Yealink SIP VP-T49G Release Notes of Version 80

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Yealink SIP IP Phones Release Notes of Version 51.80.0.100

1. Introduction

- Firmware Version:

  T49G: 51.80.0.90 upgrades to 51.80.0.100
- Applicable Models: T49G
- Release Date: July 13th, 2016.

2. New Features

None

3. Optimization

1. Optimized the feature that if you configure the idle timeout for non-office hours, the phone will enter power-saving mode after 10 minutes by default.

4. Bug Fixes

None

5. Default Value Setting Changes

Default Value Factory Setting Change Log				
Features	**Provisioning syntax**	**Description**	**Default Value of Factory Setting**	
			x. 80.0.90	**x.80.0.100**
Power Saving	features.power_saving.off_hour.idle_timeout =	It configures the time (in minutes) to wait in the idle state before IP phone enter power-saving mode during the non-office hours.	1	10

Yealink SIP IP Phones Release Notes of Version 51.80.0.90

1. Introduction

- Firmware Version:

  T49G: 51.80.0.80 upgrades to 51.80.0.90
- Applicable Models: T49G
- Release Date: May 26th, 2016.

2. New Features

1. Added the feature of Power Saving.

3. Optimization

None

4. Bug Fixes

None

5. New Features Descriptions

1 Added the feature of Power Saving.

Description: The power saving feature is used to turn off the screen to conserve energy. The IP phone enters power-saving mode after it has been idle for a certain period of time. And the IP phone will exit power-saving mode if a phone event occurs—for example, if the phone has an incoming call or message, or you press a key on the phone or tap the touch screen. If the screen saver is enabled on your phone, power-saving mode will still occur.

The parameters in the auto provision template are described as follows:

features.power_saving.enable =

features.power_saving.office_hour.idle_timeout =

features.power_saving.off_hour.idle_timeout =

features.power_saving.user_input_ext.idle_timeout =

features.power_saving.office_hour.monday =

features.power_saving.office_hour.tuesday =

features.power_saving.office_hour.wednesday =

features.power_saving.office_hour.thursday =

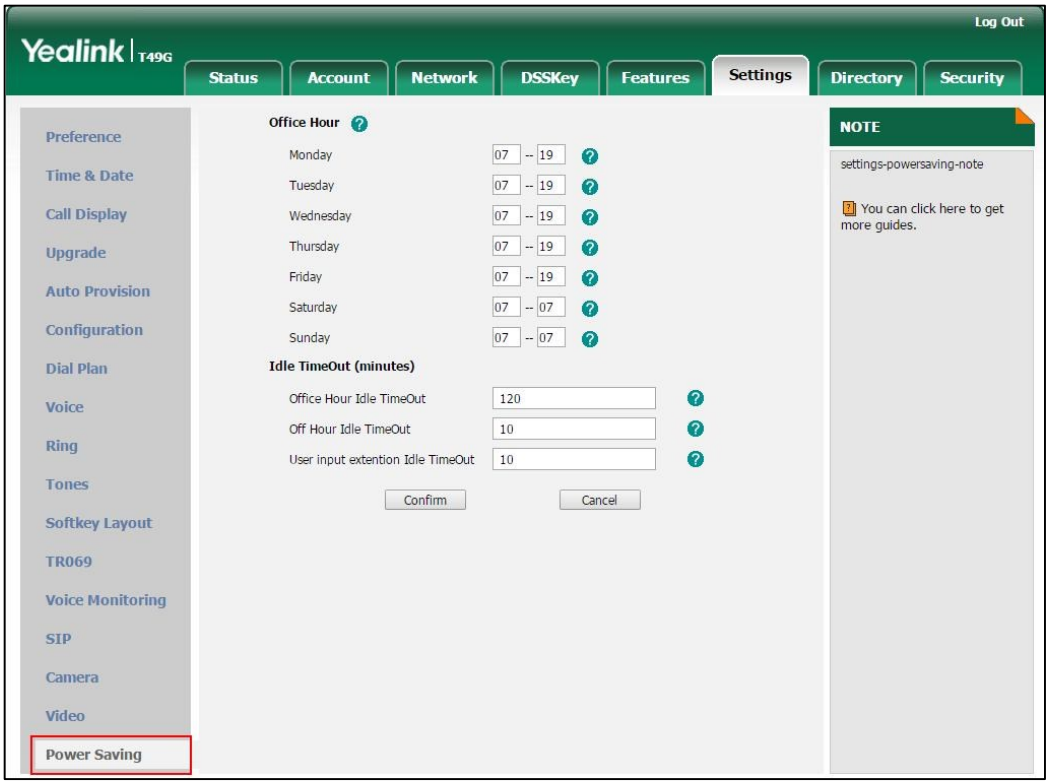
features.power_saving.office_hour.friday =

features.power_saving.office_hour.saturday =

features.power_saving.office_hour.sunday =

To configure the power saving feature via web user interface:

Click on **Settings** ->**Power Saving**.



6. Configuration Parameters Enhancements

Auto Provision Template Flies Change Log							
Firmware Version： [51. 80.0.80]-[51.80.0.90]							
Feature	Provisioning syntax Comparison		Permitted Values	Default Value	Action	Description	File
	51. 80.0.80	51.80.0.90					
Power		features.po	0 or 1	1	Add	It enables or disables the	common.

Saving		wer_saving.enable =				power saving feature. 0-Disabled 1-Enabled	cfg
Power Saving		features.power_saving.office_hour.idle_timeout =	**For SIP VP-T49G:** Interger from 1 to 240 **For SIP-T48G/T46G/T29G:** Interger from 1 to 600	**For SIP VP-T49G:** The default value is 120. **For SIP-T48G/T46G/T29G:** The default value is 480.	Add	It configures the time (in seconds) to wait in the idle state before IP phone enter power-saving mode during the office hours.	common.cfg
Power Saving		features.power_saving.off_hour.idle_timeout =	Integer from 1 to 10	1	Add	It configures the time (in seconds) to wait in the idle state before IP phone enter power-saving mode during the non-office hours.	common.cfg
Power Saving		features.power_saving.user_input_ext.idle_timeout	Integer from 1 to 20	10	Add	It configures the time (in seconds) to wait in the idle state before IP phone enter power-saving mode when pressing a key on the phone or tapping the touch screen (only applicable to SIP VP-T49G/SIP-T48G). Note: If you press a key on the phone or tap the touch screen (only applicable to SIP VP-T49G/SIP-T48G), the idle timeout that applies (User input extension Idle Timeout or Office Hours/Off Hours Idle Timeout) is the timeout with the highest value. If the phone has an incoming call or message, the User input extension	common.cfg

						Idle Timeout is ignored.	
Power Saving		features.power_saving.office_hour.monday	Integer from 0 to 23, Integer from 0 to 23	7,12	Add	It configures the starting time and duration of the day's office hour on Monday. Starting time and duration are separated by commas.	common.cfg
Power Saving		features.power_saving.office_hour.tuesday	Integer from 0 to 23, Integer from 0 to 23	7,12	Add	It configures the starting time and duration of the day's office hour on Tuesday. Starting time and duration are separated by commas.	common.cfg
Power Saving		features.power_saving.office_hour.Wednesday =	Integer from 0 to 23, Integer from 0 to 23	7,12	Add	It configures the starting time and duration of the day's office hour on Wednesday. Starting time and duration are separated by commas.	common.cfg
Power Saving		features.power_saving.office_hour.Thursday =	Integer from 0 to 23, Integer from 0 to 23	7,12	Add	It configures the starting time and duration of the day's office hour on Thursday. Starting time and duration are separated by commas.	common.cfg
Power Saving		features.power_saving.office_hour.Friday =	Integer from 0 to 23, Integer from 0 to 23	7,12	Add	It configures the starting time and duration of the day's office hour on Firday. Starting time and duration are separated by commas.	common.cfg
Power Saving		features.power_saving.office_hour.Saturday =	Integer from 0 to 23, Integer from 0 to 23	7,0	Add	It configures the starting time and duration of the day's office hour on Saturday. Starting time and duration are separated by commas.	common.cfg
Power Saving		features.power_saving.office_hour.Sunday =	Integer from 0 to 23, Integer from 0 to 23	7,0	Add	It configures the starting time and duration of the day's office hour on Sunday. Starting time and duration are separated by commas.	common.cfg

Yealink SIP VP-T49G Release Notes of Version 51.80.0.80

1. Introduction

- Firmware Version:

  T49G: 51.80.0.75 upgrades to 51.80.0.80
- Applicable Models: T49G
- Release Date: Mar 21st, 2016.

2. New Features

None

3. Optimization

1. Optimized the feature that the tooltips on the video pane will be hidden automatically when there is no activity for 15 seconds in full screen mode, and it will be redisplayed if any operations occur.
2. Changed the background color of T49G into blue during the conferencing call.

4. Bug Fixes

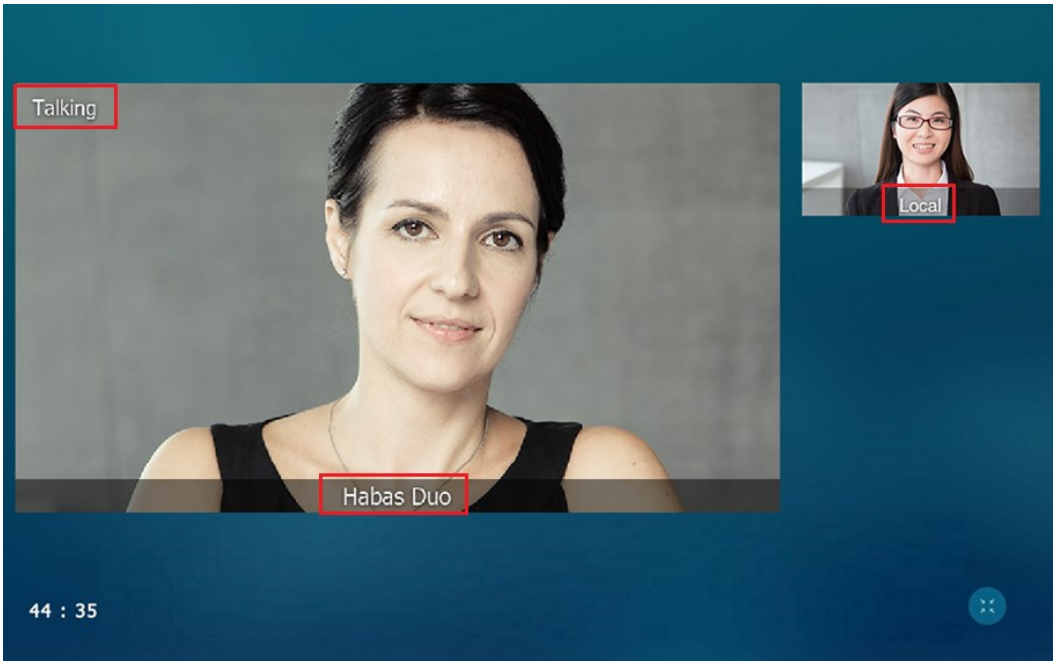
None

5. Optimization Descriptions

1. **Optimized the feature that the tooltips on the video pane will be hidden automatically when there is no activity for 15 seconds in full screen mode, and it will be redisplayed if any operations occur.**

   Description: The phone will enter the video call full-screen within 5 seconds during a call, you may tap  or blank area at the top or bottom of the screen to see the soft keys. In the latest version, after entering the full screen mode, the

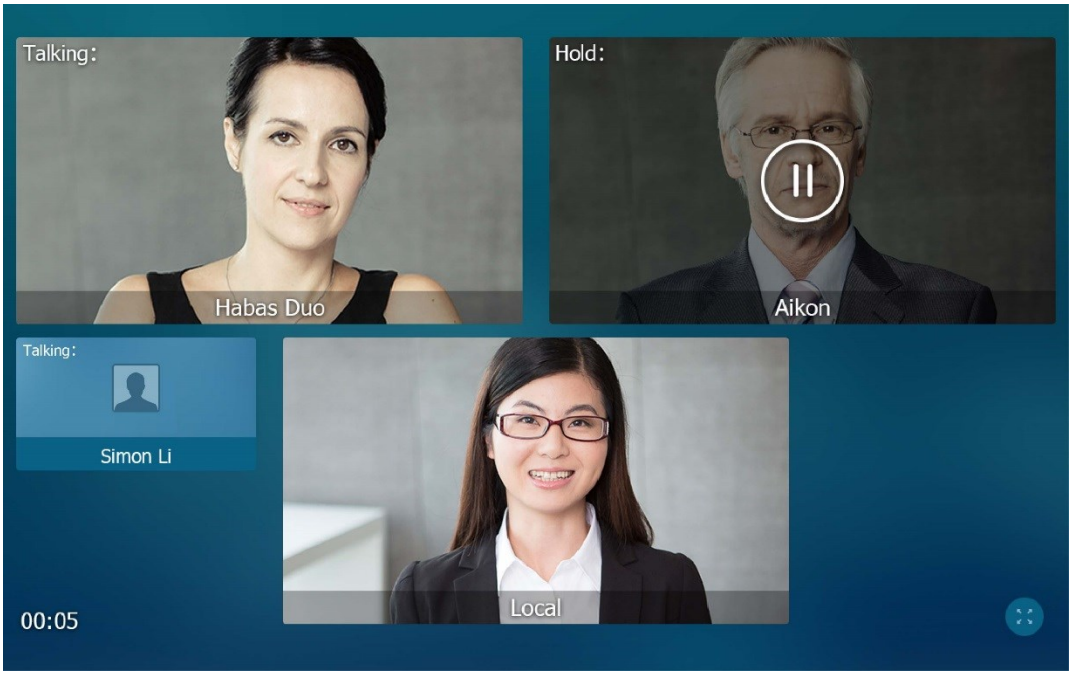
tooltips on the video pane will be hidden automatically when there is no activity for 15 seconds. As shown below, the three tooltips in the red box will be hidden.



And it will be redisplayed if any operations occur.

2. **Changed the background color of T49G into blue during the conferencing call.**

Description: In the latest version, for the participants, the background color is shown as below:



Yealink SIP VP-T49G Release Notes of Version 51.80.0.75

1. Introduction

- Firmware Version:

  T49G: 51.80.0.10 upgrades to 51.80.0.75
- Applicable Models: T49G
- Release Date: Jan 13th, 2016.

2. New Features

1. Added the configuration parameter of sending volume.
2. Added the feature of onscreen keyboard input method customization.
3. Added the supported audio codecs—G7221.C and G722.1, which you can configure them via web user interface.
4. Added the configuration parameter of auto answer tone on web user interface.
5. Added the feature of Share Call Appearance (SCA).

3. Optimization

4. Added the configuration parameter of Full Duplex 1000Mbps which can be configured via web user interface.
5. Optimized the feature that you can configure which country's 5 GHz wireless channels do the IP phones support.
6. Optimized the feature of configuring the access URL of a contact avatar file.
7. Optimized the feature that you can check the network signal strength during the call. When the packet loss is larger than 5% during the call, the touch screen will prompt "The network is unstable".
8. Optimized the feature that you can import up to 3000 mobile contacts from your Bluetooth-enabled mobile phone into the SIP VP-T49G IP Phone.
9. Optimized the feature that the touch screen will not display the prompt message frequently when the IP phone automatically disconnects or re-connects Wi-Fi.
10. Optimized the conference call interface.

11. Optimized the phone interface when you have an active call, and an incoming call arrives on the phone.
12. Optimized the feature that in the dialing interface, if you want to enter "#" after setting the input mode to  , the pound key will not perform as a send key.
13. Optimized the feature of the access URL for a remote phone book.
14. Optimized the feature of entering special characters in the dialing interface.
15. Optimized the feature that you can have a full-HD 1080P video call when you use Yealink SIP VP-T49G or Yealink VCS series.
16. Optimized the feature that the Record key and Screenshot key cannot be configured in Softkey Layout.
17. Optimized the dialing interface.

4. Bug Fixes

1. Fixed the issue that the configuration items of downlink and uplink bandwidth on phone user interface, web user interface and auto provisioning syntax are different.
2. Fixed the issue that in the talk statistics page on web user interface, it may not distinguish the video codec from H264 and H264HP as your configuration.
3. Fixed the issue that when you have a video call, the other party may receive an inverse picture on his IP phone.
4. Fixed the issue that when using Bluetooth-enabled mobile phone, the connection is unstable while answering a call, the phone interface may flash and then the call time starts over.
5. Fixed the issue that the video image may be a little darker on SIP VP-T49G.
6. Fixed the issue that during the video call, when you switch the view, the video images cannot be switched smoothly on the LCD screen.
7. Fixed the issue that it may cause image distortion when the video image maps to the LCD screen in full screen mode.
8. Optimized the subdirectory in the Directory interface.
9. Fixed the issue that in the History interface, the contact information is only shown his name without his custom avatar.
10. Fixed the issue that it may work in simplex mode with Cisco 9951 during a video call.

5. New Features Descriptions

1. Added the configuration parameter of sending volume.

Description: Sending volume allows user to adjust the sending volume of currently engaged audio devices (handset, speakerphone or headset) when the phone is in use.

The parameters in the auto provision template are described as follows:

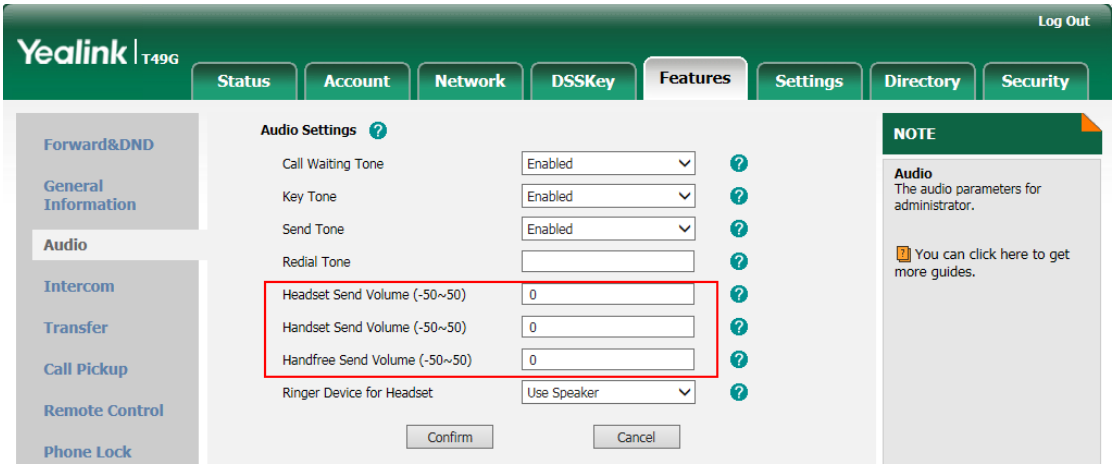
voice.handfree_send =

voice.handset_send =

voice.headset_send =

To set this feature via web user interface:

Click on Features->Audio



2. Added the feature of onscreen keyboard input method customization.

Description: By default, the SIP VP-T49G IP phone supports English and Russia onscreen keyboard. The following takes English keyboard as an example:

Change the language of the keyboard by tapping  on the onscreen keyboard. You can configure custom keyboard language files to provide other languages. The

original onscreen keyboard key only provide one character, you can configure a custom keyboard ime file to make the key provide more characters.

The character cannot be entered until you release your finger from the key.

The parameters in the auto provision template are described as follows:

gui_onescreen_keyboard.url=

For more information please refer to the *Yealink_SIP-T2_Series_T19(P)E2_T4_Series_CP860_IP_Phones_Administrator_Guide_V80_91*

3. **Added the supported audio codecs—G7221.C and G722.1, which you can configure them via web user interface.**

   Description: The audio codec that the phone uses to establish a call should be supported by the SIP server. When placing a call, the IP phone will offer the enabled audio codec list to the server and then use the audio codec negotiated with the called party according to the priority.

   The parameters in the auto provision template are described as follows:

   account.X.codec.Y.enable =

   account.X.codec.Y.payload_type =

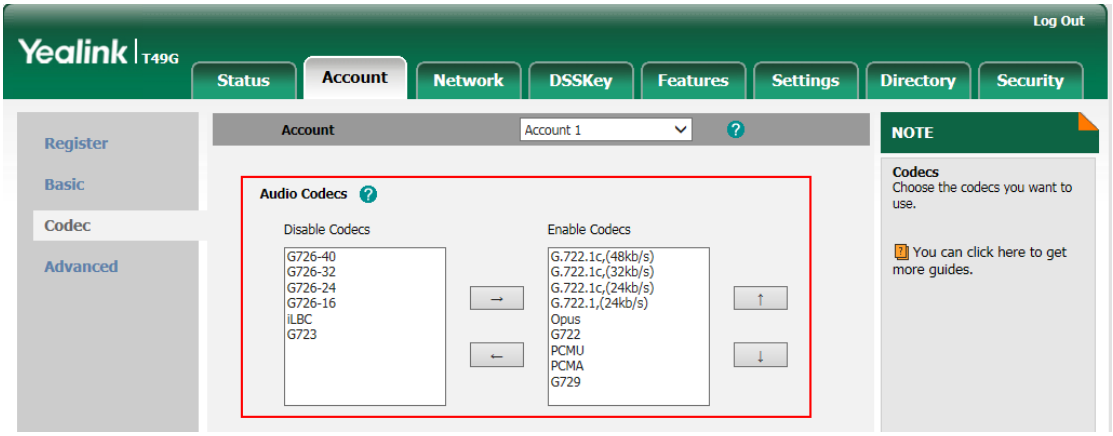
   account.X.codec.Y.priority =

   account.X.codec.Y.rtpmap =

   For more information please refer to the *Yealink_SIP-T2_Series_T19(P) E2_T4_Series_CP860 IP phones_Description of Configuration Parameters in CFG Files_V80_91*

   To configure the codecs to use and adjust the priority of the enabled codecs on a per-line basis via web user interface:

   Click on Account -> Codec



4. **Added the configuration parameter of auto answer tone on web user interface.**

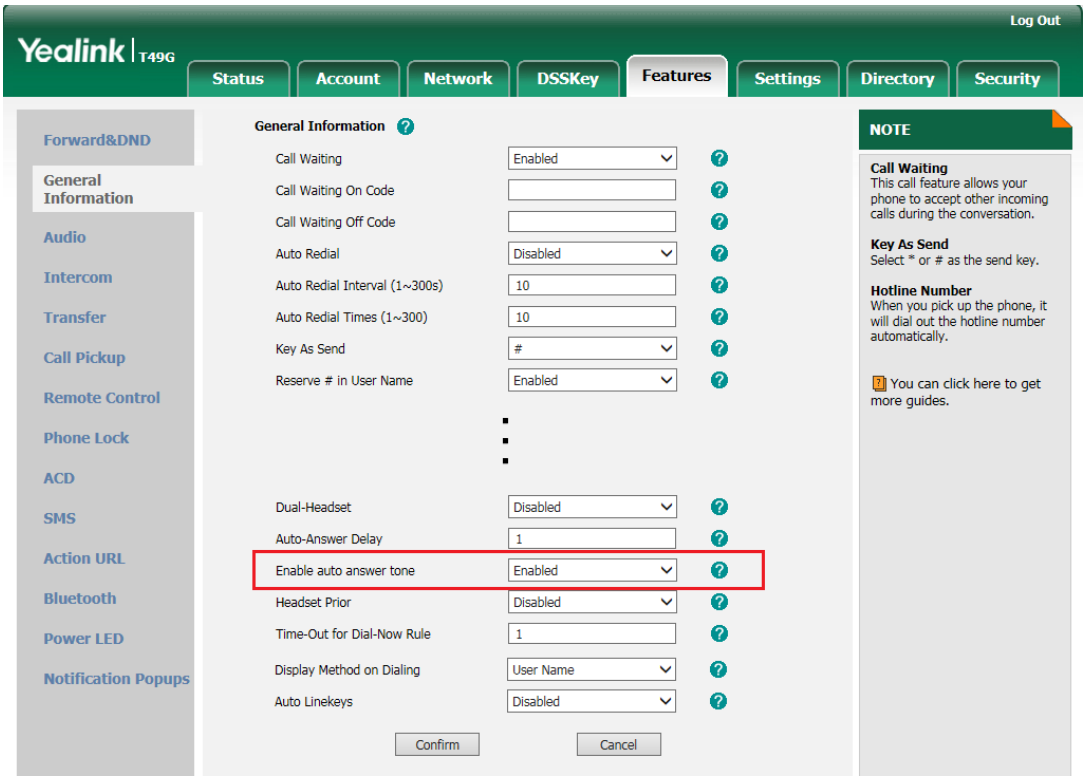
   Description: Auto answer tone allows the IP phone to play a tone when an incoming call is automatically answered. You can customize the auto answer tone or select specialized tone sets (vary from country to country) for your IP phone.

   The parameters in the auto provision template are described as follows:

   features.auto_answer_tone.enable

   To configure auto answer tone via web user interface:

   Click on Features-> General Information.

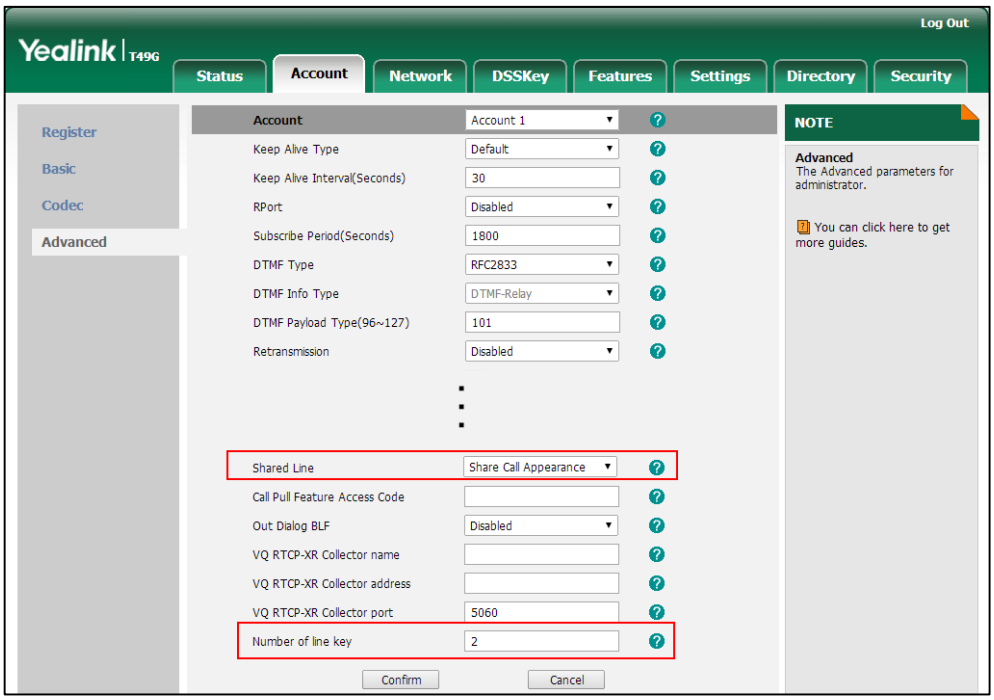


5. **Added the feature of Share Call Appearance (SCA).**

Description: You can use SCA feature to share an extension which can be registered on two or more IP phones at the same time. The shared line is indicated by a different line icon. If two phones share a line, an incoming call to this extension will cause both phones to ring simultaneously. The incoming call can be answered on either phone but not both.

To configure SCA via web user interface:

Click on Account → Advanced → Shared Line



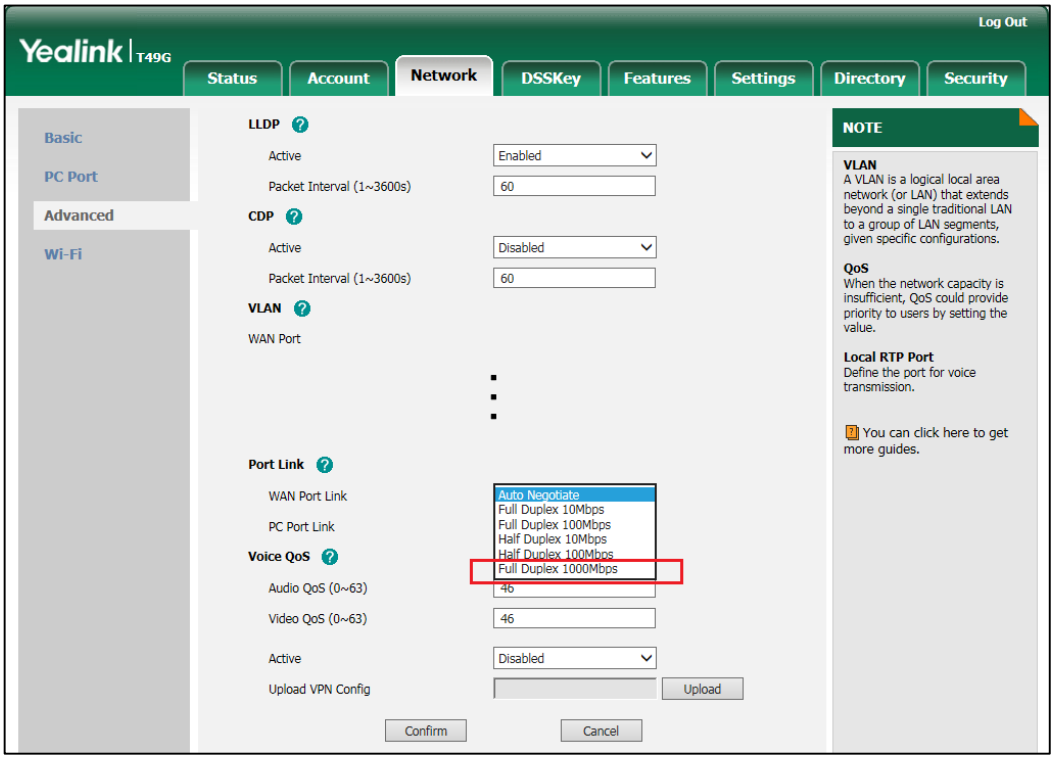
6. Optimization Descriptions

3. **Added the configuration parameter of Full Duplex 1000Mbps which can be configured via web user interface.**

   Description: Full-duplex transmission refers to transmitting voice or data in both directions at the same time; this means one device can send data on the line while receiving data. You can configure the full-duplex transmission on both Internet port and PC port for the IP phone to transmit in 10Mbps, 100Mbps or 1000Mbps. By the way, you can set the transmission speed to 1000Mbps/Auto Negotiate to transmit in 1000Mbps if the IP phone is connected to the switch supports Gigabit Ethernet.

   To configure the transmission methods of Ethernet ports via the web user interface:

Click on Network → Advanced → Pork Link



4. **Optimized the feature that you can configure which country's 5 GHz wireless channels do the IP phones support.**

   Description: Configures which country's 5 GHz wireless channels do the IP phones support. The permitted values are United States, Canada, Europe, Switzerland, Russia, Japan, Singapore, China, Israel, Korea, Turkey, Australia, South Africa, Brazil, Taiwan, or New Zealand. The default value is China.

   The parameters in the auto provision template are described as follows:

   wifi.country

   For more information please refer to the *Yealink_SIP-T2_Series_T19(P) E2_T4_Series_CP860 IP phones_Description of Configuration Parameters in CFG Files_V80_91*

   If you want to get more details about your own country, please refer to the following link:

   https://en.wikipedia.org/wiki/List_of_WLAN_channels

5. **Optimized the feature of configuring the access URL of a contact avatar file.**

   Description: You can configure the access URL of a contact avatar file. The format of the contact avatar must be *.png, *.jpg, *.bmp. The contact avatar file should be uploaded to the provisioning server in advance. If you change this parameter,

the IP phone will reboot automatically instead of rebooting manually to make the change take effect.

The parameters in the auto provision template are described as follows:

local_contact.icon.url

local_contact.icon_image.url

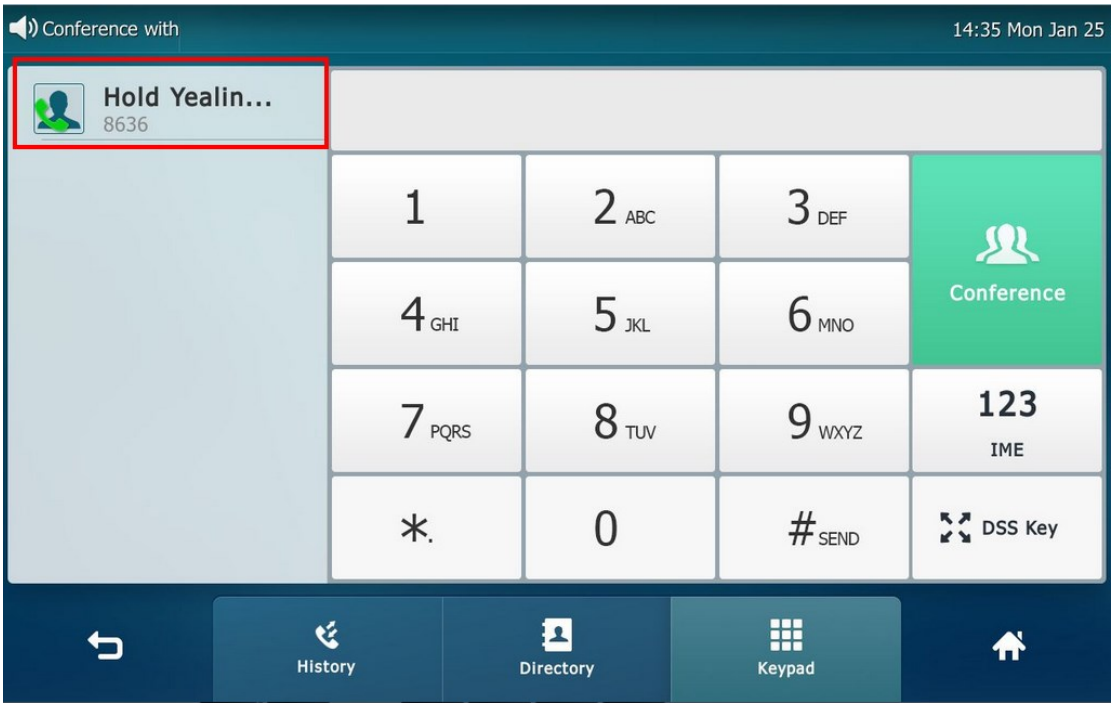
local_contact.photo.url

local_contact.image.url

local_contact.data_photo_tar.url

6. **Optimized the conference call interface.**

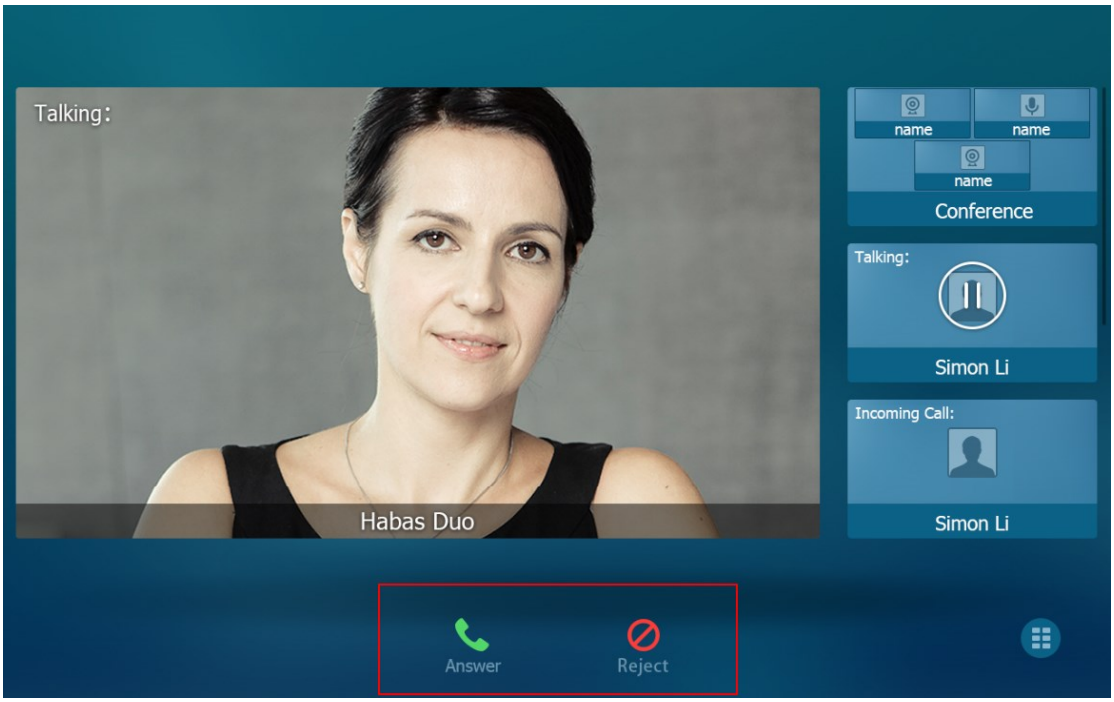
Description: You can create up to three-way video conference call and five-way mixture audio and video conference. When you have an active call and one or more calls on hold (either on the same line key or a different line key). Tap the Hold call to join the calls in the conference directly after tapping the Conference softkey. Or you can also enter the desired number to make a 3-way conference call.



7. **Optimized the phone interface when you have an active call, and an incoming call arrives on the phone.**

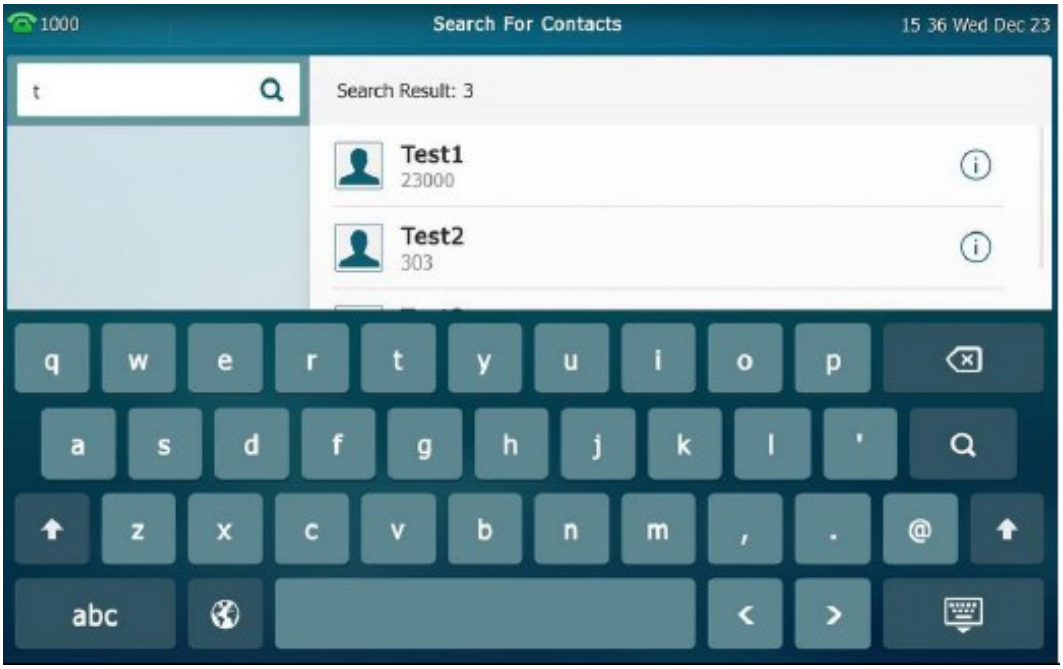
Description: When you have an active call, and an incoming call arrives on the phone. The incoming call information is displayed on the touch screen. You can tap the “Answer” or “Reject” soft key directly on the bottom of the touch screen,

instead of tapping the incoming call avatar first.



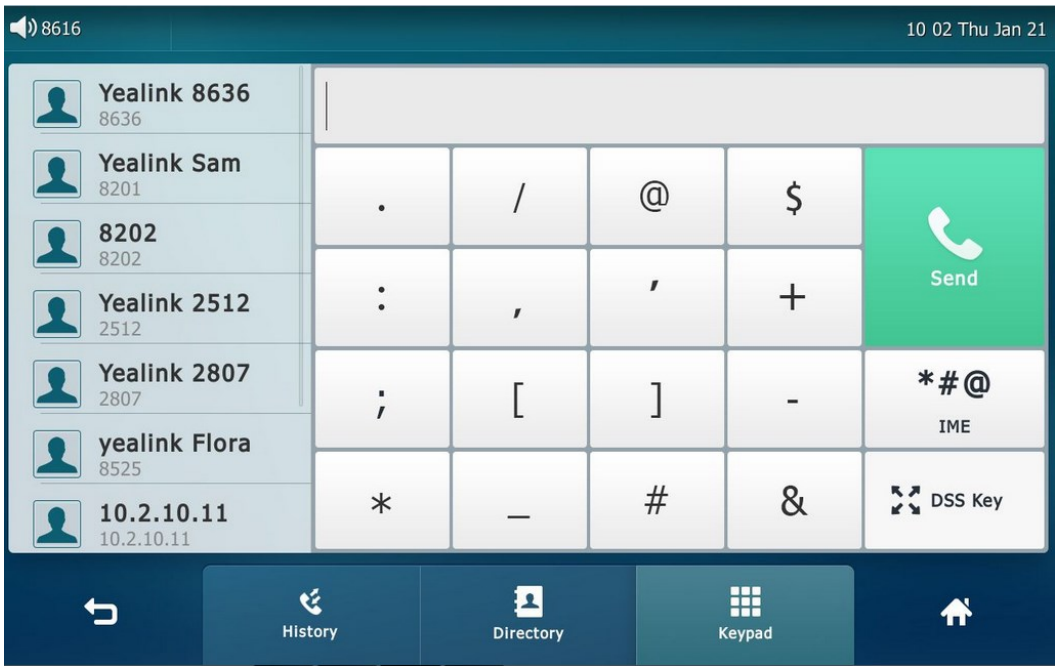
8. **Optimized the feature of the access URL for a remote phone book.**

Description: You can configure your new phone to access up to 5 remote phone books. The phone supports up to 5000 remote phone book entries. When you add "=#SEARCH" in the access URL via web user interface, for example: remote_phonebook.data.1.url = http://10.40.0.3/phonebook/?name=#SEARCH, the touch screen will automatically display all the contacts' names and numbers that meet those matching conditions as shown below:



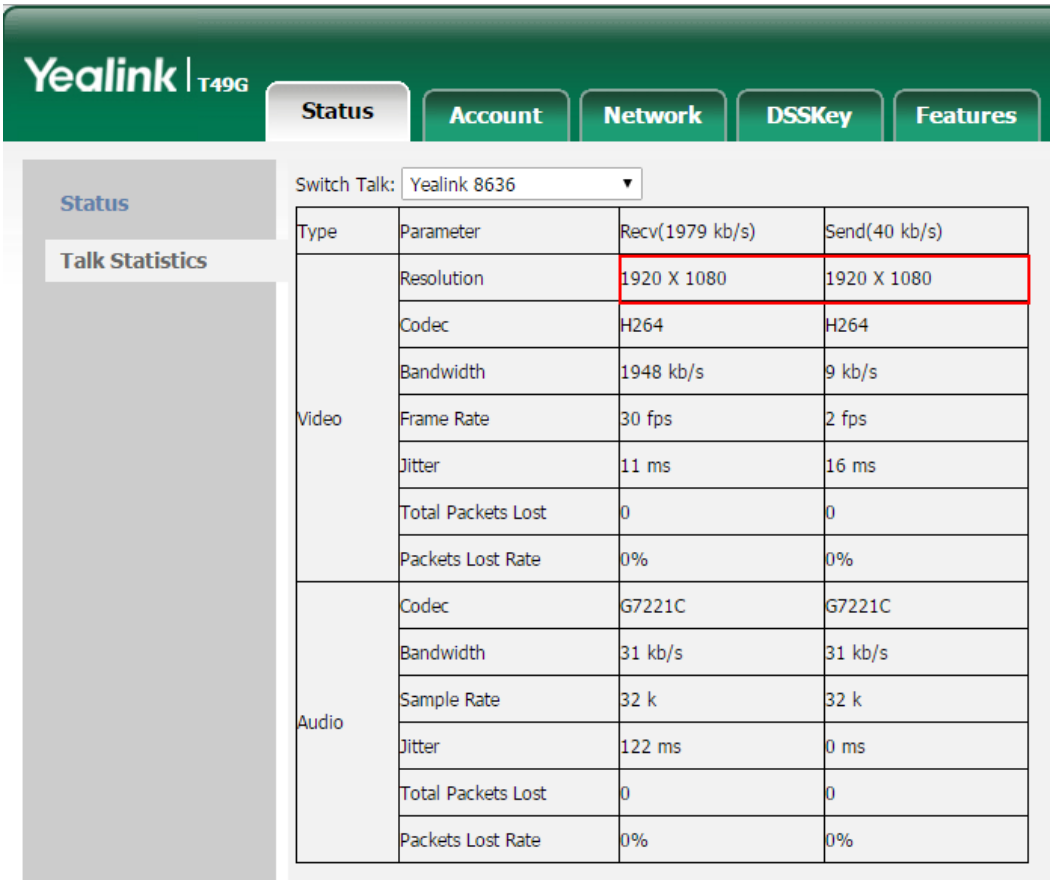
9. **Optimized the feature of entering special characters in the dialing interface.**

   Description: Tap 123 IME one or more times to set the input mode to *#@ IME. It will provide the following special characters: ./@$:,'+;[]-*_#&. You can tap the character(s) you want to enter. The phone user interface is shown as below:



10. **Optimized the feature that you can have a full-HD 1080P video call when you use Yealink SIP VP-T49G or Yealink VCS series.**

    Description: You can have a full-HD 1080P video call when you use Yealink SIP VP-T49G or Yealink VCS series. But the actual resolution depends on the performance of the remote endpoint, and is affected by the quality of the communication channel.



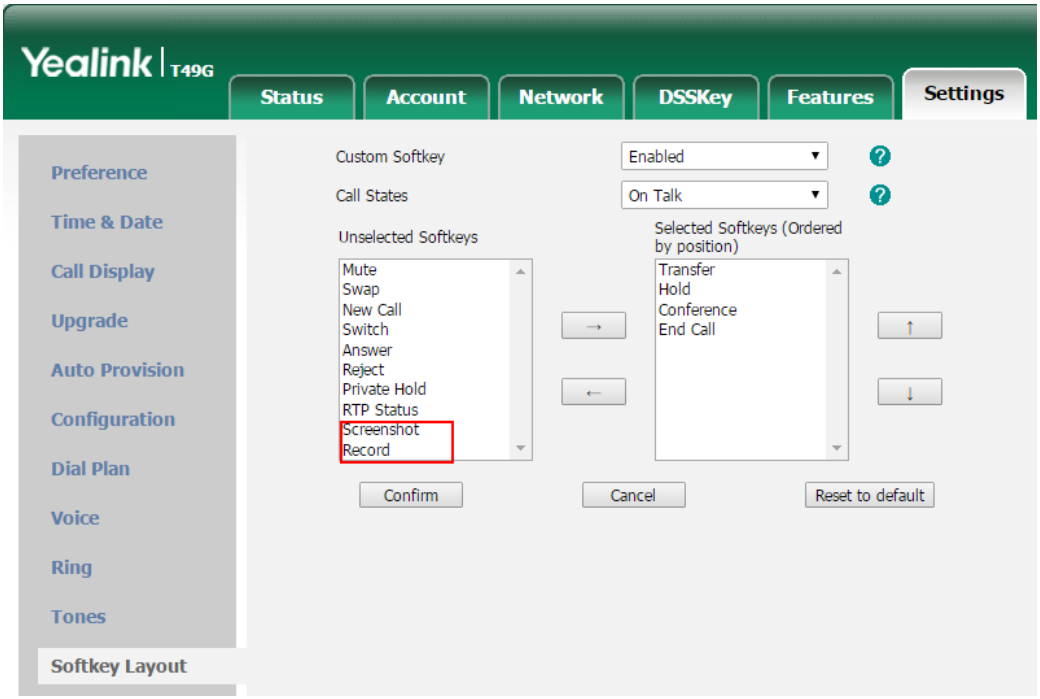
Type	Parameter	Recv(1979 kb/s)	Send(40 kb/s)
Video	Resolution	1920 X 1080	1920 X 1080
	Codec	H264	H264
	Bandwidth	1948 kb/s	9 kb/s
	Frame Rate	30 fps	2 fps
	Jitter	11 ms	16 ms
	Total Packets Lost	0	0
	Packets Lost Rate	0%	0%
Audio	Codec	G7221C	G7221C
	Bandwidth	31 kb/s	31 kb/s
	Sample Rate	32 k	32 k
	Jitter	122 ms	0 ms
	Total Packets Lost	0	0
	Packets Lost Rate	0%	0%

11. Optimized the feature that the Record key and Screenshot key cannot be configured in Softkey Layout.

Description: You can record calls by tapping a record key on the phone. The SIP VP-T49G IP phone supports record and URL record. Also, you can capture the screenshots during a video call via phone user interface.

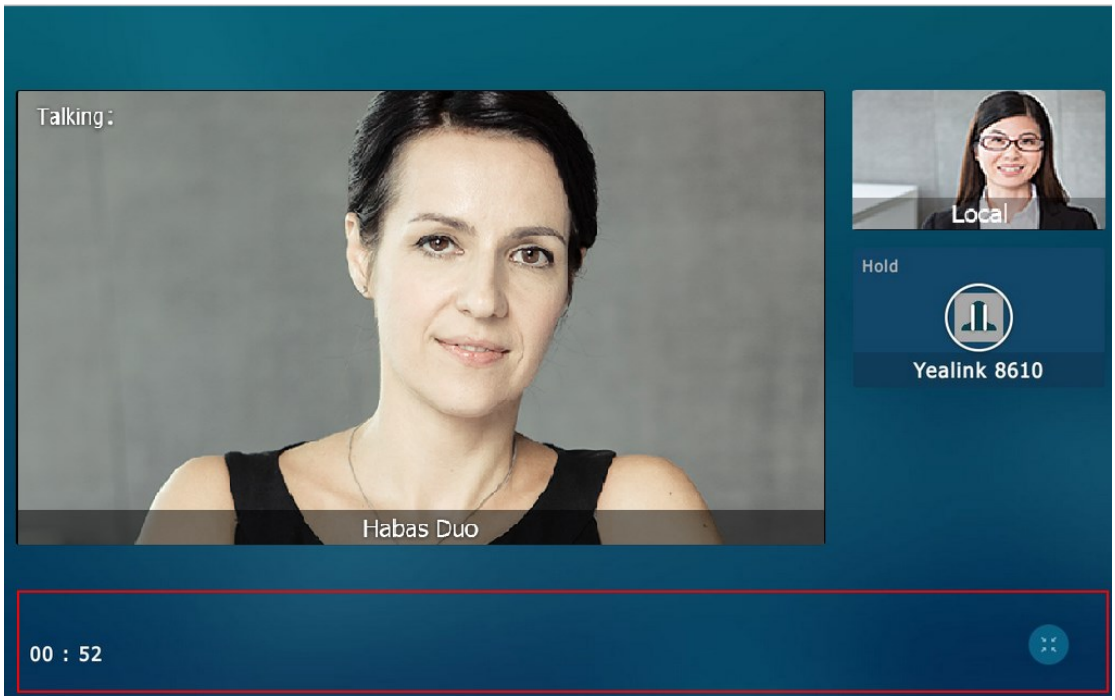
To configure the Record key and Screenshot key via web user interface:

Click on Settings -> Softkey Layout.



12. Optimized the dialing interface.

Description: During the video call, you can tap any blank space on the bottom of the touch screen, and then the softkey will be shown.



7. Configuration Parameters Enhancements

Auto Provision Template Flies Change Log							
Firmware Version：[51.80.0.10]-[51.80.0.75]							
Feature	Provisioning syntax Comparison		Permitted Values	Default Value	Action	Description	File
	51.80.0.10	51.80.0.75					
ACD	genesys.acd .after_call_ work_displa y.enable		0 or 1	0	Delete	It enables or disables the status of the ACD agent to be automatically changed to ACW (after call work) after the call. 0-Disabled 1-Enabled Note: ACW status depends on support from a Genesys server. Contact your server administrator for more information. If your server supports this status, make sure the parameter value is set to 1 (Enabled).	MAC.cfg
Custom Input Method		gui_onscreen _keyboard.url	URL within 511 characters	Blank	Add	It configures the access URL of the custom onscreen keyboard input method file. Example: gui_onscreen_keyboard.url = http://192.168.1.25/keyboa rd_lang.xml gui_onscreen_keyboard.url = http://192.168.1.25/keyboa rd_ime_german.xml gui_onscreen_keyboard.url = http://192.168.1.25/keyboa rd_layout_german.xml During the auto provisioning process, the IP phone	common. cfg

						connects to the provisioning server “192.168.1.25”, and downloads the custom files “keyboard_lang.xml”, “keyboard_ime_german.xml” and “keyboard_layout_german.xml”.	
Features_Auto Answer		features.auto_answer_tone.enable	0 or 1	0	Add	It enables or disables the phone to play a warning tone when an incoming call is automatically answered. 0-Disabled 1-Enabled Note: For the call coming from a SIP account, it works only if the value of the parameter "account.X.auto_answer" is set to 1 (Enabled). It is also applicable to IP calls.	common.cfg
Bandwidth	features.outgoing_bandwidth	features.uplink_bandwidth	0, 256, 384, 512, 640, 768, 1024, 1280, 1500, 2000, 3000 or 4000	0	Modify	It specifies the maximum transmitting bandwidth for the IP phone. 0-Auto 256-256kb/s 384-384kb/s 512-512kb/s 640-640kb/s 768-768kb/s 1024-1024kb/s 1280-1280kb/s 1500-1500kb/s 2000-2000kb/s 3000-3000kb/s 4000-4000kb/s If it is set to 0 (Auto), the IP phone will select the appropriate transmitting bandwidth automatically.	common.cfg

Bandwidth	features.incoming_bandwidth	features.downlink_bandwidth	0, 256, 384, 512, 640, 768, 1024, 1280, 1500, 2000, 3000 or 4000	0	Modify	It specifies the maximum receiving bandwidth for the IP phone. 0-Auto 256-256kb/s 384-384kb/s 512-512kb/s 640-640kb/s 768-768kb/s 1024-1024kb/s 1280-1280kb/s 1500-1500kb/s 2000-2000kb/s 3000-3000kb/s 4000-4000kb/s If it is set to 0 (Auto), the IP phone will select the appropriate receiving bandwidth automatically.	common.cfg
Genesys ACD	account.X.acd.auto_login_enable		0 or 1	0	Delete	It enables or disables the IP phone to automatically log into the ACD system when account registers, phone reboots or network reconnects. 0-Disabled 1-Enabled Note: It works only if the ACD agent successfully logged into the system before and It is only applicable to Genesys ACD feature.	MAC.cfg
Volume		voice.handfree_send	Integer from -50 to 50	0	Add	It configures the sending volume of the speaker. Note: We recommend that you modify this parameter cautiously. An unreasonable value may render the voice quality bad.	common.cfg
Volume		voice.handset_send	Integer from -50 to 50	0	Add	It configures the sending volume of the handset.	common.cfg

						Note: We recommend that you modify this parameter cautiously. An unreasonable value may render the voice quality bad.	
Volume		voice.headset_send	Integer from -50 to 50	0	Add	It configures the sending volume of the headset. Note: We recommend that you modify this parameter cautiously. An unreasonable value may render the voice quality bad.	common.cfg
Features_Wi-Fi		wifi.country	United states, Canada, Europe, Switzerland, Russia, Japan, Singapore, China, Isarel, Korea, Turkey, Australia, South Afria, Brazil, Taiwan, or New Zealand.	China	Add	It configures which country's 5 GHz wireless channels do the IP phones support.	common.cfg
RTP Port	network.port.max_rtpport		Integer from 1 to 65535	12780	Delete	It configures the maximum local RTP port. Note: The value of the maximum local RTP port cannot be less than that of the minimum local RTP port.	common.cfg
RTP Port	network.port.min_rtpport		Integer from 1 to 65535	11780	Delete	It configures the minimum local RTP port.	common.cfg

8. Default Value Setting Changes

Default Value Factory Setting Change Log			
Features	**Description**	**Default Value of Factory Setting**	
		51.80.0.10	**51.80.0.75**

Audio Codec	It enables or disables the specified codec for account X. 0-Disabled 1-Enabled Example: account.1.codec.1.enable = 1 It means that the codec PCMU is enabled on the account 1.	account.X.codec.Y.enable = When Y=1, the default value is 1; When Y=2, the default value is 1; When Y=3, the default value is 0; When Y=4, the default value is 1; When Y=5, the default value is 1; When Y=6, the default value is 0; When Y=7, the default value is 0; When Y=8, the default value is 0; When Y=9, the default value is 0; When Y=10, the default value is 0; When Y=11, the default value is 0;	account.X.codec.Y.enable = When Y=1, the default value is 1; When Y=2, the default value is 1; When Y=3, the default value is 0; When Y=4, the default value is 1; When Y=5, the default value is 1; When Y=6, the default value is 0; When Y=7, the default value is 0; When Y=8, the default value is 0; When Y=9, the default value is 0; When Y=10, the default value is 0; When Y=11, the default value is 0; When Y=12, the default value is 1; When Y=13, the default value is 1; When Y=14, the default value is 1; When Y=15, the default value is 1;
Audio Codec	It configures the codec for account X. Example: account.1.codec.1.payload_type = PCMU	account.X.codec.Y.payload_type = When Y=1, the default value is PCMU; When Y=2, the default value is PCMA; When Y=3, the default value is G723; When Y=4, the default value is G729; When Y=5, the default value is G722; When Y=6, the default value is iLBC; When Y=7, the default value is G726-16; When Y=8, the default value is G726-24; When Y=9, the default value is G726-32; When Y=10, the default value is G726-40; When Y=11, the default value is Opus;	account.X.codec.Y.payload_type = When Y=1, the default value is PCMU; When Y=2, the default value is PCMA; When Y=3, the default value is G723; When Y=4, the default value is G729; When Y=5, the default value is G722; When Y=6, the default value is iLBC; When Y=7, the default value is G726-16; When Y=8, the default value is G726-24; When Y=9, the default value is G726-32; When Y=10, the default value is G726-40; When Y=11, the default value is Opus; When Y=12, the default value is G7221 (it represents the codec G722.1c(48kb/s)); When Y=13, the default value is G7221 (it represents the codec

			G722.1c(32kb/s)); When Y=14, the default value is G7221 (it represents the codec G722.1c(24kb/s)); When Y=15, the default value is G7221 (it represents the codec G722.1(24kb/s));
Audio Codec	It configures the priority of the enabled codec for account X. Example: account.1.codec.1.priority = 2 Note: For SIP VP-T49G IP phones, numerical value 0 is defined as the highest priority in the enable codec list and disable codec list. For SIP-T48G/T46G/T42G/T41P/T40P/T29G/T27P/T23P/T23G/T21(P)E2/T19(P) E2 IP phones, the priority of codec in disable codec list is not specified, and numerical value 1 is defined as the highest priority in the enable codec list.	account.X.codec.Y.priority = When Y=1, the default value is 5; When Y=2, the default value is 6; When Y=3, the default value is 13; When Y=4, the default value is 7; When Y=5, the default value is 4; When Y=6, the default value is 12; When Y=7, the default value is 11; When Y=8, the default value is 10; When Y=9, the default value is 9; When Y=10, the default value is 8; When Y=11, the default value is 14;	account.X.codec.Y.priority = When Y=1, the default value is 5; When Y=2, the default value is 6; When Y=3, the default value is 13; When Y=4, the default value is 7; When Y=5, the default value is 4; When Y=6, the default value is 12; When Y=7, the default value is 11; When Y=8, the default value is 10; When Y=9, the default value is 9; When Y=10, the default value is 8; When Y=11, the default value is 14; When Y=12, the default value is 0; When Y=13, the default value is 1; When Y=14, the default value is 2; When Y=15, the default value is 3;
Audio Codec	It configures the rtpmap of the audio codec for account X. Example: account.1.codec.1.rtpmap = 0	account.X.codec.Y.rtpmap = When Y=1, the default value is 0; When Y=2, the default value is 8; When Y=3, the default value is 4; When Y=4, the default value is 18; When Y=5, the default value is 9; When Y=6, the default value is 106; When Y=7, the default value is 103;	account.X.codec.Y.rtpmap = When Y=1, the default value is 0; When Y=2, the default value is 8; When Y=3, the default value is 4; When Y=4, the default value is 18; When Y=5, the default value is 9; When Y=6, the default value is 106; When Y=7, the default value is 103; When Y=8, the default value is 104; When Y=9, the default value is 102; When Y=10, the default value is

		When Y=8, the default value is 104; When Y=9, the default value is 102; When Y=10, the default value is 105; When Y=11, the default value is 107;	105; When Y=11, the default value is 107; When Y=12, the default value is 121; When Y=13, the default value is 122; When Y=14, the default value is 123; When Y=15, the default value is 124;
Super_search	It enables or disables the recent call in dialing feature. 0-Disabled 1-Enabled If it is set to 1 (Enabled), you can see the placed calls list when the IP phone is on the pre-dialing screen.	super_search.recent_call=?0	super_search.recent_call=?1