

*Communicate anywhere
and anyway you want!*



vitalpbx.com

CONTENT

INTRODUCTION	3
VITALPBX PREPARATIONS	4
1.- VITALPBX SERVER PREPARATIONS	4
VITXI WEBRTC	10
1.- INSTALLATION	10
2.- MENU	12
3.- SETTINGS	14
3.1.- Settings	14
3.2.- Users	20
3.3.- Roles	21
3.4.- Groups	21
3.5.- Agent Profiles	22
3.6.- BFL Settings	22
3.7.- Social Networks	23
3.8.- Mail Settings	23
3.9.- ICE Servers	24
3.10.- PBX Settings	24
4.- CHATS	25
5.- GROUPS	26
6.- CONTACTS	27
7.- CALLS	28
7.1.- Call History	28
7.2.- Voice Mail Messages	29
7.3.- Recordings	29
7.4.- Placing Calls	30

Introduction

VitXi WebRTC is a platform that integrates with VitalPBX in a transparent manner, and allows to have the following services:

- Audio Calls
- Video Calls
 - Screen Sharing
- SIP Chat (Chat between VitXi and SIP Users)
- SMS Chat
- Chat between VitXi Users
 - Voice Notes
 - Emojis
 - File Sharing
 - Image Sharing
- Graphic Voice Mail Interaction
- Call Recordings Viewer
- Presence
- Call History
- Contact List
- And Much More!





VitalPBX Preparations

1.- VitalPBX Server Preparations

For VitXi WebRTC to work perfectly, it is necessary to have a valid FQDN to assign it security certificates. Coming up, we will show you the steps to follow once we have VitalPBX installed on our server with a valid domain.

1.1.1.- Install VitXi

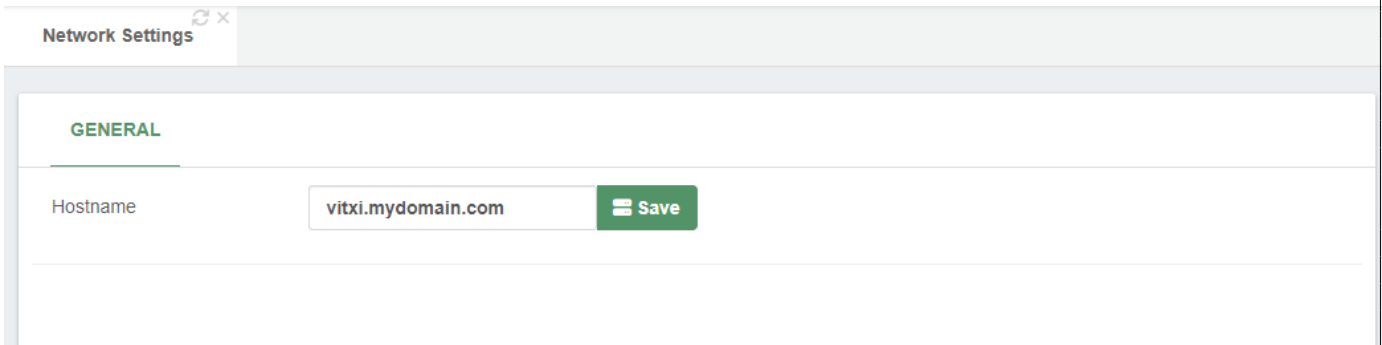
Go to Admin > Add-Ons > Add-Ons and select VitXi and install it. With this we guarantee ourselves at least two free licenses with the Community Version of VitalPBX to try out VitXi WebRTC.

VitXi	4.0.3-2	-		  
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1.1.2.- Create a certificate and enable HTTP

Remember that it is necessary to have a valid FQDN and that our VitalPBX server must have a valid certificate.

It is also very important that the host name of the server matches the FQDN that we are using to generate the Let's Encrypt certificate, so first you must go to Admin/Network/Network Settings and configure the Hostname as shown below.



The screenshot shows a web browser window titled "Network Settings" with a refresh icon and a close button. The main content area is titled "GENERAL" and contains a form with a "Hostname" label and a text input field containing "vitxi.mydomain.com". To the right of the input field is a green "Save" button with a white checkmark icon.

Now we are going to configure the certificate for which we are going to Admin/System Settings/Certificates.

The screenshot shows the 'Certificates' configuration page with the 'GENERAL' tab selected. The form contains the following fields:

Type	Let's Encrypt	Owners Email *	myemail@mydomain.com
Description *	VitXi	Country	United States
Hostname *	mydomain.com	State	Alabama
Sub-Domains	*vitxi.mydomain.com		

Afterwards, go to Admin > System Settings > HTTP Server, and force HTTPS connection and select the certificate created previously.

The screenshot shows the 'HTTP Server' configuration page with the 'GENERAL' tab selected. The form contains the following fields:

HTTP Port	80	HTTPS Enable	Yes
HTTPS Port	443	Force HTTPS	Yes
Certificate	VitXi		

Now go to Settings > PBX Settings > Mini HTTP Server and configure it as shown below. You must select the certificate we have created. Remember to save and Apply Changes. We do this so that VitXi WebRTC PJSIP WebSocket can communicate with VitalPBX.

The screenshot shows the 'Mini HTTP Server' configuration page with the 'GENERAL' tab selected. The form contains the following fields:

HTTP Bind Address	0.0.0.0	8088	Sessions Limit	1000
TLS Bind Address	0.0.0.0	8089	Enable HTTP	Yes
Certificate	VitXi		TLS Enable	Yes

1.1.3.- WebRTC Profile Settings

Go to SETTINGS > Technology Settings> Device Profiles: click on the icon  the list of profiles will be displayed:

- Default IAX2 Profile**
IAX
- Default PJSIP Profile**
PJSIP
- Default SIP Profile**
SIP
- Default WebRTC Profile**
PJSIP

Select the **default WebRTC Profile** and verify that the settings are as follows:

Device Profiles

GENERAL ADVANCED
☰

Profile Type PJSIP

General

Name *	Default WebRTC Profile	Description	Default WebRTC Profile
--------	------------------------	-------------	------------------------

Network

Transport	WSS	Use AVPF	<input checked="" type="checkbox"/>
Qualify Frequency	30	RTP Symmetric	<input checked="" type="checkbox"/>
Qualify Timeout	3	RTCP Mux	<input checked="" type="checkbox"/>
Default Expiration	3600	Asymmetric RTP Codec	<input type="checkbox"/>
Maximum Expiration	7200	Send Diversion Header	<input checked="" type="checkbox"/>
Minimum Expiration	600	Send P-Asserted-Identity	<input checked="" type="checkbox"/>
Outbound Proxy		Send Remote-Party-ID	<input checked="" type="checkbox"/>
Force rport	<input checked="" type="checkbox"/>	Support Path	<input type="checkbox"/>
ICE Support	<input checked="" type="checkbox"/>	Send Connected Line	<input type="checkbox"/>
Rewrite Contact	<input type="checkbox"/>	WebRTC	<input checked="" type="checkbox"/>
Remove Existing	<input checked="" type="checkbox"/>		

Media

Media Encryption	DTLS	RTP Timeout Hold	300
Max Video Streams	35	Direct Media	<input type="checkbox"/>
Max Audio Streams	35	Media Transport Received	<input checked="" type="checkbox"/>
Direct Media Glare Mitigation	None	Media Encryption Optimistic	<input type="checkbox"/>
RTP Keepalive	15	Disable NAT Direct Media	<input checked="" type="checkbox"/>
RTP Timeout	30		

DTLS

DTLS Certificate	VitXi	DTLS Fingerprint Hash	SHA-256
DTLS Setup	Actpass	DTLS Rekey Interval	0
DTLS Verify	Fingerprint		

Save and Apply Changes.

1.1.4.- RTP Settings

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6

Go to CONFIGURATION > PBX Configuration > RTP Configuration and make the following settings:

- **ICE Support:** Yes

The screenshot shows the 'RTP Settings' configuration page. Under the 'GENERAL' tab, the 'ICE Support' checkbox is checked 'Yes'. Below this, the 'ICE Host Settings' section is highlighted with a red border. It contains three input fields: 'Local Address', 'Advertised Address', and 'Include Local Address' (set to 'No'). An 'Add' button is located at the bottom right of this section.

- **ICE Host Settings:** This section will allow you to define which IP addresses will be exposed during ICE negotiations. This is commonly used when your **PBX server is behind a one-to-one NAT**.

Save and Apply Changes.

1.1.5.- Firewall Settings

Now, let's verify that ports 8088, 8089 "**Asterisk HTTP Daemon**" and 6001 "**VitXi WebRTC**" are enabled in the firewall services and rules, for them we enter the following path: "*Admin/Firewall/Services*"

The screenshot shows the 'Firewall Services' configuration page. A table lists various services with their ports and protocols. Red arrows point to the 'Asterisk HTTP Daemon' (port 8088) and 'VIXI WebRTC' (port 6001) entries, which have their status checkboxes checked.

Service Name	Port	Protocol	Actions
PJSIP	5060-5061	Both	<input checked="" type="checkbox"/> <input type="checkbox"/>
DNS	53	Both	<input checked="" type="checkbox"/> <input type="checkbox"/>
NTP	123	UDP	<input checked="" type="checkbox"/> <input type="checkbox"/>
DHCP	67-68	UDP	<input checked="" type="checkbox"/> <input type="checkbox"/>
HTTP	80	TCP	<input checked="" type="checkbox"/> <input type="checkbox"/>
SSH	22	TCP	<input checked="" type="checkbox"/> <input type="checkbox"/>
RTP	10000-20000	UDP	<input checked="" type="checkbox"/> <input type="checkbox"/>
IAX2	4569	UDP	<input checked="" type="checkbox"/> <input type="checkbox"/>
mDNS	5353	UDP	<input checked="" type="checkbox"/> <input type="checkbox"/>
HTTPS	443	TCP	<input checked="" type="checkbox"/> <input type="checkbox"/>
Asterisk HTTP Daemon	8088-8089	Both	<input checked="" type="checkbox"/> <input type="checkbox"/>
SIP	5062-5063	Both	<input checked="" type="checkbox"/> <input type="checkbox"/>
VPEX API	3000-3001	TCP	<input checked="" type="checkbox"/> <input type="checkbox"/>
VIXI WebRTC	6001	TCP	<input checked="" type="checkbox"/> <input type="checkbox"/>

And "*Admin/Firewall/Rules*"

Service	Ports	Source	Destination	Action	
RTP	10000-20000			ACCEPT	
PJSIP	5060-5061			ACCEPT	
HTTP	80			ACCEPT	
HTTPS	443			ACCEPT	
SSH	22			ACCEPT	
DHCP	67-68			ACCEPT	
DNS	53			ACCEPT	
NTP	123			ACCEPT	
IAX2	4569			ACCEPT	
mDNS	5353		224.0.0.251	ACCEPT	
SIP	5062-5063			ACCEPT	
VPEX API	3500-3501			ACCEPT	
VIXI WebRTC	6001			ACCEPT	
Asterisk HTTP Daemon	8088-8089			ACCEPT	

Note: in case you have another firewall, you must enable the ports mentioned above.

1.1.6.- Create a VitXi Extension

Go to PBX > Extensions > Extensions and create an extension with the following parameters:

Extensions

GENERAL | VOICEMAIL | RECORDING | ADVANCED | FOLLOW ME | INCOMING ROUTES | CONTACT INFO

Extension * Internal CID

Name * External CID

Class of Service Emergency CID

Features Password Account Code

Email Addresses Language

Devices

Technology PJSIP IAX2 VIRTUAL NONE Emergency CID

User Device * Dispatchable Location

Password * Deny

Device Description * Permit

Profile Ring Device Yes

Max Contacts WebRTC Client Yes

Codecs Mobile Client No

DTMF Mode

- **Technology**, PJSIP
- **Profile**, Default WebRTC Profile
- **WebRTC Client**, Yes

Save and Apply Changes

VitXi WebRTC

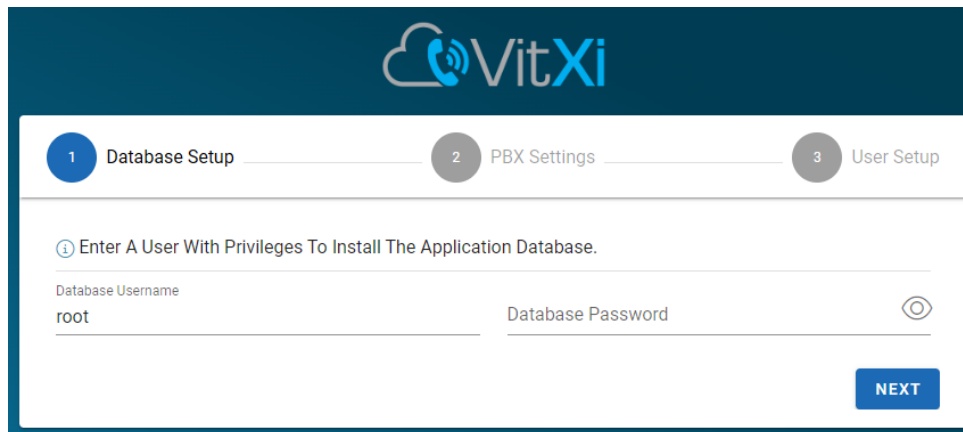
VitXi WebRTC is a full web application that integrates with VitalPBX and communicates through the PJSIP protocol over WebRTC.

1.- Installation

VitXi 4.x works with VitalPBX 4 and onwards. To install VitXi WebRTC it is necessary to go to the Add-On module (Admin > Add-Ons > Add-Ons) and press the green install button next to it. After a couple of minutes, the installation would be done, and you can proceed to its URL (<http://IP-ADDRESS/VitXi>) and proceed with the installation wizard. You can press VitXi's name on the Add-Ons module to quickly go to this URL.

Now, we will see the installation wizard that will guide us through the final steps for configuration.

a.- Database Settings:

The screenshot shows the VitXi installation wizard interface. At the top, there is a dark blue header with the VitXi logo. Below the header, there are three steps: 1 Database Setup (highlighted in blue), 2 PBX Settings, and 3 User Setup. The main content area has a light blue background and contains the following text: "Enter A User With Privileges To Install The Application Database." Below this, there are two input fields: "Database Username" with the value "root" and "Database Password" with a password icon. A blue "NEXT" button is located at the bottom right of the form.

- **Database User**, MySQL user, if we are installing on the same server as VitalPBX, you can leave the username 'root' and blank password. Else, you will need to create a new MariaDB user to access remotely.
- **Database Password**, if we are installing on the same server as VitalPBX, the root user does not have a password. When accessing remotely, you must create a new username and password in MariaDB.

b.- PBX Settings:

VitXi

1 Database Setup 2 PBX Settings 3 User Setup

Host: localhost
 Https Port: 443
 PJSIP Port: 5060
Configure The Port To Be Used For Extension Registration

WSS Port: 8089
If You Have A Firewall, Don't Forget To Allow TCP Port Through So Your Client Can Connect.

API Port: 3501
Indicate The Port Used By The API.

Application Key
If you are installing on the same server as , you can leave this field blank. a key will be automatically generated.

PREVIOUS NEXT

- **Host**, if we are installing on the same server as VitalPBX, it is recommended to leave this field as **localhost**. Else, we must input the host or remote IP Address.
- **Port**, if our server has a valid certificate and all the requests are redirected to HTTPS, we must configure port 443. If we are not using HTTPS, we configure port 80. If we are accessing the PBX remotely, remember to open these ports on the firewall.
- **Application key**, if the VitXi is located on the same server as VitalPBX, it is not necessary to generate an API key since it will be generated automatically. If VitXi is installed on a different server to VitalPBX you will need to generate an API key under Admin > Admin > Application Keys. You then copy it and paste it here.

c.- User Setup:

VitXi

1 Database Setup 2 PBX Settings 3 User Setup

Name: John Doe
 Username: admin
 Email: myemail@mydomain.com

User Password:
 Password Confirmation:
 Timezone: America/New_York

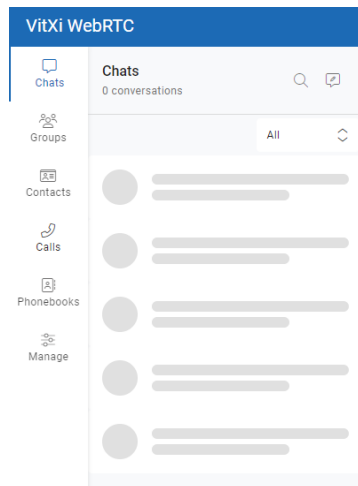
PREVIOUS FINISH

- **Name**, full name for the administrator user in VitXi.
- **Username**, is the administrator username to use in VitXi, usually 'admin.'
- **User Password y Password Confirmation**, admin user password.
- **Time zone**, here you choose a default time zone.

d.- Enter the Username and Password previously created.





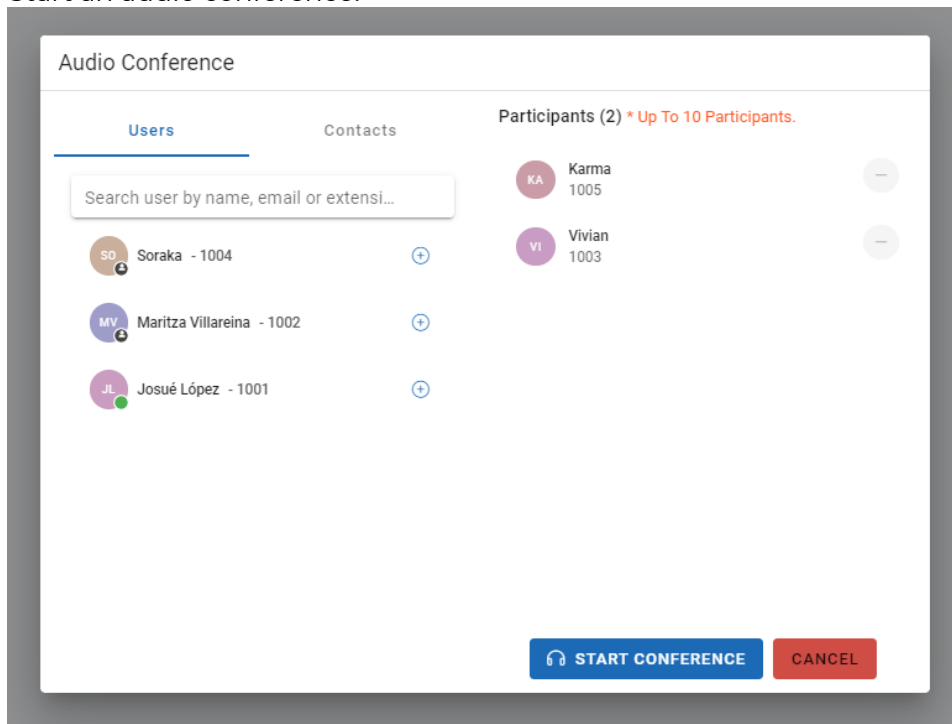
2.- Menu




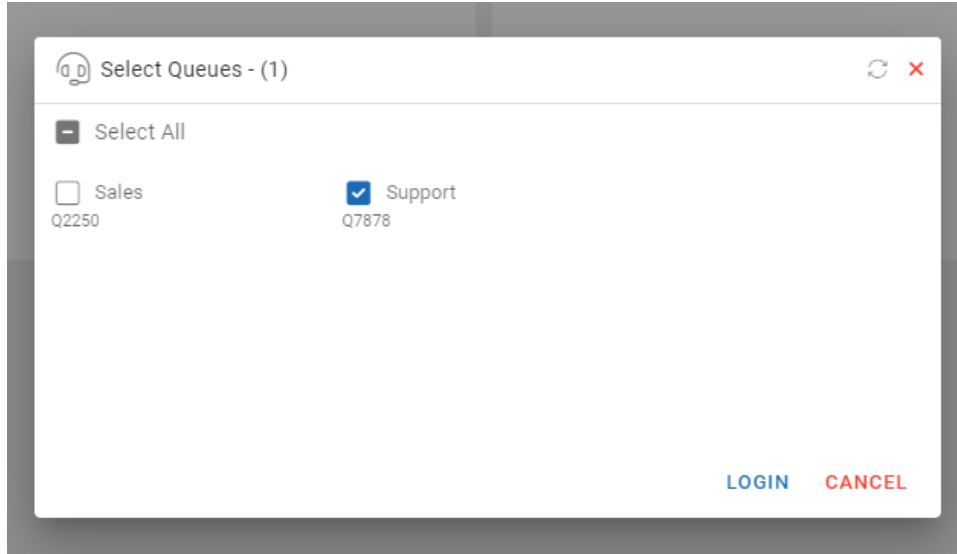
On the menu we got the following options:



- **Chats**, list of people with which the user has had some type of chat.
- **Groups**, List of available groups with their respective members.
- **Contacts**, the contact list added by the user manually. To enable/disable BLF for a number, you should remove the number and then add it again.
- **Calls**, The user's Dialpad and call history.
- **Phonebooks**, phone directories that are extracted from the VitalPBX Phonebook.
- **Manage**, main menu to configure the system.

- **Users**, Create and edit VitXi Users.
 - **Role Manager**, Create and edit privilege profiles for users.
 - **Groups**, Create and edit groups to which users belong.
 - **Agent Profile**, Create and edit profiles for call center agents.
 - **BLF Settings**, Manage the number of BLF contacts allowed per user to prevent server overload and optimize performance. Set a value between 1 and 100.
 - **Presence Status**, Create and edit the possible states of user presences.
 - **Social Networks**, Configure the URL of each Tenant's social networks.
 - **Mail Settings**, Configure the email account to send emails.
 - **ICE Server**, Configure the ICE Server. We recommend leaving the default ones.
 - **PBX Settings**, Configure VitalPBX location. It is only necessary to configure when you have a connection problem or want to use a VitalPBX that is on another server.
-  Voicemail Messages indicator.
 -  Start an audio conference.



- Allows you to initiate an audio conference with several contacts you have added.
-  Login/Logout and Pause/UnPause Agent in the Queues.
 - By clicking on the icon, the agent will automatically log in to all the queues to which it belongs. Once logged in, just click on the icon again to log out of all the queues.
 - If you want one or more agents to be able to choose which queue to log in to manually, you will have to modify or create a specific role with the corresponding permissions.



-  Preferences
 - **Settings**, Configure and view user-related settings.
 - **Profile**, Show Name, Username and Extension.
 - **Account**, Change user password.
 - **General**, Desktop Notifications and Dark Mode.
 - **Audio**, Configure audio settings.
 - **Headset Integration**,
 - **Video**, Configure video settings.
 - **Calls**, Configure Ringtone, Auto Answer and Call History Settings.
 - **Call Center**, Configure Login on Startup, Logout when closing VitXi and CRM Settings.
 - **Language**, Change Language.
 - **Shortcuts**, List of Shortcuts.
 - **About**, Show VitXi version.
 - **Troubleshooting**, clear local app data and cache.
-  VitXi client registration status.

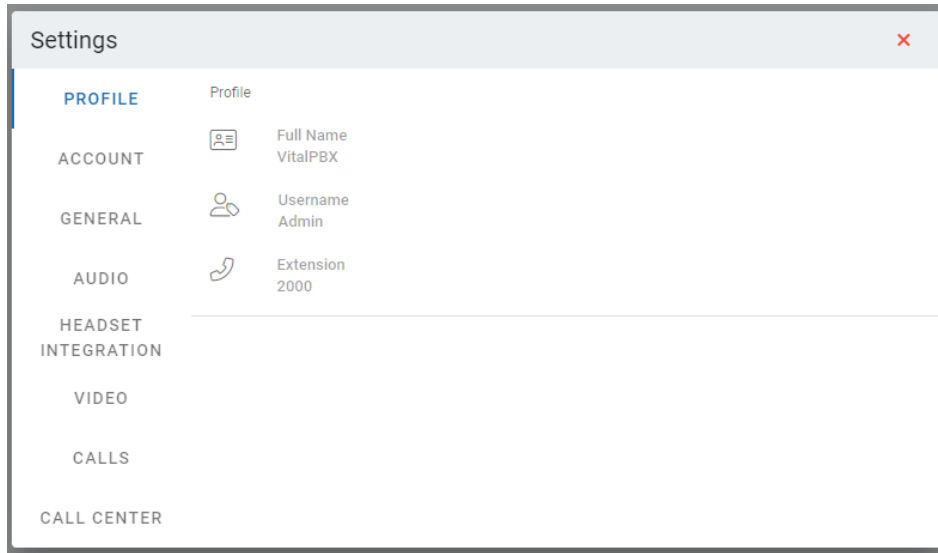
3.- Settings

3.1.- Settings

The first thing that we are going to configure is the System Settings, for which we will go to the main menu and select settings. This will be displayed in the following screen.

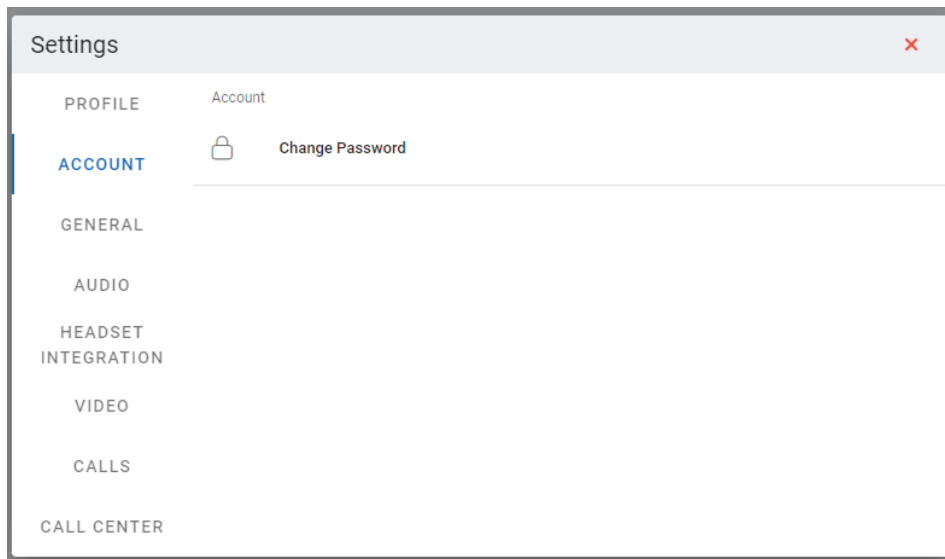
Profile

Shows the name, Username and extension number of the user.



Account

Change the user's password.

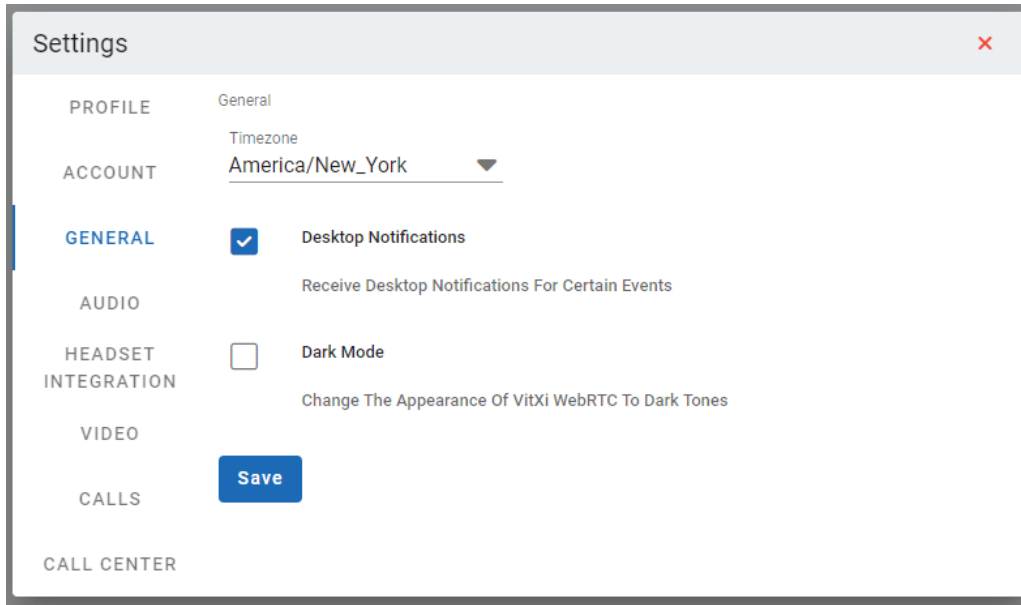


General

Timezone, the time zone for the user. This way, the user will see their own time zone, regardless of their location.

Desktop Notifications, the user can receive a notification for certain events, for example, receiving a call, a message, and more.

Dark Mode, here you can select the dark mode view of the interface.



Audio

Microphone, Select the microphone to use.

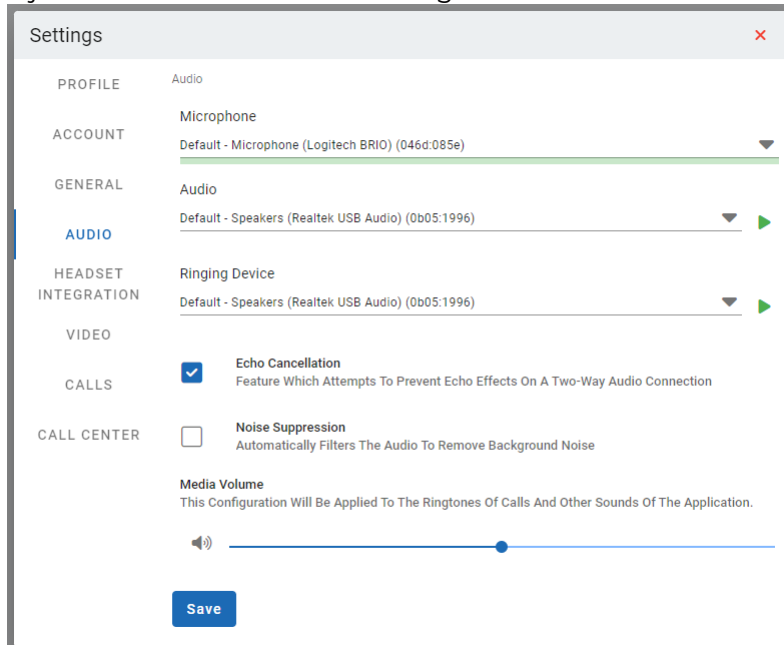
Audio, Select the audio output to listen to.

Ringling Device, Select the audio output to hear the ring when a call comes in.

Echo Cancellation, this is used to prevent the echo effect during a call.

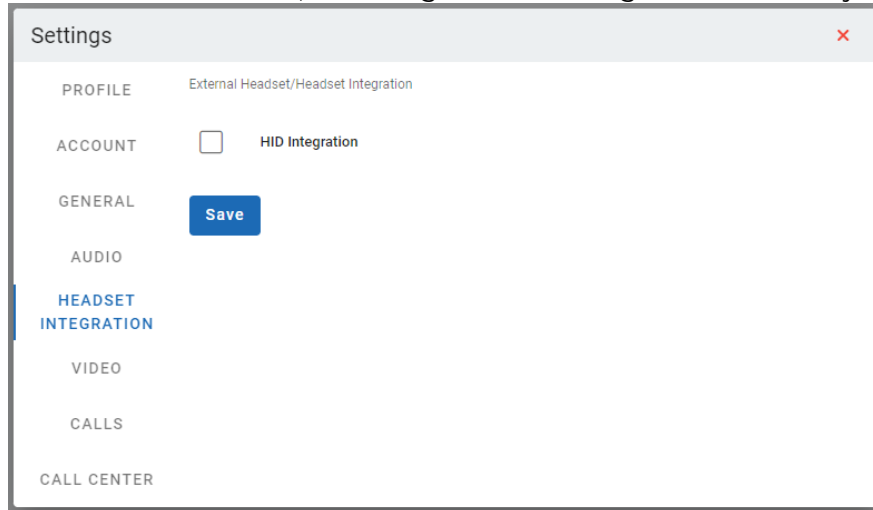
Noise Suppression automatically filters and eliminates background noise.

Media Volume, Adjust the volume to hear the ringtone when a call or notification comes in.



Headset Integration

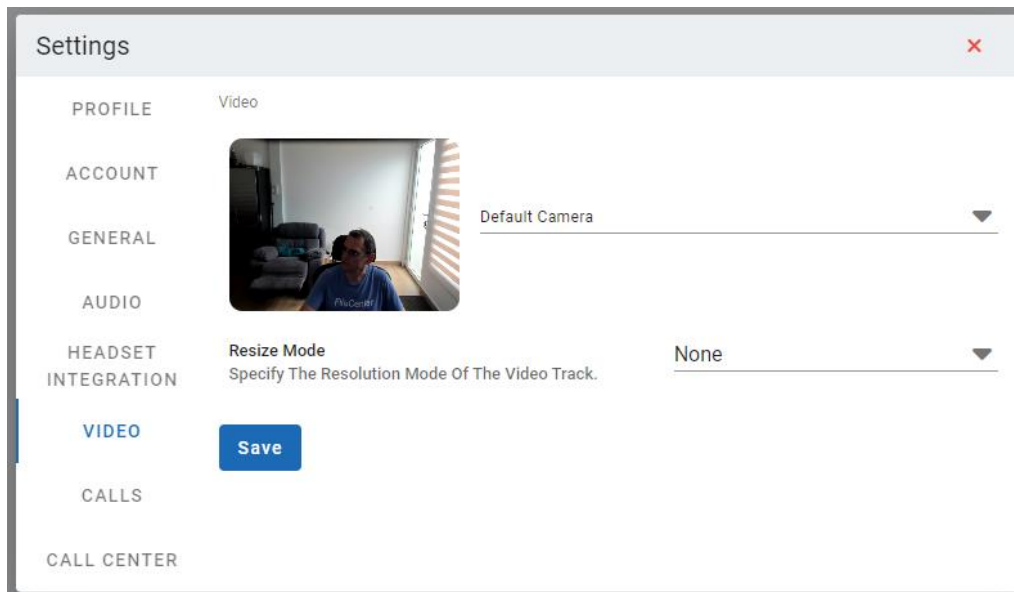
HID enablement module for streamlined call actions using headsets. Enhance the calling experience with one-touch controls, including call answering and volume adjustments.



Video

Camera, Select the camera to use.

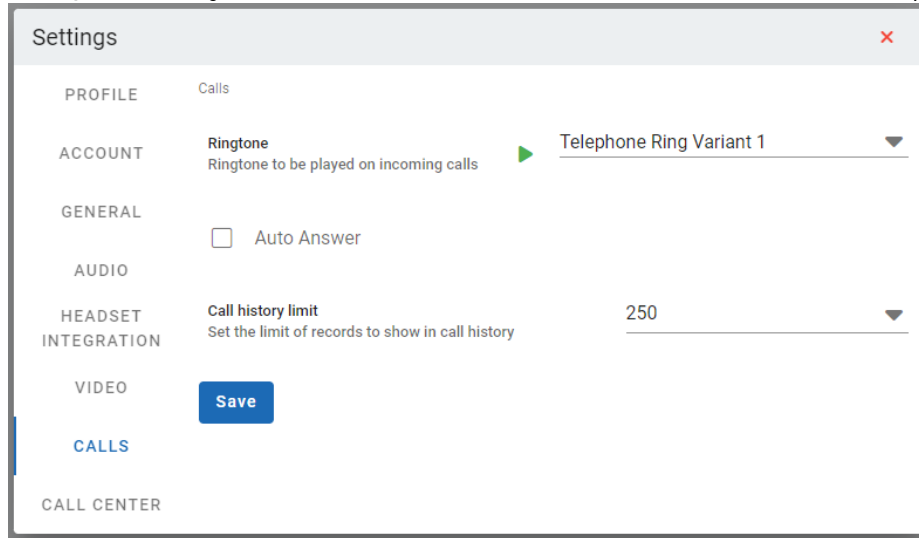
Resize Mode, Specify the resolution mode of the video track.



Calls

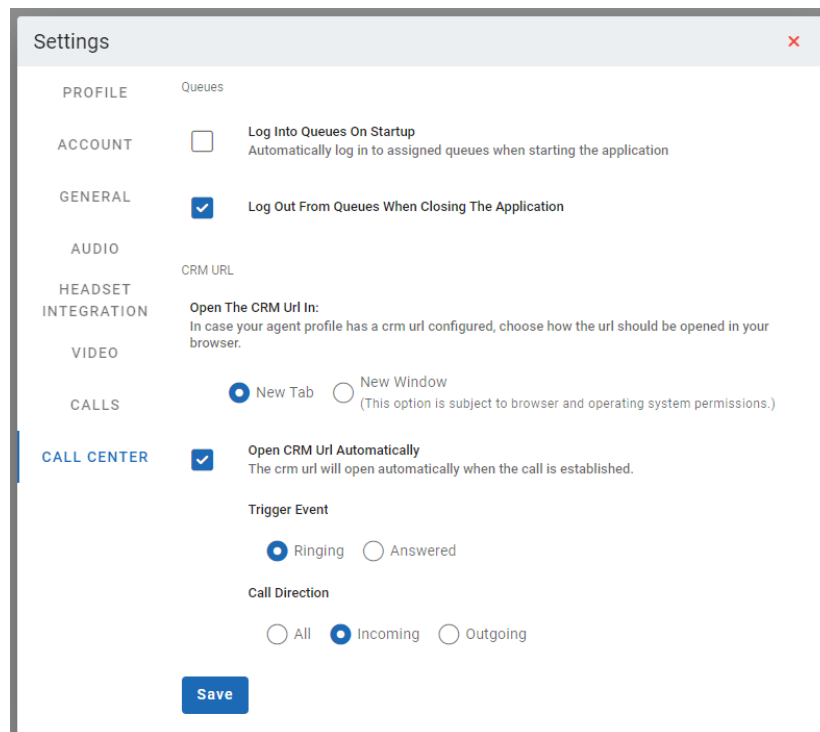
Ringtone, here you can select the ring tone to listen to when the user is receiving a call.
Auto Answer, this will answer the call automatically.

Call history limit, with this you can establish the call limit to show on the Calls option .



Call Center

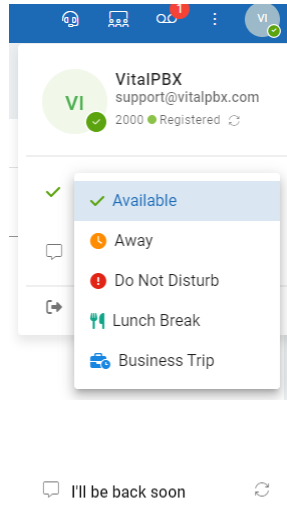
Automate Agent Login and Logout when starting and closing VitXi. CRM integration options are also configured to display pop-up windows.



Presence

Presence Status, select the presence status with which we desire to have when login in.

Additional Information, if the user wants to show some additional message that other users that are connected will see.



Here, the system administrator can create the different presence status that users can use. By default, there are five (5) statuses which are, Available, Do Not Disturb, Lunch Break, and Business Trip, however, you can add more.

A screenshot of a 'Create' form for adding a new presence status. The form has a title 'Create' and a hamburger menu icon. It contains a 'name' input field, a 'select icon' dropdown menu with 'Ad ad' selected, and a 'status color' section. The 'status color' section features a color picker with a red circle selected, a hex color input field containing '#FF0000FF', and the label 'HEX'. A green 'Save' button is located at the bottom right of the form.

3.2.- Users

Now, we will create the users and assign them to the groups we want to. A user can belong to one or more groups.

The screenshot shows a 'Create' user form with the following fields and values:

- name**: (empty)
- username**: (empty)
- email**: (empty)
- password**: (empty)
- role**: User
- tenant**: VitalPBX LLC - (main tenant)
- groups**: public list
- extension**: (empty)
- agent profile**: default profile
- phonebooks**: (empty)

A green 'Save' button is located at the bottom right of the form.

Name, user's full name.

Username, username with which the user will log into VitXi WebRTC.

Email, the email address associated with the user, is very useful to recover the password.

Password, password with which the user will log into VitXi WebRTC.

Role, the type of user. There are three types of users.

- **Super Administrator**, capable of managing all the users and tenants.
- **Administrator**, capable of managing users within its tenant.
- **User**, VitXi WebRTC end-users.
- **Agent**, When it is a Call Center Agent, we recommend this profile already activates an icon in the interface to facilitate the Login/Logout, Pause/Unpause of the Agent in queues.

Tenant, tenant to which the user belongs to.

Groups, groups to which the user belongs to.

Extension, extension associated to the user. For this, the user needs to have a VitXi Device created on VitalPBX.

Agent Profile, Assigning an agent profile to be used in case the assigned role has call center permission.

Phonebooks, phonebooks to which the user will have access to.

3.3.- Roles

To facilitate the creation of users with their respective permissions, roles are created that are assigned to different types of users.

The screenshot shows a 'Create' form for roles. It has a title bar with 'Create' and a menu icon. Below is a 'Name' input field. The 'Permissions' section contains a grid of checkboxes with labels: 'Miscellaneous', 'Management', 'Call Center', 'View Recordings', 'Roles', 'Call Center Features', 'Update Social Networks', 'Users', 'Queue Selection', and 'Agent Profiles'. Each checkbox has an information icon to its right. A green 'SAVE' button is located at the bottom right of the form.

3.4.- Groups

With the purpose that the VitXi users become visible between each other, it is necessary that each user has a group assigned to them, and afterwards, we give access to other users to see this group. This way, you can create multiple groups of users limiting them to which groups they can see. For this, we first create the group name.

The screenshot shows a 'Create' form for groups. It has a title bar with 'Create' and a menu icon. Below are two input fields: 'name' and 'description'. Underneath is a 'tenant' dropdown menu with 'VitalPBX LLC - (main tenant)' selected. A green 'Save' button is at the bottom right.

3.5.- Agent Profiles

When we are using Call Center Agents in VitXi and we want each agent group to have a different profile in terms of types of breaks or integration with CRM, we must create profiles with this information.

Below we show the interface where these profiles are created.

The screenshot shows a web interface for updating an agent profile. At the top, it says "Update" and includes a warning: "* It Is Not Allowed To Change The Tenant." Below this, another warning states: "* The Default Profile Cannot Be Deleted." The form fields include: "Name" (default profile), "Is Global" (checked), "CRM URL (Optional)" (base URL: https://your-crm.com), "Custom Website Tab" (tab name), "Parameter Name" (Value: {call_type}, Use Dynamic Variable), and "web url" (with a note: "* enter a secure web site (https://) that you can trust. this is important to keep your users safe."). Under "Reasons For Pauses", there are three entries: "Pause Administrative", "Pause Lunch", and "Pause WC". At the bottom right, there are three buttons: "UPDATE", "DELETE", and "CANCEL".

Name, Name of the profile to create.

Is Global, If it is going to be shared to be used by all Tenants.


CRM URL, you can set the URL/endpoint of your CRM. You can configure dynamic or static variables. The CRM URL will open when the agents click the CRM button in the call layout.

Custom Website Tab, configure a URL/website to be inserted as a "custom tab." This way, agents will be able to visit the site without leaving Vitxi.


Reason for Pauses, We define all possible reasons for pauses.

3.6.- BFL Settings

Manage the number of BLF contacts allowed per user to prevent server overload and optimize performance. Set a value between 1 and 100.

 **BLF Contact Control**
Manage the number of BLF contacts allowed per user to prevent server overload and optimize performance. Set a value between 1 and 100.


Limit
50





3.7.- Social Networks


Here the URLs of the social networks are configured. This can be configured by Tenants. When they are configured, some icons will appear at the bottom right.


Tenant
VitalPBX LLC

 <https://facebook.com/vitalpbx>
Set the username of your facebook page

 <https://twitter.com/vitalpbx>
Set the username for your twitter account

 <https://youtube.com/vitalpbx>
Set the username of your youtube channel

 <https://instagram.com/vitalpbx>
Set the username for your Instagram account





3.8.- Mail Settings

Configuration of the email account to be able to send email in case some users need to reset their password.

smtp	host smtp.mailtrap.io	port 2525
username required	password required	encryption required
from address	from name	Send email test

Save

3.9.- ICE Servers

Servers

Server - 1

STUN or TURN URI (1)
stun:stun1.l.google.com:19302

STUN or TURN URI (2)
stun:stun2.l.google.com:19305

Authentication

+ ADD SECONDARY SERVER

Reset **Save**

3.10.- PBX Settings

Even though this information is asked at the moment of installation, it is also possible to perform some modifications afterwards.

Settings Connected

host localhost	https port 443 <small>VitalPBX HTTPS port (useful for obtaining assets from the branding module).</small>	PJSIP Port 5060 <small>Configure the port to be used for extension registration</small>
WSS port 8089	API port 3501 <small>Indicate the port used by the API.</small>	

If you have a firewall, don't forget to allow TCP port through so your client can connect.

Edit application key

Save

4.- Chats

In Chat we can see the history, as well as start a conversation or continue a previous conversation.

The screenshot displays the VitXi WebRTC chat interface. On the left, a sidebar shows a list of 12 chat conversations. The selected chat is with 'Joseph Montes'. The main chat area shows a conversation history starting from 01/10/2023. The messages are as follows:

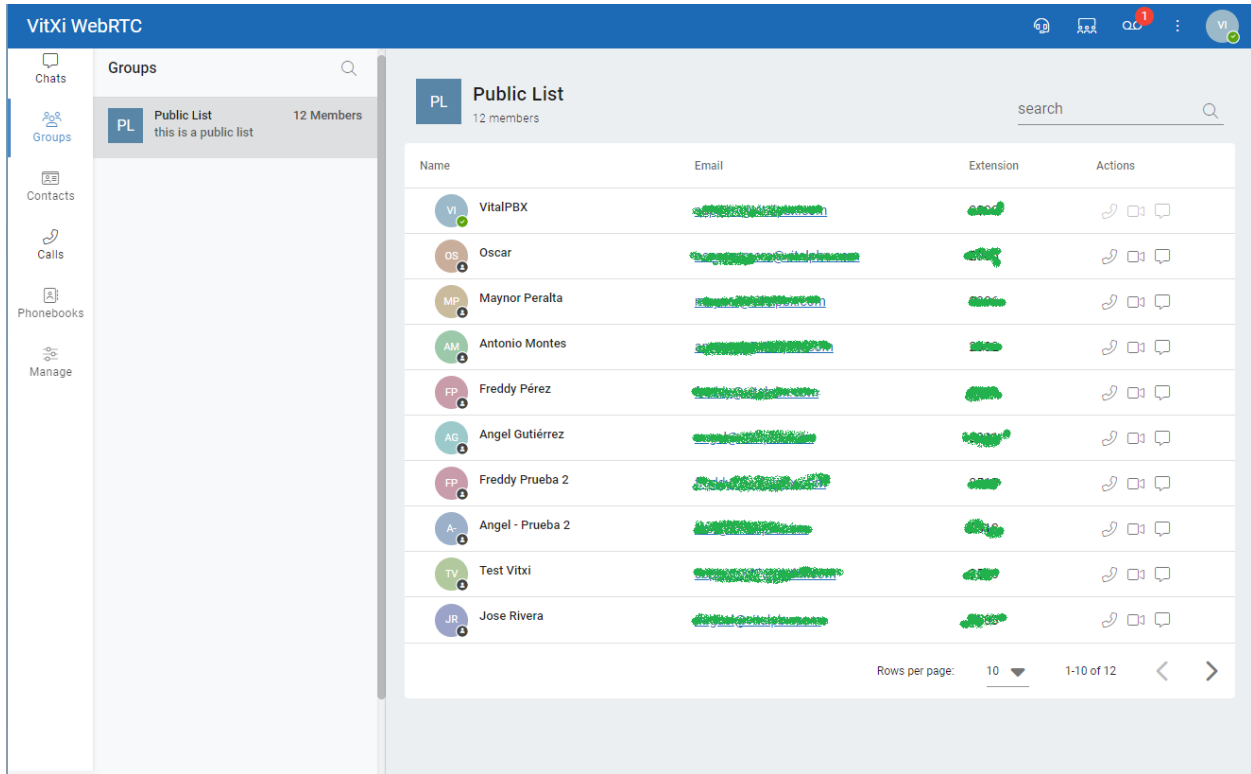
- 01/10/2023: Joseph Montes: Hello! How are you doing today?
- 01/10/2023: I'm fine, do you have the documents?
- 01/10/2023: Joseph Montes: Respondiendo, test.
- 01/10/2023: Joseph Montes: yes, I have them. Let me send them now.
- 01/10/2023: Joseph Montes: [Document: VitXi Texts.docx]
- 01/10/2023: Thanks 😊
- 04/15/2023: Test

The chat input field at the bottom contains the text 'message'.



5.- Groups

It is possible for group users so that they can be accessed only by the members of their group, thus the identity of the different sectors of a Company can be preserved.

With this option it is possible to see the members of the groups to which the user has access.



6.- Contacts

To add a new contact, it is necessary to go to 'Contacts' , and press the 'Add Contact' option  and then it will show us the following form.

Contact information

first name* required last name* required company

job title contact type* internal email

avatar

Mark as favorite

Phones

type* Internal (SIP) number BLF + ADD PHONE

* Indicates required field

CLOSE CREATE

First Name, the contact's First Name.

Last Name, the contact's Last Name.

Company, name of the Company where the Contact works at.

Job Title, the job they perform at the company they work at.

Contact Type, the type of contact, Internal or External.

Email, contact's email address.

Avatar, image to remember the contact with.

Mark as favorite, if you mark it as favorites, it will be grouped in the favorite's menu.

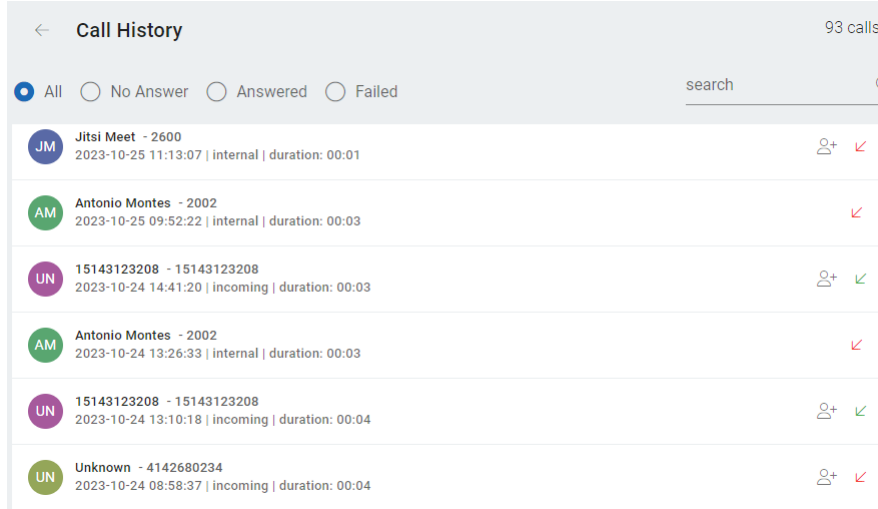
Phones, contact phone numbers, they can be SIP, cellphone, or any other type.

BLF, If the BLF option is selected (Only valid for extensions SIP on the same PBX), we will see the status of the extension in real time.

7.- Calls

7.1.- Call History

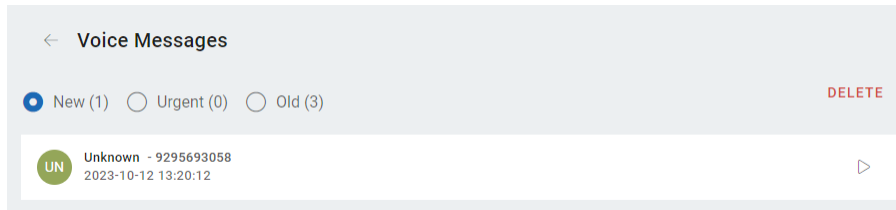
To see the call history, we must go to the Calls icon and press Call History. There we can see the last calls, for performance this history is limited by default to 250, however you can modify it.



7.2.- Voice Mail Messages

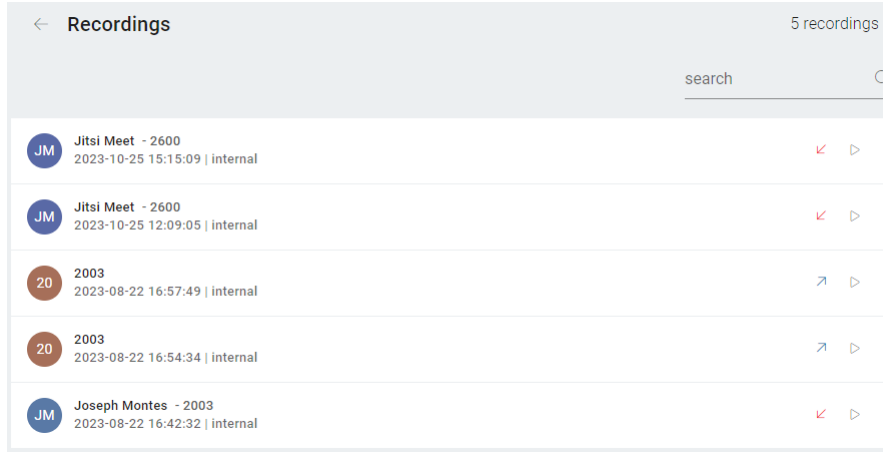
It is possible to view all your Voicemail messages from VitXi WebRTC, which allows us to perform the following actions:

- Listen
- Delete





7.3.- Recordings

It is possible to listen to the user's call recordings from the extension they have associated to them.



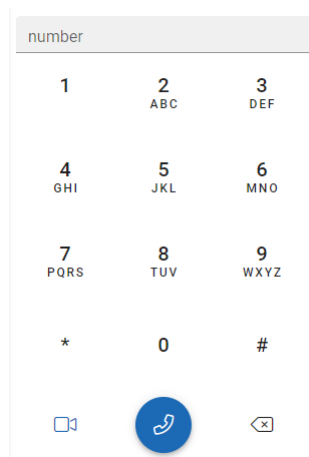
7.4.- Placing Calls

There are many ways to perform calls through VitXi WebRTC. The easiest way is by selecting the user and then pressing one of the three options available:

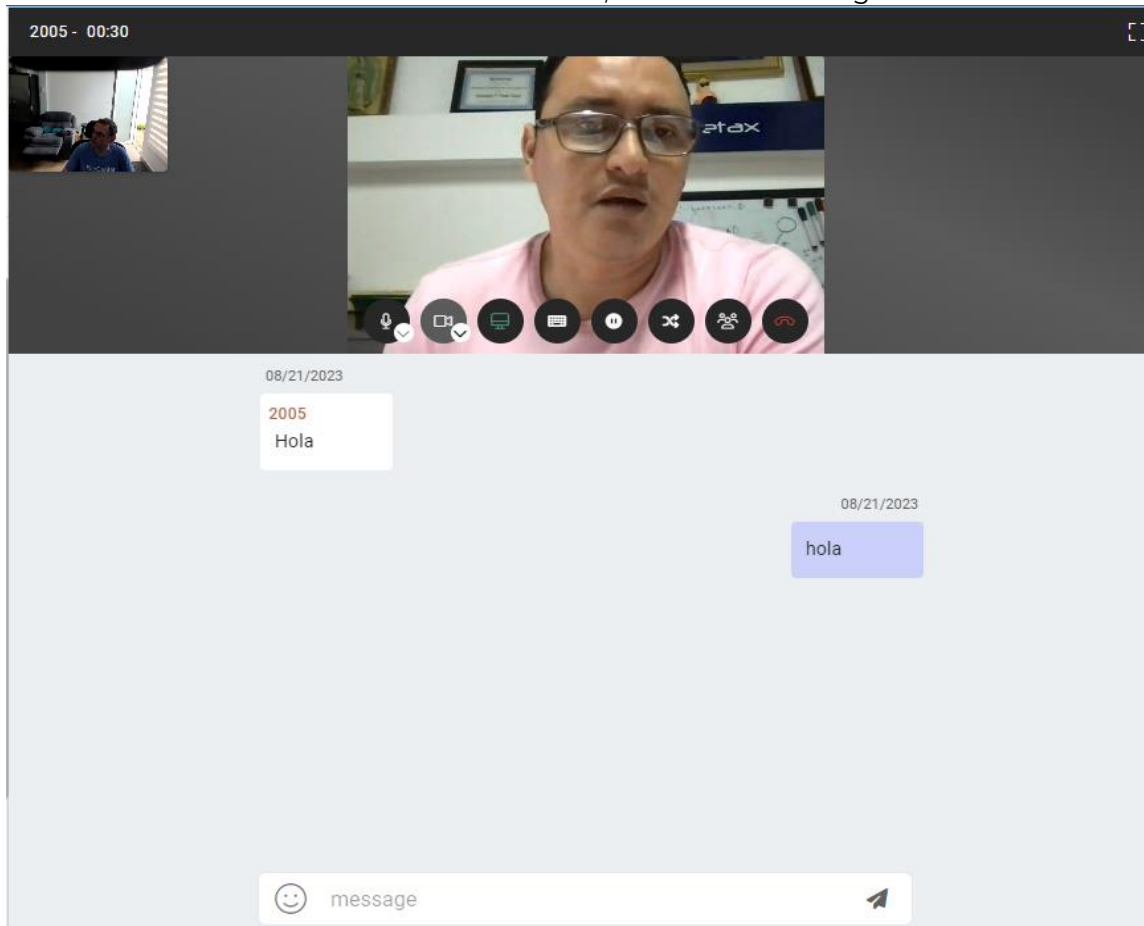
-  , Audio/Video Call
-  , Chat message











We can also use the Number Pad by pressing the  button.



When a call is established, the following screen is displayed.



When there is an established call, the following options are displayed, which we will describe below.

-  Mute the microphone by selecting a new one from a list.
-  Enable/disable video casting or select a video source from a list.
-  Share screen.
-  Keyboard for sending DTMF tones
-  Put the call on hold.
-  Call transfer.
-  Add a new call to the call to put it in conference.
-  Hang up on the current call.