

IP Phone VPS-804P



Instruction manual

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1. IP Phone VPS-804P

Thank you for purchasing the VPS-804P. It is a fully functional mobile web enables voice communication over the network. This device works like a traditional phone: You can make and receive calls and use other functions specific to a regular phone.

This phone has other features data services that are unavailable in traditional phones.

This manual contains information on the features and services available on the device.

1.1. Notice Information

Before you connect and start using your phone, please read the following information. They are important, the unit will operate in a safe and reliable.

Safety cautions

1. To use the power adapter that delivered with the phone. Other power adapters may damage the phone.

2. The phone is only for indoor use. And also avoid in high humidity, water and some other liquids.

3. Do not use the phone during thunderstorms.

Disposal of the phone



This symbol indicates that the product is classified as electrical or electronic equipment and should not be disposed of with other commercial or household waste at the end of its working life.

Cleaning

To clean the device, use an anti-static cloth. Please avoid cleaning liquids as they might damage the surface or internal electronics of the phone.

"The manufacturer reserves the right to make changes to the product without prior notice." www.slican.pl e-mail: office@slican.pl

1.2 Packing List

The package contains the following parts, please check if all the items are not missed:

PHONE



THE FOOTSTAND



HANDSET



HANDSET CORD



POWER ADAPTER



ETHERNET CABLE



QUICK INSTALLATION REFERENCE



2. Phone Installation2.1. Attach the Foot stand



2.2. Connect the Handset and optional Headset.



2.3. Connect the Network and Power



2.4.Initialization

After your phone has been powered up, the system boots up and performs the following steps:

- automatic phone initialization

The phone finishes the initialization process by loading the saved configuration. The phone LCD screen will display "Booting":



- and then show "Initializing "during the initialization process.

Initializing

By default the phone attempts to contact a DHCP server in your network in order to obtain its valid network settings, e.g. IP address, subnet mask, default gateway, DNS server.

3. Getting Familiar with Your Phone 3.1. Hardware Components Preview



COMPONENT	DESCRIPTION		
LCD screen	To Presentation all information about the date& time, accounts, so keys, messages, calls and other some information.		
Power Indicator LED	To indicate the power status		
Line keys	 The phone supported up to 2 accounts Steady green: idle interface, during a call. Blink red: a call incoming 		
Soft keys	Labels automatically to identity their context-sensitive features.		
Navigation keys	 OK Up arrow key: to move up of the selection shows on the screen. Right arrow key: to move right of the selection shows on the screen. 		

	 Left arrow key: To move left of the selection shows on the screen. Down arrow key: To move down of the selection shows on the screen. Cancel: to cancel the information or call on the screen, retirn to idle screen
Mute key	Mute the voice during the call (green light).
Conference	To place a conference call
Speaker	Press this button to place a call in hands-free mode.
Redial	To dial the previous dialed number. To act as send key.
Volume	To increase/decrease the volume.
Hold	To hold or to resume a call during a conversation.
Information	To show the accounts status and some other relevant information.
Transfer	T o transfer a call to a third party. To enable or disable Forward feature during the idle page.
Alphanumeric keypad	To enter the phone numbers, letters and so on.
Message	To indicator the New message, and press to read.

3.2. Icon preview

ICON	DESCRIPTION
	Network
6	Registered succeed
×	Unregistered
•••	Speakerphone mode
	Handset mode
	Headset mode
	Voice messages
	Text message
₩.	Mute
—	Do Not Disturb
u ×	Volume Off
0	Hold
Č×.	Missed calls
C.	Dialed Calls
Ľ	Received calls
U	Forward calls

Conference
Keypad locked
Keypad unlocked

Description of the behavior of LEDs on the phone

Power LED

STATUS LED	OPIS
Blinking green	Ringing or missed call
Steady green	Power on
Off	Power down

Account LED

STATUS LED	OPIS
Steady green	Talking or off the hook
Mrugająca czerwona	Ringing
Off	Idle

BLF LED

STATUS LED	OPIS
Steady green	Idle line
Steady red	Monitored line during a call or ringing
Blinking red	Monitored line during ringing
Off	Uknown status

4. Phone Configuration

4.1. Basic Features Configuration

This part will mainly introduce the basic configuration as the Time, Language, and Volume by the help of the LCD display.



Advanced features require knowing the password for configuration (default: admin)

Advanced phone features can be configured using a web browser. This requires knowledge of the phone IP address.

To get the IP address: Press $Menu \rightarrow Status \rightarrow Information$ (phone defaulted as a DHCP client)

Information				
1. Model: UC804				
2. IP: 192.168.0.155				
3. MAC:00:1f:c1:1a:91:d6				
4. Firmware:1.0.3.59(20 i				
Back				

Input the IP Address in the web browser.

Enter user name: admin(default) and password: admin(deafult).

Login successfully.

			logout
SUCAR	Home Account	Network Function Keys Setting D	Directory Management
itus	Version (?)		NOTE
	Product Model	VPS-804P	
	Firmware Version	BOOT1.0.3.34(2014-03-2114:28:00) IMG1.0.3.66(2014-10-2015:35:00) DSP6.1.6(Patch 1.0.0)	Version: It shows product type and the version of firmware.
			Account Status:
	Account Status		It shows the registered status of accounts.
	Account1	Register Failed	
	Account2	Disabled	Network:
	Account3	Disabled	It shows the information of WAN por and LAN ports.
	Network		System Up Time:
	WAN Port Type	DHCP	It shows the running time after
	WAN IP Address	192.168.16.51	device power up.
	Subnet Mask	255.255.0.0	Restart:
	Gateway	192.168.0.1	This button will restart the voip
	Primary DNS	8.8.8.8	approation
	Secondary DNS	8.8.8.8	
	MAC Address	b0:b3:2b:00:60:1f	
	Device Type	Bridge	

4.2. Administrator Password

The password is mainly used for login the web interface or set the advanced settings through phone interface. And the default password of the administrator is: **admin**

To change to password via Phone Interface

 $\label{eq:press} Press \quad Menu \rightarrow Setting \rightarrow Advanced \quad settings \rightarrow password(default \quad admin) \rightarrow Phone \\ Setting \rightarrow Set \ Password \\ \end{cases}$

Enter the current PWD (password), new password and confirm the new password.

Press save soft key or $\textcircled{}^{()}$ to save the new password.



To change to password via Web Interface

Management → Password

Fill the value: *Current password, new password, confirm the password* then click *Save* to save the configuration.

				<u>logout</u>
S}SIICAU	Home Acc	ount Network	Function Keys Setti	ng Directory Management
Password	User Type	admin	¥	NOTE
Upgrade	Current Password		(Max length 26)	Password:
Auto Provision	New Password		(Max length 26)	If you login as an administrator,you
Configuration	Confirm Password		(Max length 26)	can modity admin's password nere.
TLS Certs				TLS Certs:
Tools				you can import TLS certificate file here.
Restart	S	SaveSet	Cancel	
Reboot				

4.3. Language

The default phone interface language is English (both LCD Menu and WEB interface).

To change the language via Phone interface

Press: *Menu*→Settings→Basic Settings→Language.

Press: Save soft key to save the configuration.



To change the language via Web Interface

Select: Setting → Preference → Web Language

Select the necessary one.

Press : Save, to save the configuration.

Note:

All languages may not be available for selection. The available languages depend on the language packs currently loaded to the IP phone.

4.4. Time and Date

The time and date show on the idle page, and it can be set and change by SNTP server automatically or manual setting.

To configure the time and Date by SNTP setting

Press: *Menu* \rightarrow *Setting* \rightarrow *Basic setting* \rightarrow *Time* & *Date* \rightarrow *SNTP Settings*

Press (>) or (<), or info soft key to change the Time zone.

NTP server

Fill the NTP server1, NTP server2 (default: tempus1.gum.gov.pl).

Daylight Saving.

The choice: Auto, Daylight Saving Time on and off.

Press $\textcircled{}^{\textcircled{}}$ or Save soft key to save the configuration.



To configure time and date manually

To press : $Menu \rightarrow Setting \rightarrow Basic \ setting \rightarrow Time \ \& \ Date \rightarrow Manual \ Settings$

Press > and < or change the right time, or you can input the right time.

Press Or Save soft key to save the configuration.

To configure the Time & Date Format

To press: Menu \rightarrow Setting \rightarrow Basic setting \rightarrow Time & Date Format

Press > and < or press Info to change between 12 Hour or 24 Hour.

Press > and < or press Info to change among Y-M-D(year-month-day), M-D-Y(month-day-year), D-M-Y(day-month-year).

Press Or Save soft key to save the configuration.

To configure the DHCP time

To press: $Menu \rightarrow Setting \rightarrow Basic \ setting \rightarrow DHCP \ time$

Press > and < or press Info to change between Disable and Enable.

Press $\textcircled{}^{\swarrow}$ or Save soft key to save the configuration.

To configure the Time and Date by web interface

Select: Network \rightarrow Advanced \rightarrow NTP Server

Set IP address NTP server (default: tempus1.gum.gov.pl).

			<u>logout</u>
\$} <i>sucan</i>	Home Account Ne	etwork Function Keys Setting	Directory Management
Basic Advanced	LLDP Active Decide Integral	Disable v	NOTE QoS:
	Qos Set	120 (15~3600s)	When the network capacity is insufficient,QoS could provide priority to users by setting the value.
	Layer 3 Qos Layer 2 Qos	48 7	NTP Server The server which is used to synchronize the clock of the phone.
	Layer 2 Qos Data VLAN Tag	802.1p priority value 0	
	NTP Server	tempus1 gum gay pl	
	Allow DHCP Option 42 To Override NTP Server:	No O Yes	

To change the Time Zone and Date Display Format via web interface

Select: Setting \rightarrow Preference \rightarrow Time Zone

Select the necessary one. Press Save to save the configuration.

			logout
\$} <i>s</i> licar	Home Account	Network Function Keys Setting Directory	Management
Preference	Web Language	English v 🕐 NOT	E
Features	Time Zone	+1 Serbia(Belgrade) 🗸 🕐	
Tones	DHCP Time Daylight Saving Time	No OYes 7 Choose	Zone: the time zone you are in.
SMS	Time Format	● 24 Hour ○ 12 Hour Scree	nSaver Photo:
Action URL Softkey Layout	Date Display Format	 Year - Month - Day Wonth - Day - Year Day - Month - Year 	only upload screen photos t of '.bmp' and '.jpg'.

Note:

If the IP Phone cannot obtain the time and date from the Simple Network Time Protocol (SNTP) server, please contact your system administrator for more information.

4.5. Screen Saver

There are two types of screen saver:Backlight off, Time & Logo .

To enable screen saver via Phone

To press: $Menu \rightarrow Display \rightarrow Screensaver...$

Press > and < or press Switch soft key to Choose the Time-out as 1 min or 2/5/10/30 minute.

Enter the Screensaver Type to choose one of the type: Backlight off, Time & Logo.

Press or Save soft key to save the selected configuration.

To disable screen saver via Phone

To press: $Menu \rightarrow Display \rightarrow Screensaver...$

Press > and < or press Switch soft key to Choose the Time-out as off.

Press V or Save soft key to save the selected configuration.

To Enable screen saver via Webpage

Select: Setting \rightarrow Preference

To choose the Time-out as 1 min or 2/5/10/30 minute.

To select the screen Type as Backlight off or Time & Logo.

Click Saveset to save the configuration.

To Disable screen saver via Webpage

Select: Setting \rightarrow Preference To choose the Time-out as Off Click Saveset to save the configuration.

4.6. Ring Tone

You can adjust the type and volume of the ring tone(8 types).

Change the ringing tone via phone interface

To press : Press Menu \rightarrow Setting \rightarrow Basic Settings \rightarrow Ring Tone.

Press \bigcirc and \bigcirc to select the aimed one.

Press Or Save soft key to save the configuration.



Change the ringing tone via WEB interface

Selekt: Account \rightarrow Advanced \rightarrow Account Ring Tone

Make Your choice

Click to Save to save the configuration.

			logout
\$} <i>s</i> lican	Home Account N	etwork Function Keys Setting Direc	ctory Management
Preference Features Tones SMS Action URL	Web Language Time Zone DHCP Time Daylight Saving Time Time Format Date Display Format	English	NOTE Time Zone: Choose the time zone you are in. ScreenSaver Photo: You can only upload screen photos in format of '.bmp' and '.jpg'.
Softkey Layout	Keypad DTMF Tone	Month - Day - Year Day - Month - Year On Off 7	
	MIC Volume Amplification Backlight Time Screen Time Out	OdB default V 0 ? off V	
	Text Logo ScreenSaver Type Ring Tones	time & logo v Ring2 bin v	
	NO Key Entry Timeout(seconds) Dial-now Time-out (seconds)	0 0 0 0	

Assign a ring tone to a contact on your phone

Press Directory.

Select the target contact.

Press Detail soft key to edit the contact.

 $\operatorname{Press}^{}$ and $\overset{\frown}{\overset{\frown}}$ to select the wanted Ring Tone for the contact

Press Save soft key to save the contact.

4.7. Ring volume

You can adjust the volume for the phone by the volume keys: \bigcirc and \bigcirc .

To adjust the Ring tone volume

Option 1: To press \bigcirc and \bigcirc on the idle page

2013	-01-01			
01:01:00 am <mark>☎</mark> Emm				
Logs	DIR	DNE) Menu	

Option 2: To press \bigcirc and \bigcirc during the call is ringing.

To adjust the handset volume

To press \bigcirc and \bigcirc during a call in handset mode.

To press \bigcirc and \bigcirc during a call in speaker mode.

To adjust the headset volume

To press \bigcirc and \bigcirc during a call in headset mode.

4.8. Directory – edit contact list

In the directory, you can add or delete your friends, business partner or anyone others' phone number so you will not forget their number. Or put some anonymous phone No. in the blacklist to prevent from being disturbed.

To add contacts manually

Press: $Menu \rightarrow DIR$ Press: Add soft key



Enter the necessary information as name, phone number...

Press Save soft key or to add the contacts successfully.

To add contacts from Call Log

Press: History soft key or press $Menu \rightarrow Logs \rightarrow Local \ history$

Press \bigcirc and \bigcirc to select the targeted one. (Press \bigcirc and \bigcirc switched among the

All calls, Dialed calls, Received calls, Missed Calls and Forward Calls).

Edit the necessary information as Name, Phone number...

Press Save soft key or to add the contacts successfully.

To add contacts via WEB interface

Click: Directory.

Enter the name, number and some other information.

Press Add and next Save button.

	Home Account Net	twork Function Keys Setting Dire	ectory Management
Directory	BlackList	Hangup	NOTE
y Index	C Display Name Office Number	Mobile Other Account All v	Add Contact/Blacklist Fill out the contact information.User shouldn't leave contact name blank.
			Delete Contact/Blacklist
			Select the contact you want to delete in the grid, and then press the button Delect to confirm.
			Move to Contact/blacklist
			Choose the contacts you want to move in the grid, and press the button move to Contact/Blacklist to move it.
	Save	Delete Move to Contact/blacklist	Import
		_	Browse the file in XML format.
Contac	t	Import Local Contacts	Export
Name		Przeglądaj_ Nie wybrano pliku.	Click Export button and create a file with whose name you prefer to
Office N	umber	Import XML Export XML	export.
Mobile N	lumber	Przeglądaj_ Nie wybrano pliku.	
Other No	Imber	Import Csv Export Csv Show Title	
Ring	Default	7	
Group	Not In Group	GroupInfo	
Photo	Auto	Ring Auto V	

To add blacklist manually

Press: $Menu \rightarrow Directory \rightarrow Blacklist$

Enter the necessary information as Name, Phone number...

Press save soft key or to add the contacts successfully.

Import/export contact list

The phone support export and import file with a list of contacts in both the CSV and XML.

File format in XML should look like this:

Import/export XML file via WEB interface

Press: Directory → Contact → Import/export XML

Choose file to upload or download.

Import/export CSV file via WEB interface

Press: Directory→Contact→Import/export CSV

Choose file to upload or download.

Contact		Import Local Contacts
Name		Wybierz plik Nie wybrano pliku
Office Number		Import XML Export XML
Mobile Number		Wybiorz plik Nie wybrane pliku
Other Number		
Account	Auto 🔻	Import Csv Export Csv 🖤 Show Title
Ring	Default 🔻	GroupInfo
Group	Not In Group 🔻	Group
Photo	Auto 🔻	Ring Auto v
Add Edit	Search	Add Edit Delete All

Phonebook we can also download a dedicated server. Just specify the path (URL) and configure it via WEB interface.

Press: *Directory* \rightarrow *Remote phonebook* and fill the URL path for example:

http://192.168.0.254/phonebook/Phonebook.xml

4.9. Call log

The phone stores the call history divided by:

- all calls
- missed calls
- incomin calls
- outgoing calls
- forwarded calls

The maximum capacity of the call log is 100 entries.

Call history via phone interface

Press: *History*→*All calls/incoming/outgoing*

Call history via WEB interface

Choose: Directory→Call History→Dilaed List/Missed List...

4.10. Network Configuration

The phone on the housing has two network ports: Internet and PC. To connect with Internet we have to use Intrnet. Additional PC port is used as a switch to connect other devices like PC, laptop.

4.10.1. Configuration Internet port

The phone is configured as a DHCP client and after connect to network automatically get the IP address assigned by a DHCP server.

To check IP address from phone interface press: *Menu*→*Status*→*Information*

In the absence of a DHCP server, you must configure a static IP address, subnet mask, gateway, and DNS server. After the above information, please contact the network administrator.

From phone LCD menu:

 $\label{eq:press: Menu \rightarrow Settings \rightarrow Advanced \rightarrow Network \rightarrow WAN \ \textit{Port} \rightarrow \textit{Static mode}.$

From WEB interface:

Select: Network→Basic→Staic address.

You can also get the IP address logging phone to the network using PPPoE protocol. In this case, enter your account name (login) and password you received from your network administrator.

From phone LCD menu:

 $Press: \textit{Menu} \rightarrow \textit{Settings} \rightarrow \textit{Advanced} \rightarrow \textit{Network} \rightarrow \textit{WAN Port} \rightarrow \textit{PPPoE mode}.$

From WEB interface:

Select: Network \rightarrow *Basic* \rightarrow *PPPoE.*

Configure provider details for PPPoE service as a: user ID and password.

4.10.2. Configuration PC port

Additional port to connect other devices as PC or laptop. It works in the two modes:

- as bridge
- as router

In bridge mode, this port behaves as a switch. Connected devices can access to LAN. In router mode allows You to connect network devices for which the phone acts as a router assigning tchem IP address according to the configuration.

5. Configuration VoIP Account

For the realization of voice calls, it is necessary to configure the SIP accounts. After all the data for account setup, ask your network administrator or service provider VoIP.

To properly configure the SIP account as follows:

- set VoIP server address
- backup SIP server (optional)
- username (User ID)
- register name (Authenticate ID)
- password
- account name (on LCD display)

			logout
Home /	Account Network Function	Keys Setting	Directory Management
asic Account	Account 1		NOTE
Account Status	Register Failed		
* Account Active	○ No ● Yes		* fields must be filled and require a phone restart
* Primary SIP Server	10.0.0.60	3	Pasic:
Failover SIP Server		3	The Basic Parameters set for
Second Failover Sip	Server	3	adminstrator
Prefer Primary SIP S	erver 🔍 No 🔿 Yes 🕐		Codecs:
Outbound Proxy		3	Choose the codecs you want to us
Backup Outbound P	гоху	3	
* SIP Transport		· •	Advanced:
NAT Traversal	○No	alive OSTUN	adminstrator.
Label	342	3	
* SIP User ID	620	3	
* Authenticate ID	620	3	
* Authenticate Passw	ord	3	
Name		3	

5.1. Basic configuration VoIP Account

Basic configuration VoIP account requires knowledge of VoIP account login details,

such as: SIP server address, user name, service, service user password.

From phone LCD menu:

Press: $Menu \rightarrow Settings \rightarrow Advanced \rightarrow Account.$

Fill SIP server name, username, password, register name.

From WEB interface:

Select: *Account→Basic*

Fill SIP server name, username, password, register name.

```
Note:
```

After data for your VoIP account login, please contact your service provider or network administrator.

5.2. Advanced Configuration VolP Account

From the web interface you can configure advanced settings for SIP account:

• **Outband proxy:** Proxy SIP sever address provided that the VoIP service requires its settings.

- Active account: enable/disable the account registration.
- SIP Transport: type of transport protocol UDP(default), TCP, TLS.
- **Type DNS:** DNS Server search service.
- Empire of registration: expiration time of registration to the SIP server, if the register expire time of the phone is different from the time required by the server, it will automatically changed for the time recommended by the server and register again.
- **Outgoing call without registering::** permit or prohibit making calls without registration to the SIP server.
- Local SIP port: port for VoIP call signaling process the default 5060.
- Local RTP port: port for used to transfer audio frames 5004 by default.
- **Codecs:** settings for compression codecs in order to use of acoustic signals negotiated in the call signaling.
- DTMF mode: specifies how to send DTMF(in the band, SIP INFO, RFC 2833)
- **SRTP mode:** activation of RTP voice encryption frames.
- VAD(Voice Activity Detection): detection of silence during a call.
- Ringing tone for an account: assign one of 8 ringtones for your account.
- **Ring time:** phone ringing time settings for incoming calls.
- Sign # as a send: use the # key as a sign of the end of the dial.
- **Subscribe for MWI message:** MWI settings for sending messages informing about the new voice message.
- Anonymous calls: reservation number for outgoing calls CLIP.
- Anonymous Call Rejection: enable anonymous call rejection.
- Auto answer: allows to automatically answer incoming calls.
- **Rejection of the return code:** reject code defines a call in the SIP signaling.
- **Direct interception code:** service code interception defined by the administrator of the SIP server.
- Interception group code: service code interception in the group defined by the administrator of the SIP server.

NOTE:

These data are dependent on the specific VoIP server settings. Their parameters are set by the service provider or network administrator.

5.3. Codecs

Settings on the order of use preferred codecs compress acoustic signals negotiated in the call signaling.

Phone suports the following codecs:

- G.711a(PCMA)
- G.711u(PCMU)
- G.729A/B
- G.726
- G.723
- G.722
- GSM
- iLBC

6. Basic Call Features

6.1. Place a Call

There are three ways to dialing a call: Handset, Headset and Hands-free speakerphone.

To place a call by Handset

Pick up the handset, or press a line key and dial the necessary number.

Press end softkey, then the call is sending.

To place a call by Headset:

Press the (light is Green),

Enter the desired number.

Press er or press the *Send* softkey, then the call is sending.

Placing a call by hands-free speakerphone

Press the , or press the Line key, then you can hear the dial tone.

Enter the number.

Press er press the Send soft key, then the call is sending.

To place a call by call history or Directory

Press the History soft key or Menu \rightarrow History /Directory

Press \bigcirc and \bigcirc to select the targeted one.

Press @ or Send soft key to make the call.

NOTE:

The key is set to be a send key. You can set the * key as send key or set some other to be as send keys. For more information, refer to the Key as Send on page During the call, you can also change among Headset, Handset or Free-speaker mode.

6.2. End a Call

There are three ways to end a call:

To end a call by Handset

Naciśnij przycisk "Anuluj" lub odłóż słuchawkę.

To end a call under Headset Mode

Press Cancel soft key or press (Q.).

To end a call under hands-free Speakerphone Mode

Press the *Cancel* soft key or press .

NOTE:

During the conference, to end the call is same as mentioned above.

6.3. Receive a Call

There are three ways to receive a call when the phone is ringing:

To receive a call by handset

Pick up the handset the conversation is built.

To receive a call by headset

Press () and now the conversation is built.

To receive a call by hands-free speaker

Option 1: Press 🔊 directly.

Option 2: Press Answer soft key.

Option 3: Press the Line key (flashes red).

Moreover, some other action can be done by soft key when the call is coming.

To press the *Reject* soft key to reject the call. Or press \bigotimes to reject the current call.

To press Forward to forward to another phone.

To press *Silence* soft key, and then the call will keep silent, no ring tone display.

6.4. Redial a Call

To redial the last placed call from the IP Phone

Press (directly when LCD is on the idle interface.

6.5. Auto answer

Enable auto answer feature, you will answer all incoming call automatically.

To enable Auto Answer via Webpage:

To Click: Account \rightarrow Advanced

To choose Yes for the Auto Answer.

To click Saveset to save the configuration.

6.6. Call Hold

When use hold feature, the Hold icon will show on the display.

To make a call on hold during three modes:

To hold a call under handset mode:

Press Hold soft key to hold the current call.



Press Resume soft key to resume the call on hold.

To hold a call under headset mode:

Press Hold soft key or 🖭 to hold the current call.

Press Resume soft key or 🖭 to resume the call on hold.

To hold a call during the speaker mode:

Press Hold soft key or 🖭 to hold the current call.

Press Resume soft key or 🖭 to resume the call on hold.

6.7. Call Transfer

This phone supports blind, attended and Semi-Attended Transfer.

Blind Transfer

Press Transfer soft key or Conversation, the call is on hold now.

Enter the number that transfers to.

Press *Transfer* soft key or (), and now the blind transfer completed.

NOTE:

The "Enable Call Feature" must be configured to "Yes" to enable this feature.

Attended Transfer

When you use this feature, you can:

Press *Transfer* soft key or C during the conversation, the call is on hold now.

Enter the number that transfer to, and press the send soft key or

Start the second	conversation,	press the	Transfer o	or	soft key,	then,	transfer
completed.							

NOTE:

To transfer calls across SIP domains, SIP service providers must support transfer across SIP domains. Blind transfer will usually use the primary account SIP profile. To use this function, should at least one line key set as Auto.

Semi-Attended Transfer

Press the *Transfer* soft key or C during the conversation, the call is on hold now.

Enter the number transfer to, and then press *m*, then you can here the ring tone.

Press the *Transfer* soft key or (Sec., and now the Semi-attended transfer completed.

BLF Transfer

Set a Programmable Key or line key is set as BLF.

For how to set BLF, please refer to BLF

Press the *Transfer* soft key or C during the conversation, the call is on hold now.

Press BLF key then realize blind, attended and Semi-Attended Transfer. (Webpage--Setting--Features)

M				logout
Sastican	Home Account Netw	vork Function Keys Setti	ng Direc	tory Management
Preference	Forward:		3	NOTE
Features	📴 Do Not Disturb			Forward
Tones	HotLine			This feature allows you to forward an
SMS	Transfer Settings			incoming call to another phone number.
Action URI	Blind Transfer On Hook	● On Off		
	Semi-Attended Transfer	● On Off		Target:
Soffkey Layout	Attended Transfer On Hook	● On Off		calls will be forwarded.
	Transfer Mode via DSSkey	Attended Transfer 🗸		
	Hold Transfer On Hook	Attended Transfer Blind Transfer New Call		On Code: The code that will be sent to PBX
	🕂 Call Pickup			when it is switched on.
	+ Phone Lock			Off Code:
	- Call Waiting			The code that will be sent to PBX when it is switched Off.
	Alort Ping			
	Alert King			
	SaveSet	Cancel		
	Curtobi	Canton		

6.8. Call Conference

This IP Phone supports up to 5-way conference.

To set Conference

Assuming that call party A and B are in conversation. A wants to bring C, D and E in

a conference:

- 1. A press line 2 key, the call is placed on hold.
- 2. A enter the number of C and then press send soft key or #.
- 3. C answering the call.
- 4. A press *Conference* soft key or B, then A, B and C are now in a conference (and now this is **3-way conference**).
- 5 A press line 3 key, the current 3-way conference is placed on hold.
- 6 A enter the number of D and then press send soft key or #.
- 7 D answering the call.
- 8 A press the conference soft key, then A, B, C and D are now in a conference.(and now this is **4-way conference**)
- 9 A press line 4 key, the call is placed on hold.

- 10 A enter the number of E and then press send soft key or .
- 11 E answering the call.
- 12 A press the *Conference* soft key or (1), then A, B, C, D and E are now the **5**-way conference is built.
- 13 A end the call, the conference is finished.



NOTE:

If C does not answer the call, A can back to continue the conversation with B; Once A hangs up the call, the conference is ended, while if B or C drops the call, A&C or A&B conversation continues. The conference feature is not available on all servers. For more information, contact your system administrator.

To realize the 5 way conference, the line should be all available.

Press ((1)) to get all parties information.

6.9. Call Forward

This phone supports static forward (always forward, busy forward and no answer forward) and dynamic forward

To configure always forward

With this feature, all incoming calls will forward immediately to configured number.

Press: Menu \rightarrow Features \rightarrow Call forward \rightarrow Always Forward.

Press > and < or press *info* soft key to select the enable choice.



Enter the forward to number and on code (optional), off code (optional).

Press or Save soft key to save the configuration.

To configure busy forward

With this feature, the incoming calls are immediately forwarded if the phone is busy.

Press: Menu \rightarrow Features \rightarrow Call forward \rightarrow Busy forward.

Press > and < or press info soft key to select the enable choice.

Enter the forward to number and on code (optional), off code (optional).

Press or Save soft key to save the configuration.

To configure no answer forward

No Answer Forward: Incoming calls are forwarded if not answered after some time. Press: *Menu* \rightarrow *Features* \rightarrow *Call forward* \rightarrow *No answer forward*.

Press (>) and (<) or press *info* soft key to select the enable choice.

Enter the forward to number and on code (optional), off code (optional).

Press Or Save soft key to save the configuration.

When the Forward feature is enabled, the Forward Icon will display on Top of the LCD.

To configure Forward via Web Interface

Select :Setting \rightarrow Features \rightarrow *Forward*.

Click on for the Always/Busy/No Answer.

Fill the Forward to Number.

Click SaveSet to save the configuration

	Home Account Netv	vork Functio	on Keys	Setting Dire	ectory Management
Preference	Forward:			۲	NOTE
Features	Always	\bigcirc On	● Off		
Topoo	Target			3	Forward::
Tones	On Code			3	incoming call to another phone
//S	Off Code			2	number.
n URL	Busy	Oon	Off		Target:
ayout	Target			2	The number to which the incoming
	On Code			0	calls will be forwarded.
	Off Code			0	On Code:
	No Answer	Oon	• Off		The code that will be sent to PBX
	After Ring Time(seconds)			0	when it is switched on.
	Tarnet			0	Off Code:
	On Code				The code that will be sent to PBX
	On Code				when it is switched Oil.

Forward an incoming call during the ringing.

When the phone is ringing, press Forward soft key.

Enter the forward number.

Press or press the send soft key, then the call is forwarded.

NOTE:

If the Programmable Key or line key is set as BLF, when an incoming call ringing, press this BLF key directly to realize the dynamic forward.

6.10. Call Return - REDIAL

This feature allows you to dial the last phone call you received.

To configure the Call Return via phone interface

Press: Menu \rightarrow Features \rightarrow Function Keys \rightarrow Line or Memory keys as Function Keys

 \rightarrow *Memory key1*(for example)

Select the wanted Line or Memory key.

Press > and |< or press | | key to select the Key Event in the type field.

Press > and < or press $\boxed{1}$ key to select the Call Return.

Press or Save soft key to save the configuration.

To configure the Call Return via Web interface

Select: Function keys \rightarrow Line or Memory key.

Select the desired Line or Memory key and select Call Return in the Type.

Click the SaveSet to save the configuration.

6.11. Anonymous Call

Just enable the anonymous call feature, you can set your ID do not appear on the other phone that you called.

To configure anonymous call

Press: Menu \rightarrow Features \rightarrow Anonymous Call

Press > and < or press \square key to change the Line ID

Press > and < or press \square key to select the enable choice in Anonymous Call filed.

Enter the call on code (optional), call off code (optional).

Press Or Save soft key to save the configuration.

Reject anonymous call

If you do not want to be disturb by anonymous calls, you can set the reject anonymous call features, so you will not hear the unknown calls

Press: *Menu* → *Features* → *Anonymous* Call

Press > and < or press \square key to change the Line ID.

Press > and < or press \square key to select the enable choice in Rejection filed.

Press Or Save soft key to save the configuration.

6.12. Call Mute

When you use the Mute feature, the other parties will not hear your voice while you can hear their voice. Call mute applies to all modes (handset, headset, and speakerphone).

Press the mute key _____, then the mute key glows green, and the LCD display Mute Icon.

() 304	:Talking		
	00:00:5	51	6 304
	Joe		🔂 Eric
	303	Ā	
TRNF			Cancel

To disable the mute function, press () again.

6.13. Do Not Disturb DND

When you use the DND feature, the phone is to reject all incoming calls automatically and you can see the DND icon shown on the Top of the LCD on idle page.

To enable DND feature

Press the *DND* soft key when the phone is idle, and then DND iconshown on the LCD.

To disable DND feature

Press the DND soft key again, and then there is no DND icon on the LCD.

6.14. Key as Send

To speed up the process of dialing can be attribute to the end of dialing key (default #).

To configure Key as Send

Press: *Menu* \rightarrow *Features* \rightarrow *Key as Send.*

Press > and < or press \bigcirc key to select the enable choice.

Press ^(W) or Save soft key to save the configuration

6.15. No Key Entry Timeout

No Key Entry Timeout means that when you entry the number, it will dial out automatically after some time when you stop enter the number.

To configure No Key Entry Timeout

From WEB interface select: Setting \rightarrow Preference \rightarrow NO Key Entry Timeout(seconds) Fill the blank of No Key Entry Timeout: for example 5(seconds).(0 means never timeout, you should press the send key the dial out the number. Click the SaveSet to save the configuration.

SUCAR	Home Account N	etwork Function Keys Setting Dire	ctory Management
Preference	Web Language	English v	NOTE
Features	Time Zone	+1 Serbia(Belgrade) 🗸 🕐	
Topos	DHCP Time	● No O Yes 🕜	Time Zone:
Tones	Daylight Saving Time	● No O Yes 🕜	Choose the time zone you are in
SMS	Time Format	● 24 Hour 0 12 Hour	ScreenSaver Photo:
Action URL	Date Display Format	O Year - Month - Day 🕐	You can only upload screen pho
oftkey Layout		O Month - Day - Year	in format of ".omp" and ".jpg".
		Day - Month - Year	
	Keypad DTMF Tone	● On ○ Off 🕜	
	MIC Volume Amplification	0dB default 🗸	
	Backlight Time	0	
	Screen Time Out	off v	
	Text Logo	2	
	ScreenSaver Type	time & logo 🗸	
	Ring Tones	Ring2.bin V	
	NO Key Entry Timeout(seconds)	0	
	Dial-now Time-out (seconds)	0 0	

6.16. Hot Line

The hot line gives us the opportunity to direct dialing immediately after hook off programmed number with which you want to connect to.

To configure Hot Line from phone interface

Press: $Menu \rightarrow Features \rightarrow Hot Line$

Enter the number and delay time (as present, we support off hook auto dial).

Press $\textcircled{}^{\textcircled{}}$ or Save soft key to save the configuration.

To configure Hotline auto dial via Web Interface

Select: Setting \rightarrow Features.

Fill the number in the Hotline Number and Hotline Time-out.

Click SaveSet to save the configuration.

Delete the number and save the configuration, and then now the Hotline auto dial is cancelled.

SUCHIC	Home Account Network Function Keys	Setting Dire	ctory Management
Preference	Forward:	0	NOTE
eatures	Do Not Disturb		Forward::
nes	HotLine Hotline Number		This feature allows you to forward a incoming call to another phone number
on URL	Hotline Time-out(seconds)(0~180s)		
y Layout	Transfer Settings		The number to which the incoming calls will be forwarded.
	Call Pickup Phone Lock		On Code:
	Call Waiting		The code that will be sent to PBX when it is swtiched On.
	Alert Ring		Off Code:
	SaveSet Cancel		The code that will be sent to PBX when it is switched Off.

7. Advanced Features

7.1. Line

To line keys can assign SIP accounts.

To configure SIP account via WEB interface:

Choose: Function Keys→Line Keys

Choose account number then press Save.

₿slican	H	ome Account	t Netwo	rk Fund	tion Keys	Setting Di	rectory Management	<u>logout</u>
Line Key Programmable Key EXP KEY	Line Key1 Key2 Key3 Key4	Type Line ▼ Line ▼ Line ▼ N/A ▼ SaveSe	Value t	Label	Account 1 V Account 2 V Auto V Account 1 V Restart	Pickup Code	NOTE	

7.2. Voice Message

This phone supports Voicemail MWI, and when there is message, the message will will light green. Moreover, when pick up the handset, or press speaker key, you will hear some quick busy tone.



To configure the voice mail message feature via phone

Press: *Menu→Messages→Voice Mail→Set Voice Mail*.

Enter the SIP account number and set voice miał extension numer.

Press V or Save soft key to save the configuration.

To configure the voice mail message feature via WEB interface

To set function keys as a voice mail key choose: Function Keys \rightarrow Line Keys \rightarrow Voice mail. Set voice mail number and choose proper SIP account. Then press: Save.

NOTE:

This feature is not available on all servers. For more information, contact your system administrator.

7.3. Intercom

When use the intercom feature, you can quickly get access connect to the configured one.

To configure intercom feature via phone interface

 $\label{eq:press: Menu} \textbf{Press: Menu} \rightarrow \textbf{Features} \rightarrow \textbf{Function Keys} \rightarrow \textbf{Line or Memory keys as Function Keys}$

 \rightarrow Memory key1(for example)

Press > and < or press \square key to select the intercom in the type field.

Enter the targeted Number.

Press () or Save soft key to save the configuration

Then the selected Line or Memory key will work as intercom.

To configure Intercom feature via Web Interface

Select: Function keys \rightarrow Line or Memory key.

Select the wanted Line or Memory key.

Enter the desired phone number in the Value field.

Select the Account ID

Click the *SaveSet* to save the configuration.

(h arran a								logout
Sican	Hon	ne Account	Networ	k Funct	ion Keys	;	Setting Dire	ectory Management
Line Key	Line Key	Туре	Value	Label	Accour	ıt	Pickup Code	NOTE
Programmable Key	Key1	Intercom v	102		Auto	¥		
EXP KEY	Key2	Line v			Auto	۲		
	Key3	Line v			Auto	4		
	Key4	Record v		REC	Auto	\vee		
		SaveSet			Restart			

NOTE:

This feature is not available on all servers. For more information, contact your system administrator.

7.4. Speed Dial

With this feature, you can dial one directory by press the definite key.

To configure Speed Dial feature via phone interface

Press: Menu → Features → Function Keys → Line or Memory keys as Function Keys

 \rightarrow *Memory key1*(for example)

Press > and < or press key to select the Speed Dial in the type field.

Enter the targeted Number.

 $\operatorname{Press}^{\textcircled{}}$ or Save soft key to save the configuration

Then the selected Line or Memory key will work as Speed Dial.

To configure Speed Dial feature via Web Interface

From WEB interface select: Function keys \rightarrow Line or Memory key.

Select the wanted Line or Memory key and set as Speed Dial.

Enter the desired phone number in the Value field.

Select the Account ID.

Click the SaveSet to save the configuration.

7.5. Direct Pickup

With this feature, you can pick up the set line when it ringing.

To configure Direct Pickup feature via phone interface

Press: Menu \rightarrow Features \rightarrow Function Keys \rightarrow Line or Memory keys as Function Keys \rightarrow Memory key1(for example)

Press > and < or press $\boxed{1}$ key to select the Speed Dial in the type field.

Enter the value.

Press () or Save soft key to save the configuration

Then the selected Line or Memory key will work as Direct Pickup.

To configure Direct Pickup feature via Web Interface

From WEB interface select: Function keys \rightarrow Line or Memory key.

Select the wanted Line or Memory key and set as Direct Pickup.

Enter the pickup code and followed the desired phone number in the Value field.

Select the Account ID.

Click the SaveSet to save the configuration.

7.6. Group Pickup

With this feature, you can pick up the specified group that you want incoming calls.

To configure the Pick up via phone interface

Press: Menu \rightarrow Features \rightarrow Function Keys \rightarrow Line or Memory keys as Function Keys \rightarrow

Memory key1(for example)

Select the wanted Line or Memory key.

Press > and < or press | | key to select the Group Pickup in the type field.

Enter pickup code and followed the desired group numer.

Press () or Save soft key to save the configuration

To configure the Pick up via WEB interface

From WEB interface select: Function Keys \rightarrow Line Key1 \rightarrow Key1.

Select the wanted Line or Memory key.

Enter pickup code and followed the desired group numer.

Click the SaveSet to save the configuration.

7.7. BLF(Busy Lamp Field)

You can use the BLF (Busy Lamp Field) feature to monitor a specific numer whether its phone is busy or free.

When the monitored line is idle, the LED is solid green. When the monitored line is ringing, the light is blinking red, press the BLF key to pick the phone up directly. When the monitored line is calling or in a conversation, the light is steady red.

Optionally, this function gives us the ability to intercept calls to the monitored number with code capture.

To configure a BLF key by phone

Press: Menu \rightarrow Features \rightarrow Function Keys \rightarrow Line or Memory keys as Function Keys \rightarrow Memory key1(for example).

Select the targeted Line or Memory key.

Press > and < or press \square key to select the BLF in the type field.

Enter the targeted Value Number.

Press (>) and (<) or press (\square) key to select the BLF in the type field.

Enter the Pickup Code.

Press () or Save soft key to save the configuration

To configure a BLF key by WEB interface

Select: Function keys \rightarrow Line or Memory key.

Select the desired Line or Memory key and select BLF in the Type.

Enter the monitored phone number in the Value field.

Select the Account ID.

Filled the Pickup code.

Click the SaveSet to save the configuration and then restart.

ыст	Hom	ne A	Account	Network Fu	unction Keys	Setting Dire	ectory Management
Line Key	Memory Key	т	јуре	Value	Account	Pickup Code	NOTE
Nemory Key	Key1	BLF	~		Account 1 V		Кеу Туре:
ogrammable Key	Key2	N/A	~		Account 1 V		The free function key Type Speed
	Key3	N/A	~		Account 1 V		Dial, DEI, Rey Event, intercont, ORE
	Key4	N/A	~		Account 1 V		BLF:
	Key5	N/A	¥		Account 1 V		BLF setting require a phone resta
			SaveSet		Restart		

7.8. Shared Line

This feature allows subscribers to share SIP lines. Moreover it also provides status monitoring of the shared line.

To configure the line key as shared line via phone interface

Press: <i>Menu→Features→Function Keys→Line</i> or Memory keys as <i>Fur</i>	nction Keys $ ightarrow$
<i>Memory key1</i> (for example)	

Select the wanted Line or Memory key.

Press > and < or press < key to select the Shared Line in the type field.

Press > and < or press \bigcirc key to select the Account ID.

Enter the Label.

Enter the Value.

Press or Save soft key to save the configuration.

To configure the line key as shared line via Web Interface

From WEB interface select: Function keys \rightarrow Line Key or Memory key.

Select the desired Line or Memory key and select Shared Line in the Type.

Enter the Value.

Enter the Label.

Select the Account ID

Click the SaveSet to save the configuration and then restart.

₿sucar	Hom	ie Account	Network Fu	nction Keys	Setting Di	rectory Management
Line Key	Memory Key	Туре	Value	Account	Pickup Code	NOTE
Memory Key	Key1	Shared Line 🗸		Account 1 🗸		Кеу Туре:
Key	Key2	N/A v		Account 1 V		The free function key Type Speed Dial.BLF.Key Event.intercom.URL.
	Key3	N/A v		Account 1 V		
	Key4	N/A v		Account 1 Y		BLF:
	Key5	N/A v		Account 1 Y		BLF setting require a phone restart
		SaveSet		Restart		

NOTE:

This feature is not available on all servers. For more information, contact your system administrator.

7.9. Call Park

With this feature, you can put a call on hold and continue the conversation from

another phone.

To configure the Call Park via phone interface

Press : $Menu \rightarrow Features \rightarrow Function Keys \rightarrow Line$ or Memory keys as Function

Keys→Memory key1

(for example)

Select the wanted Line or Memory key.

Press > and < or press Switch soft key to select the Key Event in the type field.

Press > and < or press Switch soft key to select the Call Park.

Press Or Save soft key to save the configuration.

To configure the Call Park via Web interface

From WEB interface select: Function key \rightarrow Line or Memory key.

Select the desired Line or Memory key and select Direct Pickup in the Type.

Click the SaveSet to save the configuration.

7.10. Paging

With this feature, you can call a phone directly.

To configure the paging via phone interface

Press: $Menu \rightarrow Features \rightarrow Function Keys \rightarrow Line$ or Memory keys as Function Keys $\rightarrow Memory \ key1$ (for example)

Select the wanted Line or Memory key.

Press > and < or press $\boxed{1}$ key to select the Key Event in the type field.

Press > and < or press \square key to select the Paging.

Press () or Save soft key to save the configuration

To configure the Paging via Web interface

Select: Function keys \rightarrow Line or Memory key.

Select the desired Line or Memory key and select Paging in the Type.

Click the SaveSet to save the configuration.

7.11. DTMF

This feature allows you to send directly to the DMTF signaling lines during an ongoing call.

To configure the DTMF via Phone Interface

Press: $Menu \rightarrow Features \rightarrow Function$ Keys $\rightarrow Line$ or Memory keys as Function

Keys \rightarrow Memory key1(for example)

Select the wanted Line or Memory key.

Press > and < or press $\boxed{\square}$ key to select the DTMF in the type field.

Enter the value.

To configure the DTMF via Web interface

Click Function keys \rightarrow Line or Memory key.

Select the desired Line or Memory key and select DTMF in the Type.

Fill the value.

Click the *SaveSet* to save the configuration.

7.12. Prefix

This feature allows us to define any prefix added before the dialed number.

To configure the Prefix via Phone Interface

Press: $Menu \rightarrow Features \rightarrow Function Keys \rightarrow Line$ or Memory keys as Function Keys \rightarrow Memory key1(for example)

Select the wanted Line or Memory key.

Press > and < or press \square key to select the Prefix in the type field.

Enter the value.

Press $\textcircled{}^{\lor}$ or Save soft key to save the configuration.

To configure the Prefix via Web interface

Select: *Function key* \rightarrow *Line* or Memory key.

Select the desired Line or Memory key and select Prefix in the Type.

Fill the value.

Click the SaveSet to save the configuration.

Then when you press this key, the set value is input directly.

7.13. Action URL

This feature allows us to send useful information from the phone to the server and vice versa in order to initiate certain actions on the phone.

This gives us the opportunity to observe phone status or sending such requests:

- login / logout phone
- hook on/hook off handset
- make a call
- turn on DND service

An example of the syntax of the call from the phone by account number 1 to number 523 251 111:

http://admin:password@IP-Address/Phone_ActionURL&Command=1&Number=523251111&Account=1 where:

- admin:password user name and password
- IP-Address IP address of the phone which will be place a call
- Command=1 command call
- Number called number
- Account=1 SIP ID account to call

7.14. SMS

Additional feature to send short text messages between phones.

To send message vis phone interface:

Press: Menu \rightarrow Messages \rightarrow Text Messages \rightarrow Set SMS

Type the text and set number then press Send.

To send message vis WEB interface:

Select: Setting \rightarrow SMS

Selcet Account and target Number type the message then press Send button.

7.15. Keypad Lock

Phone lock feature allows us to completely block the function or the function keys.

To enable Keypad Lock via Phone

 $Press: \textit{Menu} \rightarrow \textit{Settings} \rightarrow \textit{Advanced Setting} \rightarrow \textit{Phone Setting} \rightarrow \textit{Lock}$

Press: Or Switch to change choose lock type:All Keys Menu Key Function Key

Lock & Answer

Press Save or OK key to save the configuration.

To disable Keypad Lock via Phone

Press: *Menu*→Settings→Advanced Setting→Phone Setting→Lock.

Press: Info or Switch to change choose Off.

Press Save or OK key to save the configuration.

To enable Keypad Lock via Webpage

From WEB interface: Setting \rightarrow Preference.

To choose the Lock keys.

To fill the unlock PIN and auto lock time

Fill the Emergence Number, when the phone is Lock, only Emergency Number can be sent.

To click Saveset to save the configuration..

	Home Account Netwo	ork Function Keys Setting	Directory Management
Preference	Forward:	(7	NOTE
Features	Do Not Disturb		Forwards
BLF Settings	E HotLine		This feature allows you to forwa
Date&Time	Transfer Settings		incoming call to another phone number.
Tones	📴 Call Pickup		
SMS	Phone Lock		The number to which the incom
Action URI	Keypad Lock	All Keys 🗸	calls will be forwarded.
	Phone Unlock Pin(0~15digial)	••••	On Code:
tey Layout	Auto Lock Time-Out(15~3600s)	15	The code that will be sent to PB
	Emergency	112	when it is switched On.
	Coll Walling		Off Code:
			The code that will be sent to PB when it is switched Off.
	🚹 Alert Ring		
	E Auto Redial		

7.16. Setting LED status

Phone let us configure behavior LED status:

- Power
- Ringing
- Missed call
- Voice message

7.17. Dial plan

Dial plan is a set of the rules to select some prefixes based on digits.

Phone suport the following digits: 1,2,3,4,5,6,7,8,9,0,*,#.

Dial plan rules:

STRING	DESCRIPTION
х	any digits from 0-9;
xx+	at least 2 digit number;
٨	exclude;
,	hear dial tone;

[3-5]	any digits from 3 to 5;
[147]	any digit 1, 4 or 7;
<2=011>	Replace digit 2 with 011.

Default dial plan: {[x*]+}.

Examples:

Example 1:

{[369]11 | 1617xxxxxx} :

Allow 311, 611, 911, and any 10 digit numbers of leading digits 1617

Example 2:

{^1900x+ | <=1617>xxxxxx}:

Block any number of leading digits 1900 and add prefix 1617 for any dialed 7 digit numbers

Example 3:

{1xxx[2-9]xxxxxx | <2=011>x+}:

Allow any length of number with leading digit 2 and 10 digit-numbers of leading digit 1 and leading exchange number between 2 and 9; If leading digit is 2, replace

leading digit 2 with 011 before dialing

Example 4:

{99,x+|[*x]+}

When user dial 99, can hear dial tone again, user continue dial, dial tone will stop.

7.18. Encrypted call handling - TLS/SRTP

The phone suport encrypted call based on TLS and SRTP protocols.

To onfigure TLS and SRTP via WEB interface

Selcet: Account \rightarrow Basic \rightarrow SIP Transport \rightarrow TLS for encryption signaling.

Select: $Account \rightarrow Advanced \rightarrow SRTP Mode$ for encryption acoustic.

To correct TLS authentication is necessary to upload the trusted certificates.

To upload certificates file select: *Management→Import Trusted Certificate Files.*

Then select Custom Certificates and press SaveSet.

esucoo				<u>loc</u>
	Home Account	Network Function Keys	Setting Dire	ctory Management
Password	Index Issued TO	Issued By Expiration	Delete	NOTE
Upgrade	1 NCP Private CA	Slican NCP Apr 3 08:43:39 203	1 GMT	Trusted CA.
Auto Provision	2			you can import TLS certificate fi
Configuration	3			here.
Trusted C4	4			
Trusted CA	5			
Server CA	6			
Tools	7			
Restart	8			
Reboot	q			
	10			
			Delete	
	Import Trusted Certificate Files	Wybierz plik Nie wybrano pliku		
		Import Trusted Certificates		
	Trusted Certificates	Default Certificates		
		 Custom Certificates All Certificates 		
	SaveS	Cancel		

7.19. VLAN and QoS service

Phone support 802.1Q protocol for VLAN and QoS. This gives us the opportunity to work with virtual network(VLAN) and set priority for voice packets.

To configure VLAN/Qos via WEB interface:

Selcet: Network $\rightarrow Advanced \rightarrow QoS set$

Set proper VLAN ID and QoS priority. This information should specify the network administrator.

\$sucan	Home Account	Network Function Keys	Setting Direct	logout tory Management
Basic Advanced	LLDP Active Packed Interval Qos Set	Disable ▼ 120 (15~3600s)		NOTE QoS: When there is insufficient network capacity, QoS priority can be provided by setting the value for the user.
	Layer 3 Qos Layer 2 Qos Layer 2 Qos Data VLAN Tag	48 802.1Q/VLAN Tag 10 802.1p priority value 2 11	°	

7.20. Additional tools

The phone has a built-in diagnostic tools to analyze any problems. One of them is the ability to retrieve logs of network traffic. Second tool to analyze phone problem is system logs.

Getting network traffic from the WEB interface:

Selcet: *Management*→*Tools*→*Pcap Feature*→*Start*

After finish getting network packets select Stop and Export to download log file.

Getting system log from the WEB interface:

Selcet: *Management*→Configuration→System log

Set proper the log level from the drop-down list(DEBUG, INFO, WARNING, ERROR) and next *Download* to get log file.

8. Management – Upgrade

8.1. Start

Login window startup web interface gives us a phone the most important status such as:

- product model
- firmware Version
- account status
- network status: address IP, DNS, MAC address
- memory free
- system up time

8.2. Factory Reset

It enables us to reset all phone settings to default factory settings.

To set Factory Reset by phone interface

Press: *Menu*→*Setting*→ *Advanced Setting(default password: admin*)→*Phone Setting*→*Factory Reset*

Press OK soft key in the warning page.

To set Factory Reset via web interface

Select: $Management \rightarrow Upgrade \rightarrow Reset To Factory.$

				<u>logout</u>
§³ <i>slican</i>	Home Account	Network Function Keys	Setting Dire	ectory Management
Password	Image Version			NOTE
Upgrade	Major Version	IMG1.0.3.77(2015-01-29 11:30:00)		Image Version:
Auto Provision	Minor Version	IMG1.0.3.74(2014-12-22 18:17:00)		Show the information of the two system image version .
Configuration	Reset To Factory	Reset To Factory		
Trusted CA Server CA	ROM Firmware Upgrade	Przeglądaj_ Nie wybrano plik Upgrade		Reset To Factory : Reset all the settings of the phone to default configruations.
Tools				Restart:
Restart				one simple operation for restart the voip application.
Reboot				

8.3. Software update

The software update can be done in several ways:



- via TFTP/http/HTTPS server(specify the path).
- through auto-provisioning server HTTP/HTTPS
- by selecting the firmware image file from the specified location (local or network drive)

Upgrade firmware from image file via WEB interface:

Selcet: *Management* → *Upgrade*→*ROM Frimware Upgrade*

Selcet proper file and press Upgrade button. Phone after this action shuld be reset to factory.

To configure the TFTP/HTTP/HTTPS server via the WEB interface

Select: $Management \rightarrow Auto Provision \rightarrow Upgrade Mode(TFTP/HTTP/HTTPS).$

Then enter the address of the TFTP/HTTP/HTTPS and server access path. After this reset the phone.

			logout
Sican	Home Account	Network Function Keys Setting Dir	rectory Management
Password	• Firmware Upgrade		NOTE
Upgrade	PnP Active	○No ®Yes 🕐	Firmwara Unarada :
Auto Provision	Upgrade Mode	●TFTP ○HTTP ○FTP ○HTTPS	The detail sets about the firmware
Configuration	Firmware Server Path	192.168.16.200/fm	upgrade for the system.
T	Config Server Path		Phonebook Download:
ISTED CA	Allow DHCP Option	128	The detail sets about the phonebool
erver CA	To Override Server:	○No ●Yes	XML download.
Tools	AUTO Upgrade:	● No ○ Yes	
Restart	Check for upgrade every	120 Minutes	
Reboot	HTTP/FTP/HTTPS UserName		
	HTTP/FTP/HTTPS Password	•••••	
	Firmware/Config File Prefix		
	Firmware/Config File Postfix		
	Upgrade Check Mode :		
	Always Check For N	Jew Firmware	
	O check new firmware	e only when F/W pre/suffix changes	
	O Always Skip The Fir	mware Check	

8.4. Download/restore configuration

This functionality allows you to import or export the current configuration of the phone in two formats:

• as a BIN(binary) file

• with an extension of XML

From WEB interface:

Selcet: Management \rightarrow Configuration \rightarrow Download Device BIN/XML Configuration. You can also restore the configuration using the following options:

Select: Management \rightarrow Configuration \rightarrow Restore Configuration XML/BIN.

8.5. Autoprovisioning HTTP/HTTPS

Automatic update method configuration functionality is available as a server. The system administrator knowing the specific data of the phone, ie.: MAC address, prepares the server level preconfiguration file that is downloaded on the phone while its initialization (reboot).

Phone support two method of autoprovisioning configuration:

- automated proces using SIP Multicast(available plug'n'play server)
- manually by entering the phone configuration path access to the server

To configure automated method via WEB interface:

Selcet: Management \rightarrow Auto Provision \rightarrow PnP Active

Select: Yes.

To configure manula method via WEB interface:

Select: Management \rightarrow Auto Provision \rightarrow Configure Server Path

9. Troubleshooting

9.1. Why is the phone LCD screen blank?

Ensure your phone is properly plugged into a functional AC outlet.

Ensure that the phone isn't plugged into a plug controlled by a switch that is off. If the phone is plugged into a power strip, try plugging it directly into a wall outlet instead. If your phone is powered from PoE, ensure you use a PoE compliant switch or hub, or contact your system administrator for more information. Check that the power LED is on to ensure the phone is powered on.

Otherwise, contact your system administrator.

9.2. Why does the phone display "Network Unavailable"?

Ensure that the Ethernet cable is plugged into the Internet port on the phone and the Ethernet cable is not loose. Ensure that the switch or hub in your network is operational.

Otherwise, contact your system administrator.

9.3. Why can't I get a dial tone?

Check for any loose connections and that the phone has been installed properly. Check whether dial tone is present on one of the audio modes(handset/headset).

Switch between the Handset, Headset (if present) or Hands-Free Speakerphone to check whether dial tone is present for one of the audio modes. If the dial tone exists on another audio mode, connect a different handset or headset to isolate the problem.

Otherwise, contact your system administrator.

10. Annex

10.1. Hardware Specifications

COMPONENT	DESCRIPTION
POWER ADAPTER	Wejście: 100-240V AC 50-60Hz Wyjście: 5v/1.2A DC
NETWORK PORT	Internet: 10/100 Base-T RJ-45(WAN/LAN) PC: 10/100 Base-T RJ-45 (LAN) Power over Ethernet(PoE) IEEE 802.3af
ADDITIONAL PORT	Handset: RJ-9(4P4C) Headset: RJ-9(4P4C)
GRAPHIC LCD	128 × 96 mm, matryca punktowa LCD
OPERATION TEMPERATURE	-10~50°C
RELATIVE HUMIDITY	10~95%
WEIGHT	0,72 kg

10.2. Technical Specifications

COMPONENT	DESCRIPTION
NETWORK PROTOCOL	TCP, UDP, ICMP, RARP, ARP, NTP, SNTP, UpnP, FTP, TFTP, DNS
VoIP PROTOCOL	SIPv2, SDP(RFC 2327), RTP (RFC 1889,1890), RTCP
VOICE CODECS	G.711A/u law, G.729A/B, G.722, G.726-32, G.723.1, GSM FR
DTMF	RFC(2833), SIP INFO, IN-BAND
SECURITY	HTTPS, TLS, SRTP, VLAN QoS(802.1pq)
QoS	VLAN QoS(802.1Q), ToS, DiffServ