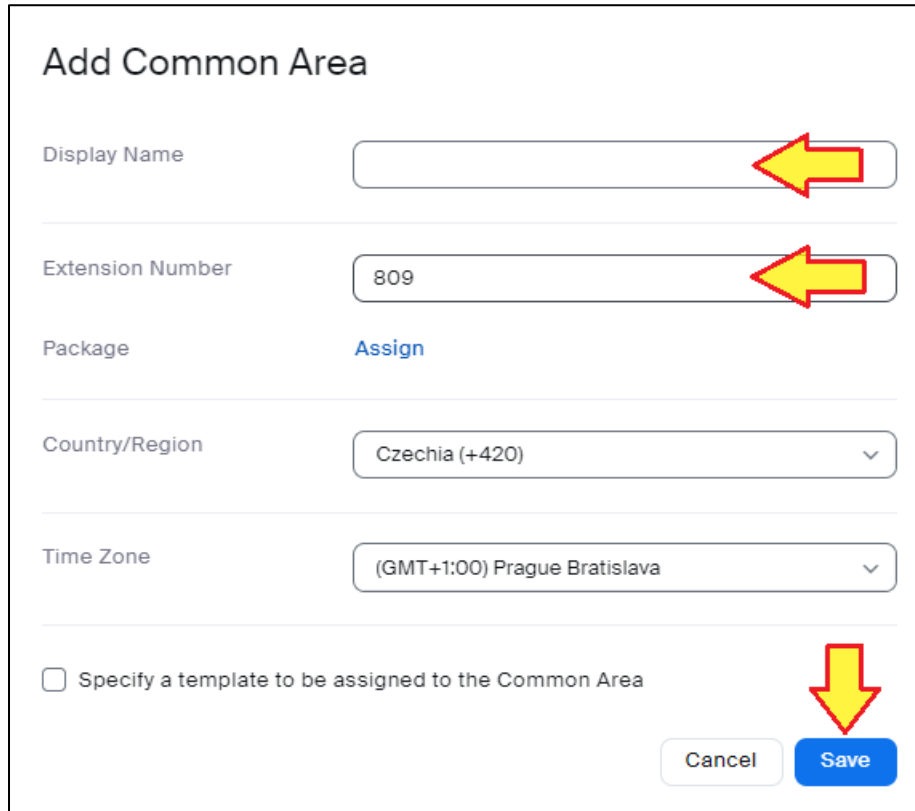
You may wish to set up an account to be associated with a 2N Device that is publicly accessible as opposed to a dedicated user account. In that case, follow these steps to create a common area. The credentials created here can be used to set up a 2N Device and link it to other created accounts for calling purposes. You can also visit the [Zoom support site](#) for more information about managing Common Area Phones.

1. **Sign in** to the Zoom Web Portal
2. Click **Phone system management** > Users & Rooms
3. Click the **Common Area** tab

The screenshot shows the Zoom Web Portal interface. The top navigation bar includes 'zoom', 'Products', 'Solutions', 'Resources', 'Plans & Pricing', 'Schedule', 'Join', 'Host', and 'Whiteboard'. The left sidebar is divided into 'PERSONAL' and 'ADMIN' sections. The 'ADMIN' section is expanded to show 'Phone System Management', which is further expanded to 'Users & Rooms'. The 'Common Areas' tab is selected in the top navigation. A message box states: 'Analog Telephone Adapter and Shared Phone management functions are now migrated to "Common Area" and "Phones & Devices" sections. Common Area section is used to create a profile/extension and then IP Phones, Analog Telephone Adapters, or Zoom Phone appliance devices can be added to the Common Area. This is a consistent workflow with managing Zoom Phone users. Note: Zoom Phone Appliance devices can be assigned to Common Area. To manage Zoom Phone Appliance, please go to Device Management'. Below the message is a 'Plan Details' section with an 'Add' button and an 'Export' button. A search bar is present with the text 'Search by Name, Ext. or Number'. A table lists common areas with columns for 'Display Name', 'Ext.', 'Package(s)', 'Number(s)', and 'Desk Phone(s)'. The table contains two rows: 'MainEntry_Test' with extension 100 and 'Common area' with extension 807. Both are assigned to 'Global Select' and have 'Online' and 'Offline' status indicators. The bottom of the table shows 'Page Size 50' and 'Total 2'. A red arrow points to the 'Common Areas' tab, and another red arrow points to the 'Add' button.

Figure 3 - Common area creation page

4. Click **Add** and enter the following information
 - a. **Display Name:** Enter a display name to identify the area
 - b. **Extension Number:** Enter an extension number to assign it to the device
 - c. **Package:** Select a Zoom Phone Package
 - d. **Country/Region:** Select the country and country code
 - e. **Time Zone:** Select the time zone and the city



The screenshot shows a form titled "Add Common Area" with the following fields and values:

- Display Name:** An empty text input field with a yellow arrow pointing to it from the right.
- Extension Number:** A text input field containing the value "809" with a yellow arrow pointing to it from the right.
- Package:** A dropdown menu with the selected option "Assign".
- Country/Region:** A dropdown menu with the selected option "Czechia (+420)".
- Time Zone:** A dropdown menu with the selected option "(GMT+1:00) Prague Bratislava".
- Specify a template to be assigned to the Common Area:** An unchecked checkbox.
- Buttons:** "Cancel" and "Save" buttons at the bottom right. A yellow arrow points down to the "Save" button.

Figure 4 - Common area creation detail

5. Click **Save**

3. Configuration Steps – 2N IP intercom

This section provides instructions on how to configure 2N IP Intercom's SIP Endpoint (does not include steps for configuration in Zoom Web Portal – refer to [2. Configuration Steps - Zoom Web Portal](#) for Zoom portal settings)

3.1 Getting SIP credentials

To register a 2N IP intercom you need to have a User or a Common area, which will be assigned to the intercom. After creating the User or Common area, follow the instructions below.

1. Sign in to the Zoom Web Portal
2. Click **Phone system management** > Phones & Devices

The screenshot shows the Zoom Admin console interface. The left sidebar contains the navigation menu with 'Phone System Management' and 'Phones & Devices' highlighted by red boxes and yellow arrows. The main content area displays a table of devices with columns for Display Name, Device Type, MAC Address, Assigned to, Status, Firmware Version, IP Address, Hot Desking (Signed In), Provision Template, and Call Control from Desktop Client. The table lists four devices: a Desk Phone, a 2N IP Style, a Simulator, and a VoIP device.

Display Name	Device Type	MAC Address	Assigned to	Status	Firmware Version	IP Address	Hot Desking (Signed In)	Provision Template	Call Control from Desktop Client
Desk Phone	Other	7c-1e-b3-12-3d-53	MainEntry_Test Ext. 100	Offline	--	--	Unsupported	Unsupported	Unsupported
2N IP Style	Other	7c-1e-b3-05-49-71	Common area Ext. 807	Offline	--	90.182.112.139	Unsupported	Unsupported	Unsupported
Simulator	Other	fc-1e-b3-00-00-95	Test Simulator Ext. 805	Offline	--	90.182.112.139	Unsupported	Unsupported	Unsupported
VoIP	Other	00-a8-59-1b-18-de	2N TELEKOMUNIKA CE Test Account Ext. 800	Offline	--	10.27.51.4	Unsupported	Unsupported	Unsupported

Figure 5 - Navigate to Phones and Devices

The screenshot shows the Zoom Admin console interface for adding a new device. The left sidebar contains the navigation menu with 'Phones & Devices' highlighted. The main content area displays the 'Assigned' tab with a search bar and filters. A red box and yellow arrow highlight the 'Add' button. Below the search bar is a table of devices with columns for Display Name, Device Type, MAC Address, Assigned to, Status, Firmware Version, IP Address, Hot Desking (Signed In), Provision Template, and Call Control from Desktop Client. The table lists two devices: a Desk Phone and a 2N IP Style.

Display Name	Device Type	MAC Address	Assigned to	Status	Firmware Version	IP Address	Hot Desking (Signed In)	Provision Template	Call Control from Desktop Client
Desk Phone	Other	7c-1e-b3-12-3d-53	MainEntry_Test Ext. 100	Offline	--	--	Unsupported	Unsupported	Unsupported
2N IP Style	Other	7c-1e-b3-05-49-71	Common area Ext. 807	Offline	--	90.182.112.139	Unsupported	Unsupported	Unsupported

Figure 6 - Add new device

3. Click **Add** and enter the following information:

The screenshot shows a form titled "Add Device" with the following fields and controls:

- Display Name:** A text input field with a red box and a yellow arrow pointing to it.
- Description (Optional):** A larger text input field.
- MAC Address:** A text input field with a red box and a yellow arrow pointing to it.
- Device Type:** A dropdown menu with "Select Brand" selected, highlighted with a red box and a yellow arrow.
- Assigned to:** A dropdown menu with "Assign" selected, highlighted with a red box and a yellow arrow.
- Buttons:** "Save" (blue) and "Cancel" (grey) buttons at the bottom right, with a yellow arrow pointing to the "Save" button.

Figure 7 - Enter device information to create new device

- a. **Display Name:** Enter a display name to identify the device
 - b. **Description(Optional):** Enter the description for the device
 - c. **MAC Address:** enter the 12-digit device MAC address.
 - d. **Device type: Select Other.**
 - e. **Assigned To:** Select a User or a Common area to be assigned to this device
4. Click **Save**
 5. Return to device overview and click **Provision** to view the SIP credentials. You will need this information to complete the provisioning on the device side. When device is created, but not assigned to user or area, it can be found in the unassigned tab.

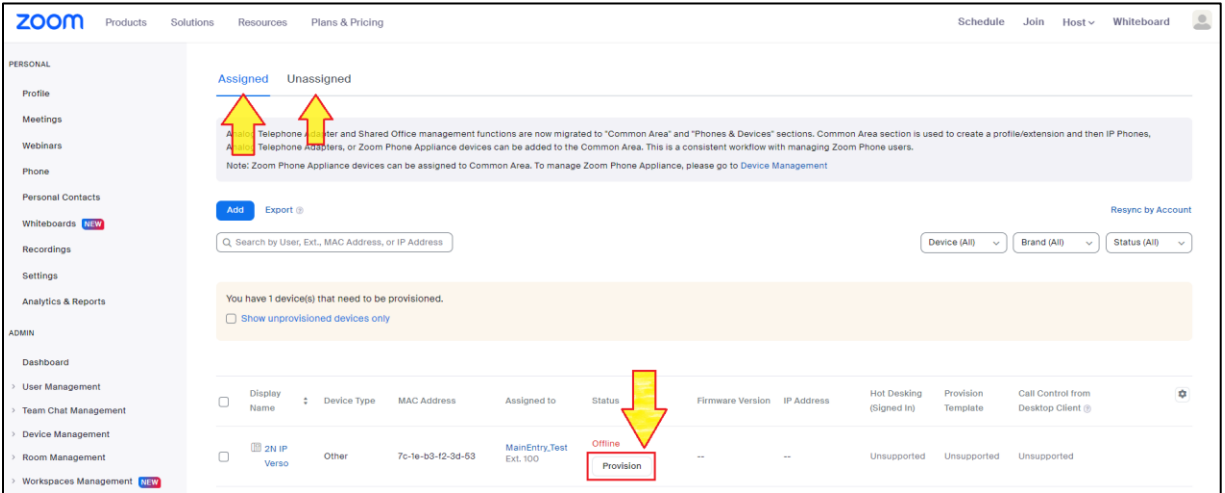


Figure 8 - Device provision information

6. Provisioning overview of the device contains the necessary SIP credentials which will be added to your 2N Intercom in part [3.2 Setting up the intercom](#) as well as 3 certificates which will need to be uploaded to your 2N device.

Provisioning

MAC Address 7c-1e-b3-03-ee-bb

Device Type Other

You will need to enable TLS1.2 for SIP registration and enable SRTP for secure calling on your IP phone. Please refer to your manufacturer's instructions for these processes.

You'll need following information for manual provisioning.

SIP Account 1:

1. SIP Domain: 10000330.zoom.us
2. Outbound Proxy: gossip0h.sc.zoom.us:5091
3. User Name: 4406[REDACTED]33
4. Authorization ID: 849687841467
5. Password: 78[REDACTED].JIV

↓ 3 certificates

Please download [DigiCert Global Root CA](#), [DigiCert Global Root G2](#), [DigiCert Global Root G3](#) and import to your IP phone if they are not in the trust list of the device.

Note: Please note that Zoom support team will not be able to troubleshoot or configure IP phones that are provisioned in this manner. Some Zoom Phone features may not work on manually provisioned phones. It may vary depending on your desk phone model.

Figure 9 - SIP Provisioning information and location of certificates

3.2 Setting up the intercom

To use intercom calling features firstly download the certificates from the Provision part and upload it to the device. You can find the guide where to download the certificates in part [3.1 Getting SIP credentials](#).

Use the instructions below to register your device to Zoom Phone.

1. Sign in on intercom's website



Figure 10 - IP Intercom Homepage - Services

2. Click **Services** > Phone
3. **Choose one of the two SIP accounts** (Figure 11 – SIP account settings) and enter the provision credentials, which have been obtained in [3.1 Getting SIP credentials](#).
 - a. if the SIP account is not enabled, use checkbox to enable the selected SIP account
 - b. **Display Name:** Enter a display name to identify the device
 - c. **Phone Number:** Enter User Name from provision details on Zoom assigned device
 - d. **Domain:** Enter SIP Domain
 - e. **Authentication ID:** Enter Authorization ID
 - f. **Password:** Enter Password
 - g. **Proxy address:** Enter Outbound Proxy without port
 - h. **Proxy Port:** Enter 5091
 - i. **Registrar Address:** Enter Outbound Proxy address

4. Click on Advanced Settings:
 - a. Sip Transport Protocol: TLS
 - b. Lowest Allowed TLS Version: TLS 1.2

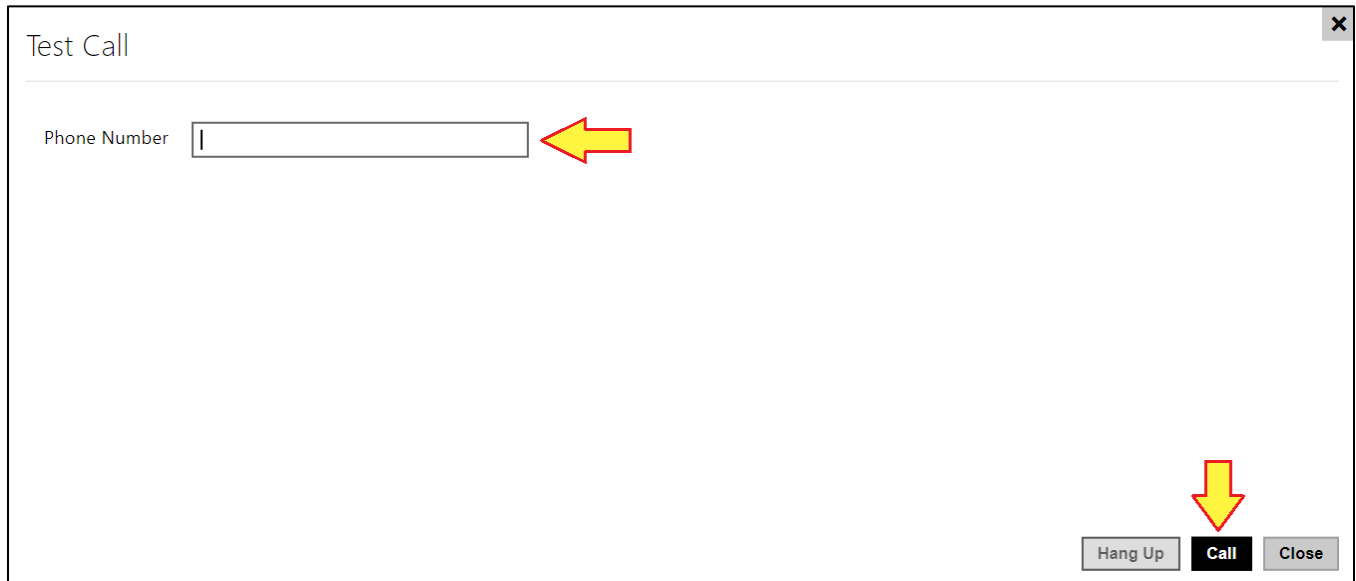
The screenshot displays the Zoom SIP account settings for 'SIP 1'. The interface includes a left-hand navigation menu with options like Phone, Access Control, Streaming, E-Mail, Automation, HTTP API, User Sounds, Web Server, Audio Test, and SNMP. The main content area is divided into several sections:

- Device Identity:** Contains fields for Display Name (2N IP Verso), Phone Number (ID) (44069394384293076533), and Domain (10000330.zoom.us). A 'Test Call' button is highlighted with a red box and an arrow.
- Authentication:** Includes a checked 'Use Authentication ID' checkbox, an Authentication ID (849687841467), and a Password field.
- SIP Proxy:** Features fields for Proxy Address (gosp0h.sc.zoom.us), Proxy Port (5091), Backup Proxy Address, and Backup Proxy Port (Default).
- SIP Registrar:** Contains a checked 'Registration Enabled' checkbox, Registrar Address (gosp0h.sc.zoom.us), Registrar Port (5091), Backup Registrar Address, Backup Registrar Port (Default), and Registration Expiry (120 [s]). The 'Registration State' is highlighted with a red box and labeled 'REGISTERED'.
- Advanced Settings:** Includes a dropdown for 'SIP Transport Protocol' (set to TLS), a dropdown for 'Lowest Allowed TLS Version' (set to TLS 1.2), and a 'Verify Server Certificate' checkbox.

Figure 11 - SIP account settings

5. Save settings
6. In case of successful registration, the registration State switches to **REGISTERED**.

2N IP Intercoms allow users to create a test call from both SIP accounts with test call button marked in screen capture image in [3.2 Setting up the intercom](#) step 4. After you have successfully set up the SIP account and the state has changed to **REGISTERED**, you can use the feature.



The screenshot shows a window titled "Test Call" with a close button in the top right corner. Below the title bar, there is a "Phone Number" label and an empty text input field. A yellow arrow points to the input field. At the bottom right of the window, there are three buttons: "Hang Up", "Call", and "Close". A yellow arrow points to the "Call" button.

Figure 12 - Enter phone number and click call

Test call: Click on Test call button in configured SIP account section. Enter another **Zoom User** or **Common area extension number** and click on **Call**.

3.3 Setting up a user on your 2N Intercom and assigning the user to a call button

To call a Zoom account from your intercom, you must set up one or more users on the 2N IP Intercom and assign it to call buttons on the device.

1. Log in to your intercom
2. Navigate to **Directory** > **Users** and create a new user

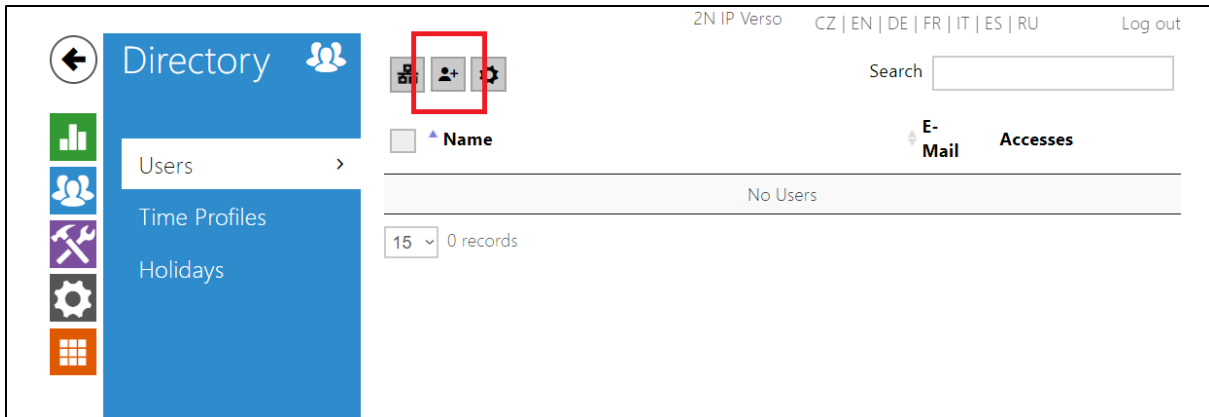


Figure 13 - Click button to create a new user

3. Enter a name of the user and a phone number
 - a. If you only have one SIP account active, the phone number is sufficient. If you have multiple SIP accounts active, make sure to set up the number followed with /1 or /2 to define which SIP account is to be used with the phone number

Figure 14 - Create user and assign Name and Phone Number

4. **Save** the new user
5. Navigate to **Hardware > Buttons** section on the intercom and add the newly created user to desired available button on your 2N IP intercom

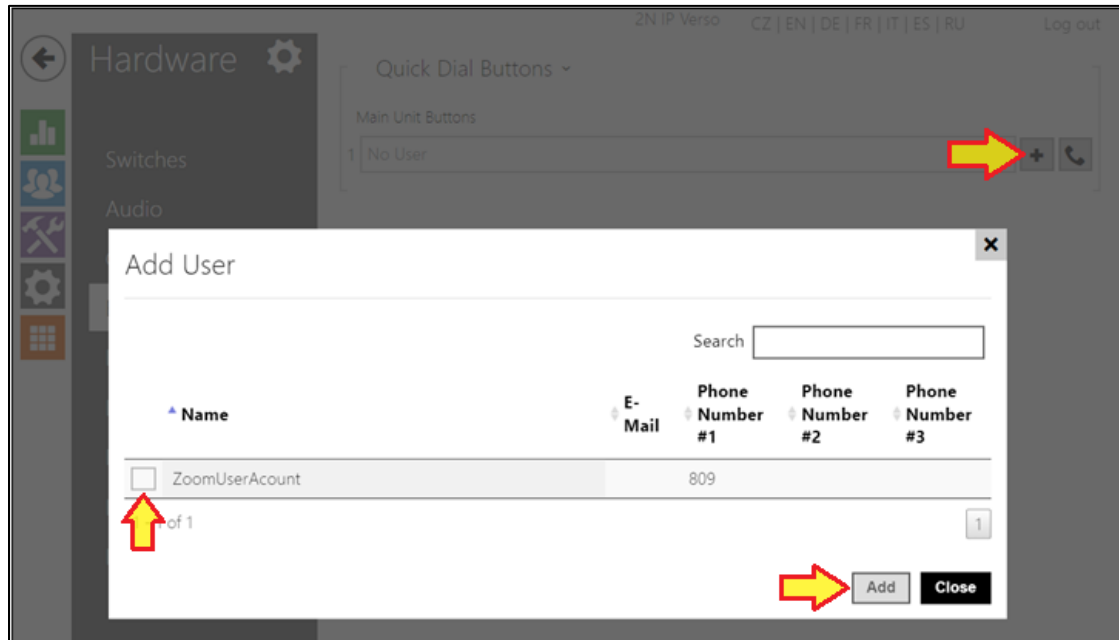


Figure 15 – Assign user to button

6. Click Add and you are ready to use your intercom for calls to your Zoom application.

3.4 Switch control

Zoom application allows you to control 2N IP Intercom's Switch and unlocking door during a call.

1. Log in to your intercom
2. Navigate to **Hardware > Switches** and create a new user



Figure 16 - Homepage - Hardware

3. Choose from Switches which one you would like to set up
 - a. **Enable switch** with checkbox
 - b. Select **controlled output**
 - c. **Test switch** – allows you to check if the controlled output works properly
 - d. Set up **activation code**
 - e. Choose accessibility type (for our purposes we selected **DTMF Only**)

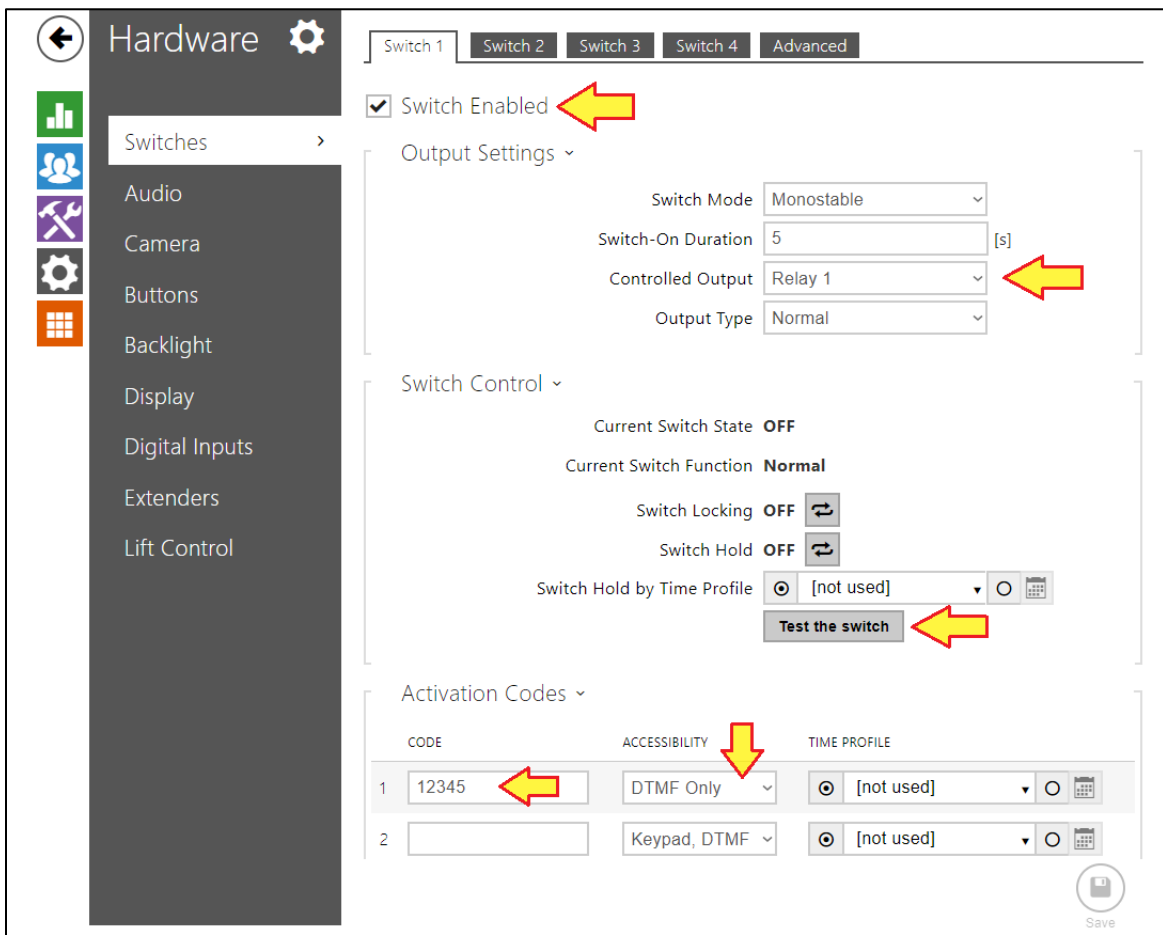


Figure 17 - Switch Control

Once you have successfully set up and tested your Switch, you can start using Zoom applications to control the switch remotely during call. This feature works both on Zoom mobile and Zoom desktop.

1. Set up a call between your 2N Intercom and Zoom application and open keypad in your application.

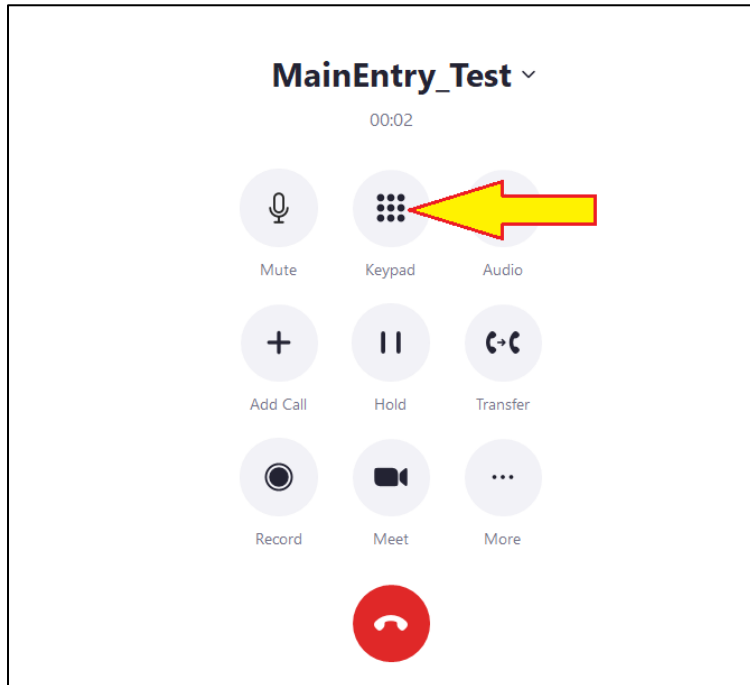


Figure 18 - Switch to keypad during call

2. On your keypad dial up the same DTMF code you set up on your IP Intercom (Figure 17 - Switch Control) and follow the code with a * sign

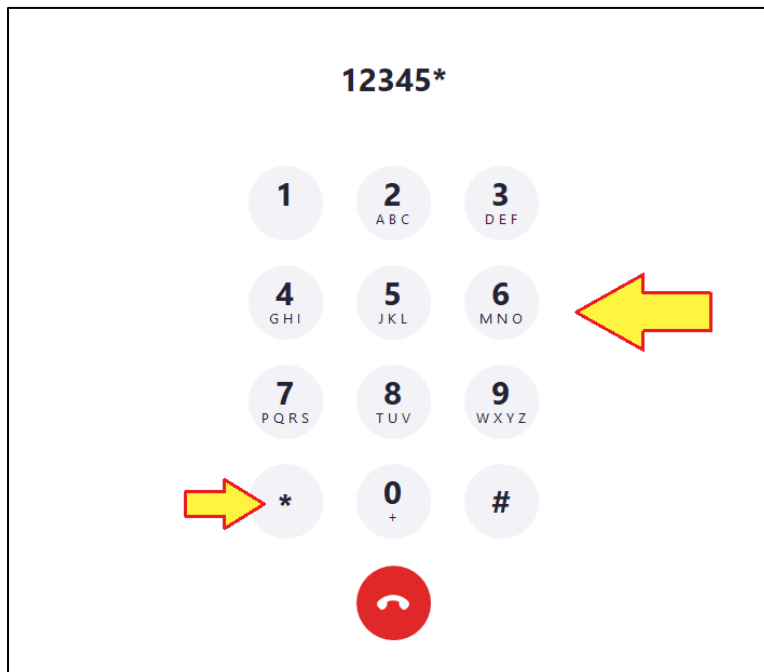


Figure 19 - Dial DTMF code followed by a * sign

4. Call Transfer, Call Conference and Call recording

Once you have set up your Zoom account with multiple users and common areas, it is possible to use additional Zoom application features to allow you increased call control within your organization. All functions except for the from **Meet** function are supported with 2N Intercoms.

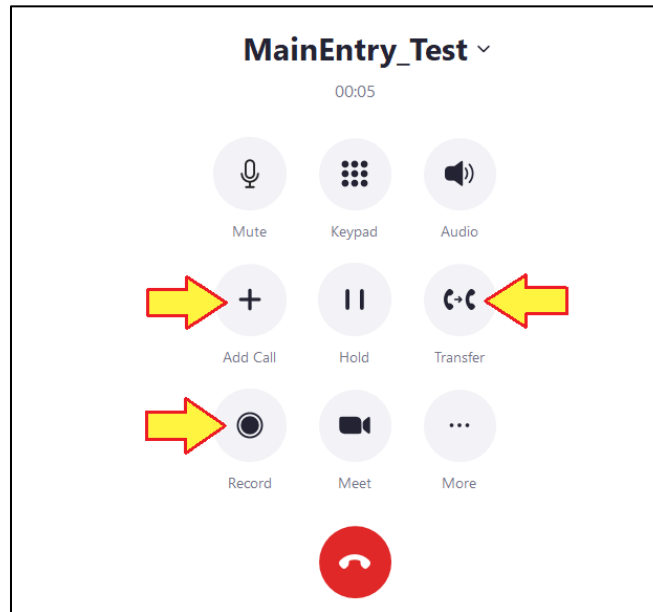


Figure 20 - Zoom call supported features

Of these, three are key with 2N IP Intercom control and access control – Call transfer, Call Conference and Call Recording.

4.1 Call Transfer

During call, you can use the Transfer button to transfer the call to a new caller within your organization, for example having a call transferred from receptionist directly to a member of your organization that is expecting a visitor.

During a call:

- a. the **Transfer** button
- b. type a name or a number where you want to transfer the call and dial it
- c. choose in which way should the call be transferred
 - i. **Warm Transfer**
 - This transfer requires additional action in form of transfer finalization
 - ii. **Blind Transfer**
 - iii. **Transfer to Voicemail**

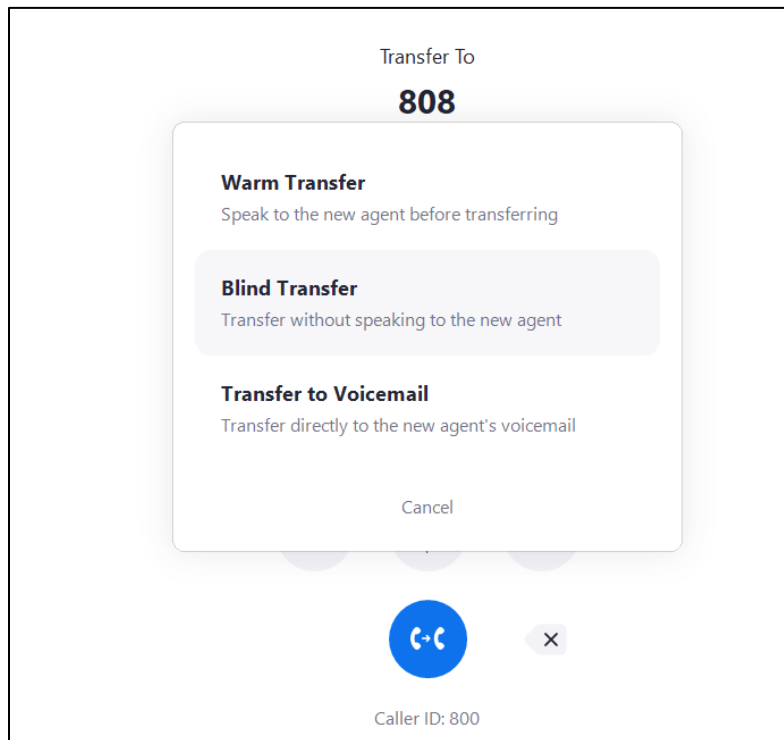


Figure 21 - Call transfer options

4.2 Call Conference

Zoom application allows you to invite another user to the call. In this case, you can invite another user to the call with a 2N Intercom.

During call:

- d. Click the **Add Call** button
- e. Type a name or a number of the contact you wish to add to your call and dial it
- f. Your original call will be on hold until the additional call is answered
- g. Once the additional call is answered, click the merge button to re-connect the original call

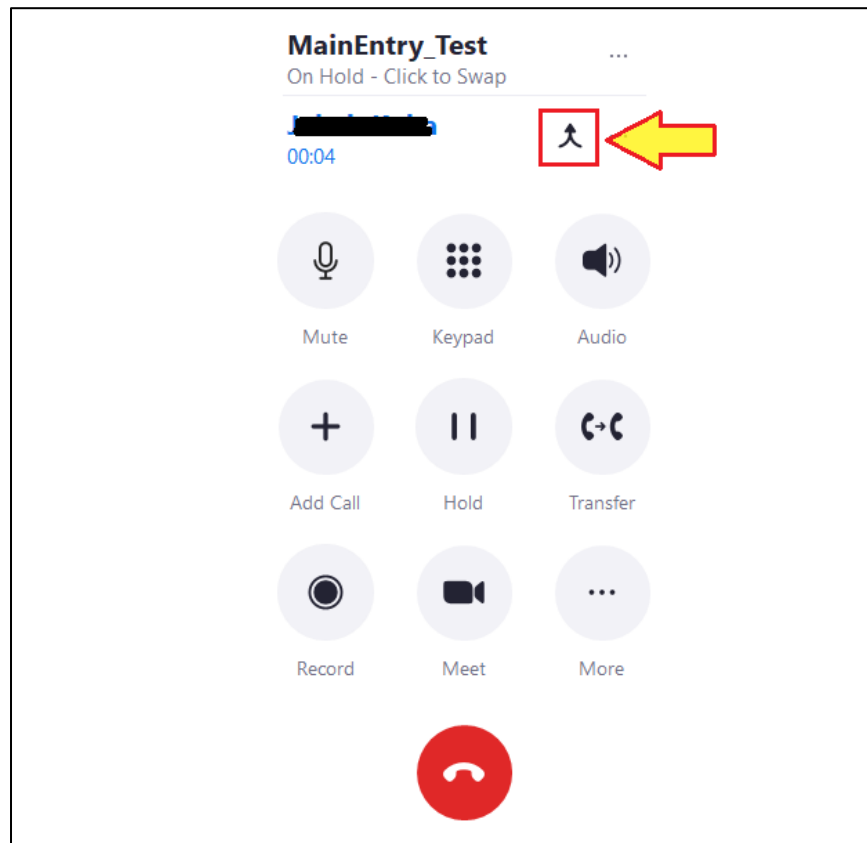


Figure 22 - Call a new contact and merge it to the original call

4.3 Call Recording

Zoom application also provides the option to record calls within the Zoom desktop application. Both parties to the call are audibly notified that a call recording has been started.

During call:

- h. Click the **Record** button
- i. You will hear an audio message that the recording has started
- j. Recording can be stopped at any time during the call with Stop Recording button, or it will be automatically stopped and saved when the call is terminated
- k. Recording from your call can be played back from call log

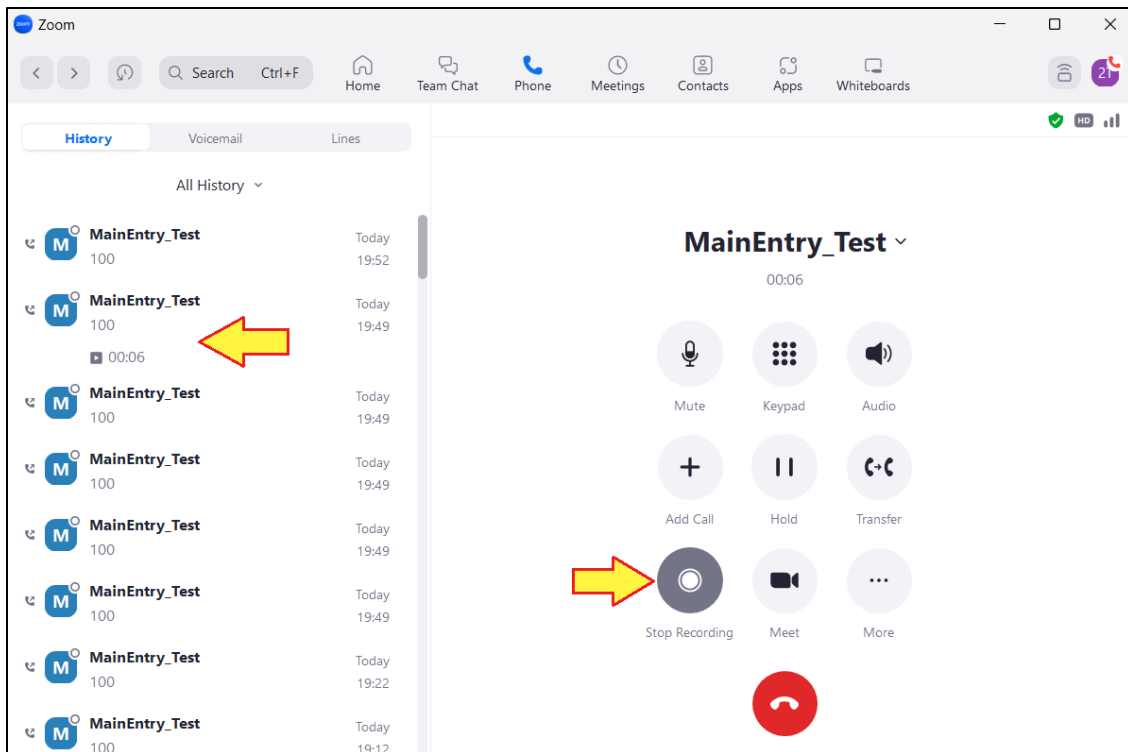


Figure 23 - Play a recording from call log. Cancel recording during call

5. Troubleshooting

Visit 2N FAQ section for dealing with troubleshooting issues: [FAQ](#).

In case of difficulties setting up the 2N IP intercom, kindly follow this [link](#) for intercom detailed installation and configuration manuals, including additional information.

To create a ticket with 2N Technical Support team, please [use this form](#). 2N Technical Support is also available during business hours through live chat with on call technicians directly on the company website at [2n.com](#).

For detailed user guide for Zoom Phone, visit [Getting started with Zoom Phone \(users\)](#) or if you are looking for help with administration of Zoom Phone, visit [Getting Started with Zoom Phone \(admin\)](#).

This section provides instructions on how to resolve common issues. Below are examples of how to format etc. (if applicable)

SIP Registration State = “NOT REGISTERED” / Failure reason: Registration failed

Definition: The server received the register request from the endpoint and responds with an unauthorized message.

- Ensure the SIP credentials (extensions, authentication ID, password) are correct.
- Under Basic Settings > SIP, click on the blue circular arrows to the right of the password field. If the Password is not what it should be, the web browser is probably auto filling the password field. If so, any change on a page containing a password could be filled in an undesired string.

SIP Registration Status = “No reply from Server”

Definition: The device is not able to communicate across the network to the phone server.

- Double check the “SIP Domain (Proxy Server)”, under Basic Settings > SIP tab fields if filled out correctly with the address of your server and port number.
- Ensure the firewall (if present) is not blocking the incoming packets from the server.
- Ensure TLS is configured for SIP Transportation Method (Advanced Settings > Advanced SIP).