

Yealink W52P Release Notes

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1. Introduction

Product Name: DECT-W52P

Firmware Version:

Base Version: 25.73.193.11 upgrade to 25.73.193.40

Release Date: July 4, 2015

2. New Features

3. Optimization

4. Bug Fixes

- 1. Fixed the issue that the software cannot be compatible with the new RFPI code, and increased the limit degradation feature in the newly version.
 - **Note:** In the newly version of (25.73.0.40/25.73.193.40), users could not upgrade to the previous versions, because the RFPI code wasn't supported in the previous versions. If the upgrade fails, please contact the Yealink FAE.
- 2. Fixed the issue that the phone can cause memory leaks when user access the page syslog frequently.

5. Description to the newly added features

- 6. Optimization description
- 7. Default Value Setting Changes
- 8. Configuration Parameters Enhancements



1. Introduction

Product Name: DECT-W52P

Firmware Version:

Base Version: 25.73.193.10 upgrade to 25.73.193.11

Handset Version: 26.73.193.2 upgrade to 26.73.193.3

Release Date: Feb 12, 2015

2. New Features

3. Optimization

4. Bug Fixes

- 1. Fixed the issue that the phone would not be configured to user the other DTMF Types when the SIP server was not supported the RFC 2833.
- 2. Fixed the issue that the incoming call record in the history was didn't match the XML remote phonebook contacts.
- 3. Fixed the issue that the authentication fails when the server which the RPS pointed to needed two-way authentication certification.

5. Description to the newly added features

6. Optimization description

7. Default Value Setting Changes



Auto Provision Template Flies Change Log

Firmware Version:

Base Version: 25.73.0.25 upgrade to 25.73.0.27

Handset Version: 26.73.0.10 upgrade to 26.73.0.11

| Feature | Provisioning syntax | | Permitted Action | | Description | File | |
|---------|-------------------------------|---|--|----------|--|------|--|
| | Comparison | | Values | | | | |
| | V73 V72 | | | | | | |
| DNS | dns_cache_a. X.ip = | account.X.d ns_cache_a. Y.ip = | IP address | Modified | It configures the IP address that the domain name in A record X maps to. The default value is blank. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ | |
| DNS | dns_cache_a. X.name = | account.X.d ns_cache_a. Y.name = | Domain name | Modified | It configures the domain name in A record X. The default value is blank. For more information, refer to Server Redundancy on Yealink IP Phones. | сом | |
| DNS | dns_cache_a. X.ttl = | account.X.d ns_cache_a. Y.ttl = | Integer from 30 to 2147483647 | Modified | It configures the time interval (in seconds) that A record X may be cached before the record should be consulted again. The default value is 300. For more information, refer to Server Redundancy on Yealink IP Phones. | сом | |
| DNS | dns_cache_n aptr.X.flags = | account.X.d ns_cache_n aptr.Y.flags = | S, A, U or P | Modified | It configures the flag of NAPTR record X. (Always "s" for SIP, which means to do an SRV lookup on whatever is in the replacement field) S-Do an SRV lookup next. A-Do an A lookup next. U-No need to do a DNS query next. P-Service customized by the user | СОМ | |



| DNS | dns_cache_n aptr.X.name = | account.X.d ns_cache_n aptr.Y.name = | Domain name | Modified | The default value is blank. For more information, refer to Server Redundancy on Yealink IP Phones. It configures the domain name to which NAPTR record X refers. The default value is blank. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |
|-----|--|--|--------------------------------------|----------|--|-----|
| DNS | dns_cache_n aptr.X.order = | account.X.d ns_cache_n aptr.Y.order = | Integer from 0 to 65535 | Modified | It configures the order of NAPTR record X. NAPTR record with lower order is more preferred. The default value is 0. For more information, refer to Server Redundancy on Yealink IP Phones. | сом |
| DNS | dns_cache_n aptr.X.prefere nce = | account.X.d ns_cache_n aptr.Y.prefer ence = | Integer from 0 to 65535 | Modified | It configures the preference of NAPTR record X. NAPTR record with lower preference is more preferred. The default value is 0. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |
| DNS | dns_cache_n aptr.X.replace = | account.X.d ns_cache_n aptr.Y.replac e = | Domain name | Modified | It configures a domain name to be used for the next SRV query in NAPTR record X. The default value is blank. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |
| DNS | dns_cache_n aptr.X.service = | account.X.d ns_cache_n aptr.Y.servic e = | String within 32 characters | Modified | It configures the transport protocol available for the SIP server in NAPTR record X. The default value is blank. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |



| DNS | dns_cache_n aptr.X.ttl = | account.X.d ns_cache_n aptr.Y.ttl = | Integer from 30 to 2147483647 | Modified | It configures the time interval (in seconds) that NAPTR record X may be cached before the record should be consulted again. The default value is 300. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |
|-----|--------------------------------|--|--|----------|--|-----|
| DNS | dns_cache_sr v.X.name = | account.X.d ns_cache_sr v.Y.name = | Domain name | Modified | It configures the domain name in SRV record X. For more information, refer to Server Redundancy on Yealink IP Phones. | сом |
| DNS | dns_cache_sr v.X.port = | account.X.d ns_cache_sr v.Y.port = | Integer from 0 to 65535 | Modified | It configures the port to be used in SRV record X. The default value is 0. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |
| DNS | dns_cache_sr v.X.priority = | account.X.d ns_cache_sr v.Y.priority = | Integer from 0 to 65535 | Modified | It configures the priority for the target host in SRV record X. Lower priority is more preferred. The default value is 0. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |
| DNS | dns_cache_sr v.X.target = | account.X.d ns_cache_sr v.Y.target = | Domain name | Modified | It configures the domain name of the target host for an A query in SRV record X. The default value is blank. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |
| DNS | dns_cache_sr v.X.ttl = | account.X.d ns_cache_sr v.Y.ttl = | Integer from 30 to 2147483647 | Modified | It configures the time interval (in seconds) that SRV record X may be cached before the record should be consulted again. The default value is 300. For more information, refer to | сом |



| | | | | | Server Redundancy on Yealink IP Phones. It configures the weight of the target host in SRV | |
|-----|------------------------------|--|-------------------------------|----------|---|-----|
| DNS | dns_cache_sr v.X.weight = | account.X.d ns_cache_sr v.Y.weight = | Integer from 0 to 65535 | Modified | record X. When priorities are equal, weight is used to differentiate the preference. Higher weight is more preferred. The default value is 0. For more information, refer to Server Redundancy on Yealink IP Phones. | СОМ |

8. Configuration Parameters Enhancements



1. Introduction

Product Name: DECT-W52P

■ Firmware Version:

Base Version: 25.73.193.5 upgrade to 25.73.193.10

Release Date: Nov 11, 2014

2. New features

3. Optimization

- 1. Optimized a configuration item to configure the channel mode when an incoming intercom call is answered through the handset. The headset should be connected in advance.
- 2. Optimized the SIP XSI verifying feature that the phone would send the XSI message when the register name was changed.

4. Bug Fixes

- Fixed the issue that the configuration item of "features.voice_mail_tone_enable ="could not be take effect.
- 2. Fixed the issue that the voice tone could not be take effect when it was set as 0 while the "voice.tone.country =" was set as Custom.
- Fixed the issue that the configuration item of "auto_provision.handset_configured.enable =" could not be take effect.
- 4. Fixed the issue that the dial tone would still be effective if user set this parameter go beyond the ranges.
- 5. Optimized the display of outgoing calls, the LCD can now display the number and name.



5. Description to the newly added features

6. Optimization description

1. Added a configuration item to configure the channel mode when an incoming intercom call is answered through the handset. The headset should be connected in advance.

The auto provision template are as follows:

features.intercom.headset_prior.enable =

0-Speaker Mode

1-Headset Mode

The default value is 0.

For more information please refer to the

Yealink_W52P_IP_DECT_Phones_Auto_Provisioning_Guide_V73_20.

7. Default Value Setting Changes

8. Configuration Parameters Enhancements

| Auto Provision Template Flies Change Log | | | | | | | | |
|--|--|--------------|-----------|--------|---|------|--|--|
| Firmware Ve | Firmware Version: [25.73.193.5]-[25.73.193.10] | | | | | | | |
| Feature | Provisioning syr | ntax | Permitted | Action | Description | File | | |
| | Comparison | | Values | | | | | |
| | 25.73.193.5 | 25.73.193.10 | | | | | | |
| Phone Features | features.interco m.headset_prior. enable = | | 0 or 1 | Added | It configures the channel mode when an incoming intercom call is answered through the handset. The headset should be connected in advance. O-Speaker Mode 1-Headset Mode The default value is 0. | СОМ | | |



1. Introduction

Product Name: DECT-W52P

Firmware Version:

Base Version: 25.50.193.1 upgrade to 25.73.193.5

Handset Version: 26.50.193.1 upgrade to 26.73.193.2

■ Release Date: September 9, 2014

2. New features

- 1. Added the feature of Protect personalized settings after auto provisioning and upgrade.
- 2. Added Off-hook hot line dialing feature.
- 3. Added the Backup Server.
- 4. Added support for the Broadsoft SIP XSI verifying feature.
- 5. Added the support of wildcards (\$PN and \$MAC) in the provisioning server URL.
- 6. Added support of Network Conference.

3. Optimization

- 1. Optimized the Intercom Call feature that it can now support for Metaswitch platform.
- 2. User can now cancel the handset of the registration via web interface.
- 3. Add the auto answer message format.
- 4. Optimized the behavior problems after Enable/Disable of LLDP.
- 5. Optimized the computer response speed when user checking the contacts and telephone records.
- 6. Optimized the display experience of Broadsoft contacts.
- 7. Optimized the display experience of call log.
- 8. Optimized the verification of URL address/Name/Password on the LCD interface.



- 9. Added the Russian language.
- 10. Unified the auto provisioning files (Com.cfg/MAC.cfg).
- 11. Optimized the feature of Busy Tone.
- 12. Optimized the feature of Trouble Shooting.
- 13. Optimized the sequence of codec.
- 14. Optimized the control of answering/rejecting anonymous calls on the handset interface.
- 15. As of firmware version 73, some of the Broadsoft features are only applicable to the UC-One firmware version.
- 16. Optimized the feature of Time zone.
- 17. As of firmware version 73 and the UC-One firmware version the RPS feature was enable by default.
- 18. Optimize the standby time of the battery.
- 19. File encryption of config.bin.
- 20. Modified the Daylight Saving Time in the file of DST.

4. Bug Fixes

- 1. Fixed the issue that users can't fill in these two special characters of "&"and"!" When registered account label.
- 2. Fixed the issue that the SIP server address does not accept rnd-1 format.
- 3. Fixed the issue that the SIP server can't support Special domain name without ".".
- 4. Fixed the issue that the W52P couldn't support auto answer (via notify).

5. Description to the newly added features

 Added the feature of Protect personalized settings after auto provisioning and upgrade.

Description: In the new version, when a user customized feature configuration settings via web/phone interface, the personalized configuration settings could not be covered after auto provisioning.

The auto provision template are as follows:

It enables or disables the phone to synchronize personalized configuration settings with the auto provisioning server.

If it is set to 1 (Enabled), personalized configuration settings via web/phone user interface are saved in the MAC-local.cfg file on the phone,



Then the phone will upload the MAC-local.cfg file to the provisioning server. During auto provisioning, the phone requests to download the MAC-local.cfg file on the provisioning server.

0-Disabled, 1-Enabled. The default value is 0.

auto provision.custom.sync =

It enables or disables the phone to protect personalized configuration settings even after auto provisioning.

If it is set to 1 (Enabled), personalized configuration settings via web/phone user interface will be protected and remained after auto provisioning.

0-Disabled, 1-Enabled. The default value is 0.

auto provision.custom.protect =

For more information please refer to the

Yealink_W52P_IP_DECT_Phones_Deployment_Guide_for_BroadSoft_UC-One_En vironment_V73_2.

2. Added Off-hook hot line dialing feature.

Description: The account need to use through authentication, user should dial out a number for authentication, and the account can be use after the success of the input user name and password authentication.

The auto provision template are as follows:

#!version:1.0.0.1

auto_provision.custom

account.1.auto_dial_num =

account.1.auto_dial_enable =

For more information please refer to the

Yealink_W52P_IP_DECT_Phones_Deployment_Guide_for_BroadSoft_UC-One_En vironment V73 2.

3. Added the Backup Server.

Description: The W52P is designed to work in concert with the SIP Phone of version 72 that the user account can automatic switch to Backup server when the current server did not work.

The auto provision template are as follows:

#!version:1.0.0.1



```
account.1.reregister enable =
account.1.naptr_build =
account.1.fallback.redundancy_type =
account.1.fallback.timeout =
account.1.sip server.1.address =
account.1.sip_server.1.port =
account.1.sip_server.1.expires =
account.1.sip server.1.retry counts =
account.1.sip server.1.failback mode =
account.1.sip server.1.failback timeout =
account.1.sip_server.1.register_on_enable =
account.1.sip server.2.address =
account.1.sip server.2.port =
account.1.sip_server.2.expires =
account.1.sip_server.2.retry_counts =
account.1.sip server.2.failback mode =
account.1.sip_server.2.failback_timeout =
account.1.sip server.2.register on enable =
account.1.dns_cache_type =
account.1.dns_cache_a.1.name =
account.1.dns cache a.1.ip =
account.1.dns cache a.1.ttl =
account.1.dns cache srv.1.name =
account.1.dns_cache_srv.1.port =
account.1.dns cache srv.1.priority =
account.1.dns cache srv.1.target =
account.1.dns cache srv.1.weight =
account.1.dns cache srv.1.ttl =
account.1.dns cache naptr.1.name =
account.1.dns cache naptr.1.flags =
account.1.dns cache naptr.1.order =
account.1.dns cache naptr.1.preference =
account.1.dns cache naptr.1.replace =
```



```
account.1.dns cache naptr.1.service =
account.1.dns_cache_naptr.1.ttl =
account.1.dns cache type =
account.1.dns cache a.2.name =
account.1.dns_cache_a.2.ip =
account.1.dns cache a.2.ttl =
account.1.dns cache srv.2.name =
account.1.dns cache srv.2.port =
account.1.dns cache srv.2.priority =
account.1.dns_cache_srv.2.target =
account.1.dns cache srv.2.weight =
account.1.dns cache srv.2.ttl =
account.1.dns cache naptr.2.name =
account.1.dns_cache_naptr.2.flags =
account.1.dns_cache_naptr.2.order =
account.1.dns cache naptr.2.preference =
account.1.dns cache naptr.2.replace =
account.1.dns cache naptr.2.service =
account.1.dns_cache_naptr.2.ttl =
account.1.static_cache_pri =
```

4. Added support for the Broadsoft SIP XSI verifying feature.

Description: W52P IP DECT phones supports implementing the Xtended Services Interface (XSI) to access the BroadSoft directory and call log. To access the BroadSoft directory and call log, the IP phone must sends the XSI user authentication information to Xtended Services Platform (XSP) to authenticate the XSI user. IP phones running firmware version 73 or later support two XSI authentication methods:

- User Login Credentials for XSI Authentication: The IP phone uses the XSI user login credentials (web portal login user ID and password) for XSI authentication.
- SIP Credentials for XSI Authentication: As of BroadWorks release 20.0, the IP phone can use the XSI user ID along with SIP authentication credentials for XSI authentication. SIP credentials are the user name and password of



the SIP account registered on the first line of the phone. This method is not applicable to IP phones with firmware version prior to 73.

The auto provision template are as follows:

#!version:1.0.0.

```
sip.authentication_for_xsi =
xsi.port =
xsi.server_type =
```

For more information please refer to the

Yealink_W52P_IP_DECT_Phones_Deployment_Guide_for_BroadSoft_UC-One_En vironment_V73_2.

5. Added the support of wildcards (\$PN and \$MAC) in the provisioning server URL.

Description: when the phone tries to get IP address and DNS address via DHCP, it will also receive a fixed standard URL at the same time. For example, http://10.2.4.1/\$PN/\$MAC.cfg.

The auto provision template are as follows:

#!version:1.0.0.

auto provision.url wildcard.pn =

For more information please refer to the

Yealink_W52P_IP_DECT_Phones_Deployment_Guide_for_BroadSoft_UC-One_En vironment_V73_2.

6. Added support of Network Conference.

Description: This feature on the W52P is designed to work in concert with the SIP Phone of version 72.

The auto provision template are as follows:

#!version:1.0.0.1

```
account.X.conf_type =
account.X.conf_uri =
```

For more information please refer to the

Yealink_W52P_IP_DECT_Phones_Deployment_Guide_for_BroadSoft_UC-One_En vironment_V73_2.



6. Optimization description

- 1. Optimized the Intercom Call feature that it can now support for Metaswitch platform.
- 2. User can now cancel the handset of the registration via web interface.
- 3. Add the auto answer message format.

Description: Auto Answer now works correctly when the receiving message form as Alert-Info: Auto Answer and Call-Info: answer-after=0

4. Optimized the behavior problems after Enable/Disable of LLDP.

Description: The phone now acquires the correct VLAN either in the environment of the LLDP or VLAN.

- 5. Optimized the computer response speed when user checking the contacts and telephone records.
- 6. Optimized the display experience of Broadsoft contacts.

Description: The phone automatically deletes the same contacts in the Group and Enterprise.

7. Optimized the display experience of call log.

Description: The phone clock format (12 hrs – 24hrs) can now be synchronized to phone Call log.

8. Optimized the verification of URL address/Name/Password on the LCD interface.

Description: Added the autop code feature that the username and password of auto provision URL can now be verified via the LCD interface. At the same time, the web interface will make automatic synchronization of these username and password in the server URL field.

9. Added the Russian language.

Description: Added the language support of Russian on the base and web interface.

10. Unified the auto provisioning files (Com.cfg/MAC.cfg).

Descrption: The auto provisioning files were unified in concert with the SIP Phone of V73.

For more information please refer to the

W52P_Auto_Provision_Template_files_Changelog_v1 1

11. Optimized the feature of Busy Tone.

Description: Busy tone can be configured by the following configuration



parameters.

The auto provision template are as follows:

#!version:1.0.0.1

features.busy_tone_delay =

12. Optimized the feature of Trouble Shooting.

Description: User can quickly positioning problem through deriving the Syslog or caught. When the user configures the System Log Level through web interface, the phone would not restart, but when the user configures it by auto provisioning, the phone will still restart.

13. Optimized the sequence of codec.

Description: Optimized the sequence of G722 codec to the highest, the phone will prefer using G722 codec to receive audio and the sound quality is better.

14. Optimized the control of answering/rejecting anonymous calls on the handset interface.

Description: Added support of configuring parameter to control Anonymous call and anonymous call rejection feature.

The auto provision template are as follows:

#!version:1.0.0.1

account.X.anonymous_call_mode(X ranges from 1 to 5.)

##It configures the anonymous call feature mode when users configure anonymous call feature via the handset for account X.

0-Local

1-Server

The default value is 0.

For more information please refer to the

Yealink_W52P_IP_DECT_Phones_Deployment_Guide_for_BroadSoft_UC-One_En vironment V73 2.

15. As of firmware version 73, some of the Broadsoft features are only applicable to the UC-One firmware version.

Description: As of firmware version 73, the following features are only applicable to the UC-One firmware version (25/26.73.193.x.rom). The following sections are deleted in this version:

- Network Directory
- Network Call Log
- Shared Call Appearance

To preserve backward compatibility of firmware version 50, the above features



will not be removed from the phone when upgrading the old firmware version to the firmware version 73 or later, unless you reset the phone to factory defaults.

16. Optimized the feature of Time zone.

Description: The user can configure separately with the time zone and time zone name via auto provisioning, only in the daylight saving time, the time zone name would be used to determine the daylight saving time belongs to which country or region.

The Time zone and daylight-saving time are set up separately on the web UI/LCD/Auto provisioning, and the optional values of time zones was optimized at the same time.

The available time zones which on the IP phone was now added every half an hour from -12 to +13, and the +5:45 and +12:45 time zone was added at the same time.

17. As of firmware version 73 and the UC-One firmware version the RPS feature was enable by default.

18. Optimize the standby time of the battery.

Description: The standby time of handset has extended by 20%.

19. File encryption of config.bin.

Description: In order to protect user information better, in the newly version of 73 or later, the file of config.bin was now encrypted, and it was only supported to be upload with new version but not the old one.

20. Modified the Daylight Saving Time in the file of DST.

Description: The Daylight Saving Time of 7 US-Mountain Time in the file of DST is now to be modified as iType="1" szStart="3/2/7/2" szEnd="11/1/7/2" szOffset="60"

7. Default Value Setting Changes

| Default Val | Default Value Changelog from V50 to V73 | | | | | |
|-------------|--|----------------------------------|---|--|--|--|
| Feature | Description | Default Value of Factory Setting | | | | |
| | | x.50.193.1 | x.73.193.5 | | | |
| VLAN | It enables or disables the IP phone to obtain IP address with lower preference of VLAN assignment method or disable VLAN feature when the IP phone cannot obtain IP address with the current VLAN assignment method. | None | network.vlan.vlan_change.en able = 0 | | | |



| 0 -Disabled | |
|---|---------------------------------------|
| 1-Enabled | |
| | |
| The default value is 0. | |
| It enables or disables the IP phone to | |
| periodically | |
| (every 5 minutes) upload the | |
| <mac>-local.cfg file</mac> | |
| CustomProt to the provisioning server and | auto_provision.custom.sync |
| download the None | = 0 |
| <mac>-local.cfg file from the</mac> | |
| provisioning server during auto | |
| provisioning. | |
| 0-Disabled | |
| 1-Enabled | |
| Enables or disables the IP phone to | |
| protect personalized settings after | |
| auto provisioning. | |
| 0-Disabled | |
| CustomProt 1-Enabled | auto_provision.custom.prote |
| If it is set to 1 (Enabled), personalized None | c = 0 |
| settings of the base station configured | |
| via the handset or web user interface | |
| and the handset settings configured | |
| via the handset will be remained after | |
| auto provisioning. | |
| Configures the way the IP phone | |
| uploads the <mac>-local.cfg file to the</mac> | |
| CustomProt provisioning server (for HTTP/HTTPS None | auto_provision.custom.uploa |
| ect server only). | d_method = 0 |
| 0-PUT | |
| 1-POST | |
| It configures the host name of the IP | |
| DHCP phone for DHCP option 12. option1 | 2_value = network.dhcp_host_name = |
| Hostname The default value is SIP-W52P. | SIP-W52P |
| It takes effect after a reboot. | |
| It enables or disables 100 reliable | |
| 100 reliable retransmission feature for account X. | |
| | 100rel = ?1 account.X.100rel_enable = |
| ion 1-Enabled | |
| The default value is 0. | |
| CallNumFilt It configures the characters the IP | |
| phone filters when dialing. None | features.call_num_filter = - |
| If the dialed number contains | |



| | and and the section of the section of | | |
|--------------|--|--------------------|-------------------------------|
| | configured characters, the IP phone | | |
| | will automatically filter these | | |
| | characters when dialing. | | |
| | The default value is "-,". | | |
| | It configures the time format. | | |
| | 0-12 Hour | | |
| | 1-24 Hour | | |
| | If it is set to 0 (12 Hour), the time will | | |
| | be displayed in | | |
| DateFormat | 12-hour format with AM or PM | None | custom.handset.time_format |
| Dater or mat | specified. | None | = 1 |
| | If it is set to 1 (24 Hour), the time will | | |
| | be displayed in | | |
| | 24-hour format (e.g., 2:00 PM displays | | |
| | as 14:00). | | |
| | The default value is 1. | | |
| | It configures the begin time of the day | | |
| | for the IP phone to perform an auto | strWeeklyBeginTime | auto_provision.schedule.tim |
| autop | provisioning process weekly. | = ?02:00 | e_from = 00:00 |
| | The default value is 00:00. | | _ |
| | | | |
| | It configures the end time of the day | | |
| autop | for the IP phone to perform an auto | strWeeklyEndTime | auto_provision.schedule.tim |
| | provisioning process weekly. | = ?03:00 | e_to = 00:00 |
| | The default value is 00:00. | | |
| | It configures the authentication | | |
| | method the IP phone uses for XSI | | |
| | access. | | |
| | 0- User Login Credentials for XSI | | |
| | Authentication | | |
| | 1-SIP Credentials for XSI | | |
| | Authentication | | sin authoritisation for vsi - |
| Broadsoft | If it is set to 0, the IP phone uses the | None | sip.authentication_for_xsi = |
| | XSI user ID and password for XSI | | 0 |
| | authentication. | | |
| | If it is set to 1, the IP phone uses the | | |
| | XSI user ID, the user name and | | |
| | password of the SIP account 1 for XSI | | |
| | authentication. | | |
| | The default value is 0. | | |
| | It configures the access port of the | | |
| BroadSoft | Xtended Services | Nege | wei namt 00 |
| Phonebook | Platform server. | None | xsi.port = 80 |
| | Example: | | |
| | · · | | |



| | xsi.port = 80 | | |
|----------------------------------|---|---|---|
| | The default value is 80. | | |
| BroadSoft Phonebook | It configures the access protocol of the Xtended Services Platform server. Example: xsi.server_type = http The default value is http. | None | xsi.server_type = http |
| Busy Tone | It configures the duration time (in seconds) for the busy tone. When one party releases the call, a busy tone is audible to the other party indicating that the call connection breaks. If it is set to 3 (3s), a busy tone is audible for 3 seconds on the IP phone. The default value is 0. | None | features.busy_tone_delay = 0 |
| BroadSoft | It enables or disables BroadSoft features for IP phones. 0-Disabled 1-Enabled The default value is 0 for the neutral firmware version (x.73.0.x), and 1 for the UC-One firmware version (x.73.193.x). It takes effect after a reboot. | bw.enable = 1 | bw.enable =0 bw.enable =1(BSFT UC ONE) |
| codec | It configures the priority of the enabled codec for account X. It configures the anonymous call feature mode when users configure | account.1.codec.1.priorit y = 1 account.1.codec.2.priorit y = 2 account.1.codec.5.priorit y = 3 account.1.codec.6.priorit y = 4 | account.x.codec.1.priority = 2 account.x.codec.2.priority = 3 account.x.codec.5.priority = 4 account.x.codec.6.priority = 1 |
| anonymous _call_mode reject_anon | anonymous call feature via the handset for account X. 0-Local 1-Server The default value is 0. It configures the anonymous call | None | account.x.anonymous_call_ mode = 0 account.x.anonymous_call_ |
| ymous_call | feature mode when users configure | None | mode = 0 |



| _mode | anonymous call feature via the | | | |
|------------|---------------------------------------|------|-------------------------|--|
| | handset for account X. | | | |
| | 0-Local | | | |
| | 1-Server | | | |
| | The default value is 0. | | | |
| | It configures the conference type for | | | |
| | account X. | | | |
| Network | 0-Local Conference (Conferences are | | account.x.conf_type = 0 | |
| Conference | set up on the IP phone locally). | None | | |
| Comerence | 2-Network Conference (Conferences | | | |
| | are set up by the server). | | | |
| | The default value is 0. | | | |
| | It configures the network conference | | | |
| Network | URI for account | None | account y conf uri - | |
| Conference | X. | None | account.x.conf_uri = | |
| | The default value is blank. | | | |

8. Configuration Parameters Enhancements

| Firmware Ve | ersion: [X.50.193 | .1]-[X.73.193.5] | | | | |
|----------------------|---|----------------------------------|---------------------|----------|--|------|
| Feature | Provisioning syr | ntax | Permitted Values | Action | Description | File |
| | V73 | V50 | | | | |
| Auto Provisioning | auto_provision.p ower_on = | auto_provision. mode = | 0 or 1 | Modified | Enables or disables the IP phone to perform an auto provisioning process when powered on. | COM |
| Auto Provisioning | auto_provision.w eekly.dayofweek = | auto_provision. weekly.mask = | 0 or 1 | Modified | Configures the days of the week for the IP phone to check new configuration weekly. | COM |
| Auto Provisioning | auto_provision.c ustom.protect = | None | 0 or 1 | Added | Enables or disables the IP phone to protect personalized settings after auto provisioning. | COM |
| Auto Provisioning | auto_provision.c ustom.upload_m ethod = | None | 0 or 1 | Added | Description: Configures the way the IP phone uploads the <mac>-local.cfg</mac> | COM |



| Phone | sip.send_respons | | | | provisioning. It configures from | СОМ |
|----------------------|--|------|-----------------------------------|-------|--|-----|
| Auto Provisioning | auto_provision.r eboot_force.ena ble = | None | 0 or 1 | Added | It enables or disables the IP phone to reboot after auto provisioning when there is no specific configuration requiring reboot. It is especially useful when there is no specific configuration requiring reboot in the configuration files, but you want to force the IP phone to reboot after auto | COM |
| Auto Provisioning | auto_provision.u rl_wildcard.pn = | None | String within 32 characters | Added | It configures the characters to replace the wildcard \$PN and \$MAC in the received URL of the provisioning server. The default value is W52P. | СОМ |
| Auto Provisioning | auto_provision.c ustom.sync = | None | 0 or 1 | Added | (for HTTP/HTTPS server only). 0-PUT 1-POST Enables or disables the IP phone to periodically (every 5 minutes) upload the <mac>-local.cfg file to the provisioning server, and download the <mac>-local.cfg file from the provisioning server during auto provisioning.</mac></mac> | COM |
| | | | | | file to the provisioning server | |



| 1 | | | | | |
|--------------------------------|--|--|--|--|---|
| | | | | retrieves the | |
| | | | | | |
| | | | | for response. The IP | |
| | | | | phone will then send | |
| | | | | all SIP response | |
| | | | | messages to the | |
| | | | | destination address. | |
| | | | | It enables or disables | сом |
| sip.use_23_as_p | None | 0 or 1 | Addad | the IP phone to | |
| ound = | None | 0011 | Audeu | reserve the pound | |
| | | | | sign when dialing out. | |
| | | | | It enables or disables | сом |
| sin wfs2E42 hold | | | | the IP phone to use | |
| _ | None | 0 or 1 | Added | RFC 2543 (c=0.0.0.0) | |
| = | | | | outgoing hold | |
| | | | | signaling. | |
| | | | | It configures the | сом |
| | | | | phone behavior when | |
| sip.notify_reboot _enable = | Nisas | 0.42 | A -1 -11 | the IP phone receives | |
| | None | 0, 10r 2 | Added | a SIP NOTIFY message | |
| | | | | containing the header | |
| | | | | "Event: check-sync". | |
| | | | | It configures the | сом |
| | | | | emergency numbers. | |
| phone_setting.e | | String | | Multiple emergency | |
| mergency.numbe | None | within 99 | Added | numbers are | |
| r = | | characters | | separated by commas. | |
| | | | | The default value is | |
| | | | | blank. | |
| factcias | | | | It enables or disables | сом |
| | Nisas | 0 1 | A -1 -11 | the IP phone to play a | |
| | None | 0 or 1 | Added | warning tone when | |
| = | | | | receiving a voice mail. | |
| | | | | It configures the | сом |
| | | | | duration time (in | |
| | | | | seconds) of the busy | |
| | | | | tone. When one party | |
| faction 1 | | | | releases the call, a | |
| . — | None | 0, 3 or 5 | Added | busy tone is audible | |
| ne_delay = | | | | to the other party | |
| | | | | indicating that the call | |
| | | | | connection breaks. | |
| | | | | If it is set to 3 (3s), a | |
| | | | | busy tone is audible | |
| | <pre>ound = sip.rfc2543_hold = sip.notify_reboot _enable = phone_setting.e mergency.numbe</pre> | sip.rfc2543_hold = None sip.notify_reboot _enable = None phone_setting.e mergency.numbe r = None features.voice_m ail_tone_enable = None features.busy_to None | sip.rfc2543_hold = None 0 or 1 sip.notify_reboot _enable = None 0, 1or 2 phone_setting.e mergency.numbe r = String within 99 characters features.voice_m ail_tone_enable = 0 or 1 features.busy_to None 0.3 or 5 | ound = None 0 or 1 Added sip.rfc2543_hold = None 0 or 1 Added sip.notify_reboot enable = None 0, 1or 2 Added phone_setting.e mergency.numbe r = None String within 99 characters Added features.voice_m ail_tone_enable = None 0 or 1 Added features.busy_to None 0 or 1 Added | destination address for response. The IP phone will then send all SIP response messages to the destination address. Sip.use_23_as_p ound = |



| | | | | | for 3 seconds on the | |
|----------------------|---------------------------------|------|-----------------------------------|-------|--|-------|
| | | | | | IP phone. | |
| Phone Features | features.direct_i p_call_enable | None | 0 or 1 | Added | It enables or disables the IP phone to place direct IP calls. A direct IP call is placed by dialing the IP address of the destination phone. | СОМ |
| Jitter Buffer | voice.jib.adaptiv e = | None | 0 or 1 | Added | It configures the type of jitter buffer. | COIVI |
| Jitter Buffer | voice.jib.min = | None | Integer | Added | It configures the minimum delay (in milliseconds) of jitter buffer. The default value is 0. | СОМ |
| Jitter Buffer | voice.jib.max = | None | Integer | Added | It configures the maximum delay (in milliseconds) of jitter buffer. | СОМ |
| Jitter Buffer | voice.jib.normal = | None | | Added | It configures the normal delay (in milliseconds) of jitter buffer. The default value is 120. | СОМ |
| Echo Cancellation | voice.vad = | None | 0 or 1 | Added | It enables or disables VAD (Voice Activity Detection) feature on the phone. | СОМ |
| Echo Cancellation | voice.cng = | None | 0 or 1 | Added | It enables or disables CNG (Comfortable Noise Generator) feature on the phone. | СОМ |
| Security Settings | security.user_na me.user = | None | String within 32 characters | Added | It configures the user name of the user for phone's web user interface access. The default value is user. | COM |
| Security Settings | security.user_na me.admin = | None | String within 32 characters | Added | It configures the user name of the administrator for phone's web user | СОМ |



| Contacts | bw_phonebook. group_common_ | None | String within 99 | Added | display name on the | COIVI |
|----------------------|------------------------------------|------|-----------------------------------|-------|--|-------|
| Contacts | bw_phonebook. group_displayna me = | None | String within 99 characters | Added | It configures the display name on the LCD screen for the group directory. The default value is Group. It configures the | СОМ |
| Contacts | xsi.server_type = | None | "http" or "https" | Added | It configures the access protocol of the Xtended Services Platform server. | СОМ |
| Contacts | xsi.port = | None | Integer from 1 to 65535 | Added | It configures the port of the Xtended Services Platform server. | СОМ |
| Broadsoft Feature | bw.enable = | None | 0 or 1 | Added | It enables or disables the Broadsoft feature. | СОМ |
| User Mode | web_item_level. url = | None | URL within 511 characters | Added | It configures the access URL of the file which defines 3-level access permissions. The default value is blank. It takes effect after a reboot. For more information, refer to Yealink IP phones Configuration Guide for User Access Level. | COM |
| User Mode | security.var_ena ble = | None | 0 or 1 | Added | It enables or disables the 3-level permissions (admin, user, var). | СОМ |
| Security Settings | security.user_na me.var = | None | String within 32 characters | Added | admin. It configures the user name of the var for phone's web user interface access. The default value is var. | СОМ |
| | | | | | interface access. The default value is | |



| | displayname = | | characters | | LCD screen for the | |
|-----------|-------------------|--------|------------------|------------|-------------------------|-------|
| | uispiayilaille = | | citatacters | | | |
| | | | | | group common | |
| | | | | | directory. | |
| | | | | | The default value is | |
| | | | | | GroupCommon. | |
| | | | | | It configures the | COM |
| | bw_phonebook. | | String | | display name on the | |
| Contacts | enterprise_displa | None | within 99 | Added | LCD screen for the | |
| | yname = | | characters | | enterprise directory. | |
| | , | | | | The default value is | |
| | | | | | Enterprise. | |
| | | | | | It configures the | СОМ |
| | bw_phonebook. | | | | display name on the | |
| | enterprise_com | | String | | LCD screen for the | |
| Contacts | mon_displaynam | None | within 99 | Added | enterprise common | |
| | e = | | characters | | directory. | |
| | | | | | The default value is | |
| | | | | | EnterpriseCommon. | |
| | | | | | It configures the | сом |
| | h mhanahaali | | Chuin | | display name on the | |
| Cambasta | bw_phonebook. | None | String within 99 | Added | LCD screen for the | |
| Contacts | personal_display | None | | Added | personal directory. | |
| | name = | | characters | | The default value is | |
| | | | | | Personal. | |
| | | | | | It configures the | сом |
| | sip.authenticatio | | 0 or 1 | Added | authentication | |
| Contacts | n_for_xsi = | None | | | method the IP phone | |
| | | | | | uses for XSI access. | |
| | | | | | Enables or disables | сом |
| | | | | | the base station to | |
| | auto_provision.h | | | | deliver handset | |
| Handset | andset configure | None | 0 or 1 | Added | settings configured | |
| | d.enable = | | | | via auto provisioning | |
| | | | | | to the registered | |
| | | | | | handset. | |
| | | | | | It configures the | СОМ |
| Handset | custom.handset.l | None | Integer | Added | language of the | 20171 |
| Tidilaset | anguage= | 14011C | from 0 to 11 | Added | handset | |
| | | | | | It configures the color | СОМ |
| Handset | custom.handset. | None | 0 or 1 | Added | scheme of the | COIVI |
| Tianuset | color_scheme = | INUITE | 0011 | Audeu | handset. | |
| | | | | | | 6014 |
| Hondest | custom.handset. | None | 0.12 | ا ما ما ما | It configures whether | COM |
| Handset | auto_intercom = | None | 0, 1 or 2 | Added | the IP phone | |
| | | | | | automatically answers | |



| Handset | missed_call_noti fy_light.enable = | None | 0 or 1 | Added | the message key LED to flash red when the | |
|-----------|---------------------------------------|--------|--------|------------|--|-------|
| | custom.handset. | | | | It enables or disables | СОМ |
| | | | | | voice mail. | |
| Tanaset | y_light.enable = | INOTIC | 0011 | Added | handset receives a | |
| Handset | custom.handset. voice_mail_notif | None | 0 or 1 | Added | the message key LED to flash red when the | |
| | anakana kan l | | | | It enables or disables | СОМ |
| | | | | | batteries is low. | |
| | .enable = | | | | capacity of the | |
| Handset | ow_battery_tone | None | 0 or 1 | Added | tone when the | |
| | custom.handset.l | | | | the handset to play a | |
| | | | | | It enables or disables | СОМ |
| | | | | | the charger cradle. | |
| | ne.enable = | | | | places the handset in | |
| Handset | confirmation_to | None | 0 or 1 | Added | saves settings or | |
| | custom.handset. | | | | tone when a user | |
| | | | | | the handset to play a | |
| | | | | | It enables or disables | СОМ |
| | ble = | | | | pressed. | |
| Handset | keypad_tone.ena | None | 0 or 1 | Added | tone when any key is | |
| | custom.handset. | | | | the handset to play a | COIVI |
| | | | | | It enables or disables | СОМ |
| | DIE – | | | | when any key is pressed. | |
| rianuset | ble = | INUITE | 0 or 1 | Added | | |
| Handset | keypad_light.ena | None | | | on the keypad light | |
| | custom.handset. | | | | It enables or disables the handset to turn | СОМ |
| | | | | | charging state. | CON4 |
| | = | | | | when it is not in the | |
| Handset | _charger.enable | None | 0 or 1 | Added | turn on the backlight | |
| llowds st | backlight_out_of | None | 0.07.1 | ا ما ما ما | the handset to always | |
| | custom.handset. | | | | It enables or disables | СОМ |
| | | | | | charging state. | |
| | rger.enable = | | | | when it is in the | |
| Handset | backlight_in_cha | None | 0 or 1 | Added | turn on the backlight | |
| | custom.handset. | | | | the handset to always | |
| | | | | | It enables or disables | СОМ |
| | ble = | | | | | |
| Handset | screen_saver.ena | None | 0 or 1 | Added | screen saver feature. | |
| | custom.handset. | | | | It enables or disables | сом |
| | | | | | plays a warning tone. | |
| | | | | | intercom call and | |
| | | | | | an incoming internal | |



| | | | | | handset misses a call. | |
|------------------|---|------|-----------------------------------|-------|---|-----|
| Handset | custom.handset. auto_answer.ena ble = | None | 0 or 1 | Added | It enables or disables a user to answer incoming calls by lifting the handset from the charger cradle without having to press the off-hook key. | СОМ |
| Handset | custom.handset. eco_mode.enabl e = | None | 0 or 1 | Added | It enables or disables the eco mode to greatly reduce the transmission power and signal output when the phone is in the talk mode. | СОМ |
| ECO | base.eco_mode. enable= | None | 0 or 1 | Added | It enables or disables the Eco mode+ to turn off the transmission power when the phone is in the standby mode. | COM |
| Repeater | base.repeater_m ode.enable= | None | 0 or 1 | Added | It enables or disables the repeater mode to extend the radio coverage of the base station. | СОМ |
| Features DTMF | features.dtmf.hid e = | None | 0 or 1 | Added | It enables or disables the IP phone to suppress the display of DTMF digits. If it is set to 1 (Enabled), the DTMF digits are displayed as asterisks. | СОМ |
| Features DTMF | features.dtmf.hid e_delay = | None | 0 or 1 | Added | It enables or disables the IP phone to display the DTMF digits for a short period before displaying asterisks. | СОМ |
| DHCP Host | network.dhcp_h ost_name = | None | String within 99 characters | Added | It configures the host name of the IP phone for DHCP option 12. | СОМ |



| | | | | | The default value is SIP-W52P. It takes effect after a reboot. | |
|---|--------------------------------------|---|---------------------------------|---------|---|-----|
| Network | network.802_1x. client_cert_url = | None | | Added | Configure the username and password for 802.1x authentication | СОМ |
| Auto Provisioning | | auto_provision. pnp_domain_n ame = | | Deleted | | СОМ |
| Auto Provisioning | | auto_provision. pnp_event_ven dor = | | Deleted | | СОМ |
| Auto Provisioning | | auto_provision. common_file_n ame = | | Deleted | | СОМ |
| Phone Settings | | features.fwd.all ow = | 0 or 1 | Deleted | | СОМ |
| Phone Settings | | features.dnd.all ow = | 0 or 1 | Deleted | | СОМ |
| Customized Factory Configuration s | configuration.url = | | URL within 511 characters | Added | It configures the access URL for the custom config files. The default value is blank. It takes effect after a reboot. | СОМ |
| Customized Factory Configuration s | custom_mac_cfg .url = | | URL within 511 characters | Added | It configures the access URL of the custom MAC-Oriented CFG file. | СОМ |
| SIP server | | account.1.sip_s erver_host | | Deleted | The default value is blank. | MAC |
| Account1 Settings | | account.1.sip_s erver_port = | | Deleted | | MAC |
| Account1 Settings | | account.1.back up_outbound_ host = | | Deleted | | MAC |
| Account1 Settings | | account.1.back up_outbound_ port = | | Deleted | | MAC |
| Account1 Settings | | account.1.prec ondition = | | Deleted | | MAC |



| Account1 | | account.1.enab le_signal_enco | | Deleted | | MAC |
|---------------------------|---|-----------------------------------|-----------------------------------|---------|---|-----|
| Settings | | de | | | | |
| Account1 Settings | | account.1.signa I_encode_key = | | Deleted | | MAC |
| SCA | account.1.shared _line = | | 0 or 1 | Added | It enables or disables BroadSoft shared call appearance feature. 0-Disabled 1-BroadSoft_SCA The default value is 0. | MAC |
| Account1 Settings | account.1.anony mous_call_mode = | None | 0 or 1 | Added | It configures the anonymous call feature mode when users configure anonymous call feature via the handset for account X. O-Local 1-Server The default value is 0. | MAC |
| Account1 Settings | account.1.reject_ anonymous_call _mode = | None | 0 or 1 | Added | It configures the anonymous call rejection mode when users configure anonymous call rejection feature via the handset for account X. O-Local 1-Server The default value is 0. | MAC |
| Account1 Settings | account.1.send_ anonymous_cod e = | None | 0 or 1 | Added | Configure the on code and off code of the anonymous call feature | MAC |
| Account1 Settings | account.1.send_ anonymous_reje ction_code = | None | 0 or 1 | Added | Configure the on code and off code of the reject anonymous call feature | MAC |
| Off Hook hot line dialing | account.1.auto_ dial_num = | None | String within 32 characters | Added | It configures the number that the phone automatically | MAC |



| | | | | | dials out first when the user places a call. The default value is blank. | |
|------------------------------|---|------|--------|-------|---|-----|
| Off Hook hot line dialing | account.1.auto_ dial_enable = | None | 0 or 1 | Added | It enables or disables the phone to automatically dial out a pre-configured number first when the user places a call using the account X. O-Disabled 1-Enabled The default value is 0. | MAC |
| Account1 Settings | account.1.cid_so urce_privacy = | None | 0 or 1 | Added | It enables or disables the IP phone to process the PRIVACY header field in the 180 or 200 OK message for account X. 0-Disabled 1-Enabled The default value is 1. | MAC |
| Account1 Settings | account.1.cid_so urce_ppi = | None | 0 or 1 | Added | It enables or disables the IP phone to process the P-Preferred-Identity header for caller identity presentation when receiving an incoming call for account X. 0-Disabled 1-Enabled The default value is 1. | MAC |
| Account1 Settings | account.1.cp_so urce = | None | 0 or 1 | Added | It configures the presentation of the callee's identity for account X. | MAC |
| Failback | account.X.sip_se rver.Y.register_on _enable = | None | 0 or 1 | Added | It enables or disables the IP phone to send registration requests to the secondary | MAC |



| | | | | | server for account X when encountering a failover. 0-Disabled 1-Enabled The default value is 0. | |
|----------|--|------|-------------------------------------|-------|---|-----|
| Failback | account.1.naptr_ build = | None | 0 or 1 | Added | It configures the way of SRV query for the IP phone to be performed when no result is returned from NAPTR query for account X. 0-SRV query using UDP only 1-SRV query using UDP, TCP and TLS. The default value is 0. | MAC |
| Failback | account.1.fallbac k.redundancy_ty pe = | None | 0 or 1 | Added | It configures the registration mode for account X in fallback mode. 0-Concurrent registration 1-Successive registration The default value is 0. | MAC |
| Failback | account.1.fallbac k.timeout = | None | Integer from 10 to 2147483647 | Added | It configures the time interval (in seconds) for the IP phone to detect whether the working server is available by sending the registration request for account X after the fallback server takes over call control. It is only applicable to successive registration mode. The default value is 120. | MAC |



| Failback | account.1.sip_se rver.1.address = | None | IP address or domain name | Added | It configures the IP address or domain name of the SIP server Y for account X. | MAC |
|----------|---|------|-------------------------------------|-------|--|-----|
| Failback | account.1.sip_se rver.1.port = | None | Integer from 1 to 65535 | Added | It configures the port of SIP server Y for account X. The default value is 5060. Note: The old parameter "account.X.sip_server _port" is also applicable to IP phones. | MAC |
| Failback | account.1.sip_se rver.1.expires = | None | Integer from 30 to 2147483647 | Added | It configures the registration expiration time (in seconds) of SIP server Y for account X. The default value is 3600. | MAC |
| Failback | account.1.sip_se rver.1.retry_coun ts = | None | Integer from 0 to 20 | Added | It configures the retry times for the IP phone to resend requests when the SIP server Y is unavailable or there is no response from the SIP server Y for account X. The default value is 3. | MAC |
| Failback | account.1.sip_se rver.1.failback_m ode = | None | 0, 1, 2 or 3 | Added | It configures the mode for the IP phone to retry the primary server in failover mode for account X. | MAC |
| Failback | account.1.sip_se rver.1.failback_ti meout = | None | Integer 0, from 60 to 65535 | Added | It configures the timeout (in seconds) for the IP phone to retry to send requests to the primary server after failing over to the current working | MAC |



| Failback | che_a.1.ip = | None | IP address | Added | record Y maps to for account X. | MAC |
|----------|---|------|----------------|-------|--|-----|
| | account.1.dns_ca | | | | address that the domain name in A | |
| Failback | account.1.dns_ca che_a.1.name = | None | Domain name | Added | It configures the domain name in A record Y for account X. The default value is blank. It configures the IP | MAC |
| Failback | account.1.dns_ca che_type = | None | 0, 1 or 2 | Added | It configures the content that the DNS cache records for account X. O-perform real-time DNS query rather than using DNS cache. 1-Use DNS cache, but do not record the additional records. 2-Use DNS cache and record the additional records. The default value is 1. | MAC |
| Failback | account.1.sip_se rver.1.register_o n_enable = | None | 0 or 1 | Added | Y.failback_mode" is set to 3 (duration). It enables or disables the IP phone to send registration requests to the secondary server for account X when encountering a failover. O-Disabled 1-Enabled The default value is 0. | MAC |
| | | | | | server for account X when the value of the parameter "account.X.sip_server. | |



| | che_a.1.ttl = | | from 30 to 2147483647 | | interval (in seconds) that A record Y may be cached before the record should be consulted again for account X. The default value is 300. | |
|----------|---|------|-------------------------------------|-------|--|-----|
| Failback | account.1.dns_ca che_srv.1.name = | None | Domain name | Added | It configures the domain name in SRV record Y for account X. It takes effect after a reboot. | MAC |
| Failback | account.1.dns_ca che_srv.1.port = | None | Integer from 0 to 65535 | Added | It configures the port to be used in SRV record Y for account X. The default value is 0. | MAC |
| Failback | account.1.dns_ca che_srv.1.priority = | None | Integer from 0 to 65535 | Added | It configures the priority for the target host in SRV record Y for account X. Lower priority is more preferred. The default value is 0. | MAC |
| Failback | account.1.dns_ca che_srv.1.target = | None | Domain name | Added | It configures the domain name of the target host for an A query in SRV record Y for account X. The default value is blank. It takes effect after a reboot. | MAC |
| Failback | account.1.dns_ca che_srv.1.weight = | None | Integer from 0 to 65535 | Added | It configures the weight of the target host in SRV record Y for account X. | MAC |
| Failback | account.1.dns_ca che_srv.1.ttl = | None | Integer from 30 to 2147483647 | Added | It configures the time interval (in seconds) that SRV record Y may be cached before the record should be | MAC |



| Failback | account.1.dns_ca che_naptr.1.servi ce = | None | String within 32 characters | Added | It configures the transport protocol available for the SIP server in NAPTR | MAC |
|----------|--|------|-----------------------------------|-------|---|-----|
| Failback | account.1.dns_ca che_naptr.1.repl ace = | None | Domain name | Added | It configures a domain name to be used for the next SRV query in NAPTR record Y for account X. The default value is blank. | MAC |
| Failback | account.1.dns_ca che_naptr.1.pref erence = | None | Integer from 0 to 65535 | Added | It configures the preference of NAPTR record Y for account X. NAPTR record with lower preference is more preferred. The default value is 0. | MAC |
| Failback | account.1.dns_ca che_naptr.1.orde r = | None | Integer from 0 to 65535 | Added | It configures the order of NAPTR record Y for account X. NAPTR record with lower order is more preferred. The default value is 0. | MAC |
| Failback | account.1.dns_ca che_naptr.1.flags = | None | S, A, U or P | Added | It configures the flag of NAPTR record Y for account X. (Always "s" for SIP, which means to do an SRV lookup on whatever is in the replacement field) | MAC |
| Failback | account.1.dns_ca che_naptr.1.nam e = | None | Domain name | Added | It configures the domain name to which NAPTR record Y refers for account X. The default value is blank. It takes effect after a reboot. | MAC |
| | | | | | consulted again for account X. The default value is 300. | |



| | | | | | record Y for account X. The default value is blank. | |
|--------------|---|------|-------------------------------------|-------|--|-----|
| Failback | account.1.dns_ca che_naptr.1.ttl = | None | Integer from 30 to 2147483647 | Added | It configures the time interval (in seconds) that NAPTR record Y may be cached before the record should be consulted again for account X. The default value is 300. | MAC |
| Failback | account.1.static_ cache_pri = | None | 0 or 1 | Added | It configures whether preferentially to use the static DNS cache for domain name resolution of the SIP server for account X. | MAC |
| Call Forward | account.1.always _fwd.off_code = | None | | Added | 0-Use domain name resolution from server | MAC |
| Call Forward | account.1.always _fwd.on_code = | None | | Added | Configures the always forward on code to activate the server-side always forward feature for account X. | MAC |
| Call Forward | account.1.busy_f wd.off_code = | None | | Added | Configures the busy forward off code to deactivate the server-side busy forward feature for account X. | MAC |
| Call Forward | account.1.busy_f wd.on_code = | None | | Added | Configures the busy forward on code to activate the server-side busy forward feature for account X. | MAC |
| Call Forward | account.1.timeo ut_fwd.off_code = | None | | Added | Configures the no answer forward off code to activate the server-side no answer forward feature for | MAC |



| | | | | | account X. | |
|-----------------------|--|-------------------------|--------|-------------|---|-----|
| Call Forward | account.1.timeo ut_fwd.on_code = | None | | Added | Configures the no answer forward on code to activate the server-side no answer forward feature for account X. | MAC |
| DND Mode | features.dnd_mo de = | features.dnd_ mode = | | Remove d | Configures the DND mode for the IP phone. If it is set to 0 (Phone), DND feature is effective for the IP phone. | MAC |
| DND | account.1.dnd.o n_code = | None | | Added | Configures the DND on code to activate the server-side DND feature for account X. | MAC |
| DND | account.1.dnd.of f_code = | None | | Added | Configures the DND off code to deactivate the server-side DND feature for account X. | MAC |
| Network Conference | account.1.conf_t ype = 0 | None | 0 or 2 | Added | ##It configures the conference type for account X.0-Local Conference,2-Networ k Conference. The default value is 0. | MAC |
| Network Conference | account.1.conf_u ri = | None | | Added | It configures the network conference URI for account X. The default value is blank. | MAC |
| Time Settings | local_time.manu al_time_enable = | None | 0 or 1 | Added | It configures the phone to obtain time from NTP server or manual settings.0-Manual,1-N | MAC |
| Time Settings | local_time.manu al_ntp_srv_prior = | None | 0 or 1 | Added | It enables or disables the phone to use manually configured NTP server preferentially | MAC |





| | time = | | | to | to start DST. | |
|---------------|------------------------------|------|---------------------------------|------------------------|--|-----|
| | | | | MAC.cfg | | |
| Time Settings | local_time.end_t ime = | None | Time | Moved to MAC.cfg | It configures the time to end DST. | MAC |
| Time Settings | local_time.offset _time = | None | Integer from -300 to 300 | Moved to MAC.cfg | It configures the offset time (in minutes). The default value is 60. | MAC |
| Time Settings | local_time.dhcp_ time = | None | 0 or 1 | Moved to MAC.cfg | It enables or disables the phone to update time with the offset time obtained from the DHCP server. It is only available to the time zone 0. | MAC |
| Time Settings | auto_dst.url = | None | URL within 511 characters | Moved to MAC.cfg | It configures the access URL of the DST file (AutoDST.xml). | MAC |



Yealink W52P Release Notes of Version 50

1. Introduction

Product Name: DECT-W52P

Firmware Version:

Base Version: 25.40.193.2 upgrade to 25.50.193.1

Handset Version: 26.40.193.2 upgrade to 26.50.193.1

■ Release Date: March 24, 2014

2. New Features

- 1. Added the feature that the W52P can now support TLS and SSL certificates.
- 2. Added the feature of Broadsoft Shared Call Appearance (SCA).
- 3. Added the feature that the time and date format of the W52P can now be configured by auto provisioning.
- 4. Added the feature of Suppress DTMF Display.

3. Optimization

- 1. <u>Fixed the issue that the phone triggers local ringback and ringtones at the same time.</u>
- 2. <u>Increased a select menu when the user transfer the second call to the first conversation.</u>
- 3. Optimized the version information user experience.
- 4. Removed the Limit of number of lines on Auto Answer feature.
- 5. Optimized the way of configuring the value of DNS.The value can now be configured through the web interface and the handset.
- 6. Removed the feature of SNMP.

4. Bug Fixes

1. Fixed the issue that the phone would not be registered successfully if the network environment ontains two gateway addresses.



- 2. Fixed the issue that the phone will cause abnormal display if the contact number contains letters.
- 3. Fixed the issue that only part of the handset rang when the call was forwarded from one Hunt Group to another Hunt Group.
- 4. Fixed the issue that the broadsoft calllog and directory did not take effect.
- 5. Fixed the issue that the telephone ring would be interrupted for 5 seconds when the call transferred to third party.
- 6. Fixed the issue that the phone failed to be deployed with the CFG files using the RPS feature.
- 7. Fixed the issue that the handset needs to reboot with the '1' digit key long pressed when a user wants to access the voice mail.
- 8. Fixed the issue that the W52P couldn't make a new call when an existing call was placed on hold.
- 9. Fixed the issue that the W52P directly goes to talking interface before the called party answers the phone.
- Fixed the issue that the translation of Chinese traditional was missing in the handset.

5. New Features instructions

Added the feature that the W52P can now support TLS and SSL certificates.

Description: The Dect phones from the factory will be preset with an SSL Certificate based on MAC.

The corresponding auto configuration:

#!version:1.0.0.1
security.ca_cert = 0/1/2(the default value is 2-All Cert)
security.dev_cert = 0/1(the default value is 0-Disable)
security.cn_validation = 0/1 (the default value is 0-Diaable)

2. Added the feature of Broadsoft Shared Call Appearance (SCA).

Description:The users can share their office extension on their mobile phone or the desk phone in their home. User can achieve remote office through this feature.

3. Added the feature that the time and date format of the W52P can now be configured by auto provisioning.

The corresponding auto configuration:



#!version:1.0.0.1

custom.handset.time_format = 0/1 (by the default value is 1) custom.handset.date_format = 0^6 (by the default value is 0)

4. Added the feature of Suppress DTMF Display.

Description:It is supported to enable or disable the phone to suppress the display of DTMF digits. If it is set to enabled, the DTMF digits are displayed as asterisks.

6. Optimization description

1. Fixed the issue that the phone triggers local ringback and ringtones at the same time.

Description:The phone will close the local ringback when the handset received 183 message and 180 message at the same time.

2. Increased a select menu when the user transfer the second call to the first conversation.

Description:when the user transfers the second call to the first called party, the handset will bring up a prompt interface for the user to select the corresponding talk line. The Dect phone will transfer the activity party to the held party.

3. Optimized the version information user experience.

Description:The version information of the view now is designed to handset and base separately.

4. Removed the Limit of number of lines on Auto Answer feature.

Description:The Dect phone will removed the incoming call prompt interface, and pick up the call automatically when the phone receive the Auto Answer SIP message.

5. Optimized the way of configuring the value of DNS. The value can now be configured through the web interface and the handset.

Description:The user can also change the value through the web interface and handset after the Dect phone obtain DNS message automatically.



Yealink W52P Release Notes of Version 40

1. Introduction

Product Name: DECT-W52P

Firmware Version:

Base Version: 25.30.193.6 upgrade to 25.40.193.2

Handset Version: 26.30.193.6 upgrade to 26.40.193.2

■ Release Date: December 05,2013

2. New Features

- 1. Added more access URL for users to access the Remote Phone Book.
- 2. Added the feature that users could access the missed call list via the idle screen.
- 3. Added the feature of dynamic VLAN.
- 4. Added the feature of OTA (Over-The -Air).

3. Optimization

- 1. Optimized the user experience, when the accounts were registered full.
- 2. Optimized the note, when locking the Keypad.
- 3. Optimized the user experience of the Battery electric power display.
- 4. Optimized the feature that the "R" Key could be used for Transferring.
- 5. Optimized the security of W52P.

- 1. Fixed the issue that the user could hear some noise when answering a call.
- 2. Fixed the issue that the second party didn't ring when the first party transferring a call to it.
- 3. Fixed the issue that both of the parties couldn't hear each-other during a call.
- 4. Fixed the issue that the W52P didn't support the Replace feature.



5. New Features instructions

1. Added 4 URLs for users to access the Remote Phone Book.

Description: The number of the access URL was added to five for the user to use the remote Phone Book.

The corresponding auto configuration:

```
#!version:1.0.0.1
remote_phonebook.data.X.url =
remote_phonebook.data.X.name =
```

2. Added the feature that users could access to the missed call list directly via the idle screen.

Description: Users could access to the missed call list via the idle screen.

Features: When the handset received a missed call, the users could press the "View" soft key to access the missed call list via the idle screen.

3. Added the feature of dynamic VLAN.

Description: The W52P could obtain the VALN information dynamically.

Features: The base can access the IP address normally, when the LLDP feature is enabled, despite it was turned to the network without VLAN.

The corresponding auto configuration:

```
#!version:1.0.0.1
network.vlan.dhcp_enable =
network.vlan.dhcp_option =
```

4. Added the feature of OTA (Over-The -Air).

Description: Both the base and handset support OTA feature, upgrade via web user interface or auto provisioning.

Features:

4.1 To trigger an OTA upgrade via web user interface:

- 1) Click on Phone->Upgrade.
- 2) In the Select and Upgrade Handset Firmware block, click Browse to locate the handset firmware file from local system.
- 3) Click Upgrade to trigger the OTA update.

4.2 Users can also trigger an OTA upgrade via auto provisioning.



The corresponding auto configuration:

```
#!version:1.0.0.1
over_the_air.url =
over_the_air.base_trigger =
over the air.handset tip =
```

6. Optimization description

1. Optimized the user experience, when the accounts were registered full.

Description: When the base was registered for five accounts, the web page of Status-Handset &VoIP would prompt "registered full".

2. Optimized the note, when locking the Keypad.

Description: If the keypad is locked, the LCD screen would prompt "Keypad Locked, Press and hold '#' to unlock!" when the user entered the On-hook key or others.

3. Optimized the user experience of the Battery electric power display.

Description: Optimized the user experience that the battery indicator was incorrect.

4. Added the feature that the "R" Key could be used for Transferring.

Description: Users could using the transfer (Blind Transfer/Semi-Attended Transfer/Attended Transfer) during a call by pressing the Redirect Key.

5. Optimized the security of W52P.

Description:

- 5.1. Any configuration items about password were cleared when the configuration file was exported.
- 5.2. The web will lock out prospective users for 3 minutes if they fail to enter the correct password three times in a row. And the lock time would redoubles up to one hour if continuous disoperation.



Yealink W52P Release Notes of Version 25.30.193.6

1. Introduction

Product Name: DECT-W52P

■ Firmware Version:

Base Version: 25.30.193.5 upgrade to 25.30.193.6

Handset Version: 26.30.193.5 upgrade to 26.30.193.6

■ Release Date: September 30,2013

2. Optimization

- 1. Optimized the battery charging strategy.
- 2. Optimized the logic of matching contacts.
- 3. Optimized the multipath delay issue.
- 4. Optimized the delay issue of Hunt Group.
- 5. Optimized the issue of battery gets hot when charging.

- 1. Fixed the issue that the W52P didn't send the DTMF-RFC 2833 when it received the 183 message.
- 2. Fixed the issue that the W52P didn't ring warning tone with "beep" when dialing the Voice mail.
- 3. Fixed the issue that the statement of dial plan in M7 was inoperative.
- 4. Fixed the issue that the W52P would request to the DNS server when the TTL was overtime.
- 5. Fixed the issue that if four handsets were registered to a same Base, and the user answering the second incoming call, there would be 2-4 seconds latency.
- 6. Fixed the issue that there were two seconds latency occurred when the users answering the incoming calls in the hunt group.
- 7. Fixed the issue that W52P didn't match the local contact.



- 8. Fixed the issue that the handset didn't support Chinese traditional characters.
- 9. Fixed the issue that the voice mail was played through the speaker phone mode rather than the handset mode.
- 10. Support auto answer feature when W52P receives the NOTIFY.

4. Optimization instructions

1. Optimized the battery charging strategy.

Description: Battery won't get too hot again when continues charging, and the total charging time doubled.

2. Optimized the logic of matching contacts.

Description: The contacts could be matched automatically when coming calls, regardless of which handset the phonebook is imported to.

3. Optimized the multipath delay issue.

Description: Optimized the issue that Latency occurred when any handset (H2-H5) answered a new call during the call of handset of H1.

4. Optimized the delay issue of Hunt Group.

Description: The handset received Hunt Group calls, and picked up, the call will not delay.



Yealink W52P Release Notes Of Version 25.30.0.40

1. Introduction

Product Name: DECT-W52P

■ Firmware Version:

Base Version: 25.30.0.20 upgrade to 25.30.0.40

Handset Version: 26.30.0.10 upgrade to 26.30.0.40

■ Release Date: August 17th, 2013

2. New Features

- 1. Added TR069 feature.
- 2. Added AutoP Code feature.
- 3. Added SNMP feature.
- 4. Added Auto Intercom feature.
- 5. Added Repeater feature.(Note: Only if the Base and Handset have upgraded to the latest software 25.30.0.40 and 26.30.0.40 can use the Repeater feature.)

3. Optimization

- 1. Optimized code of software.
- 2. Optimized the logic when performing a factory reset.
- 3. Optimized the logic of Anonymous Call feature.
- 4. Optimized warning tone.
- 5. Optimized the base's flash in the hardware, modified the software to be compatible with.

- 1. Fixed the issue that W52P Caller ID display incompletely.
- 2. Fixed the issue that base.pin_code=1234 can't be applied via AutoP.



- 3. Fixed the no voice issue when using W52P as calling party.
- 4. Fixed the no voice issue when W52P is placed behind SonicWall firewall.
- 5. Fixed the issue that accounts and password are lost when AutoP.
- 6. Fixed the issue that pin code can't be modified via AutoP.
- 7. Fixed the issue that the voice of W52P is badly damaged.
- 8. Fixed the issue that W52P displays 'Line Busy' during a call.
- Fixed the issue that the handset can't answer or make a call, but automatically recovers after some time.
- 10. Fixed the issue that 'Daylight Saving Time' doesn't work when set to 'Automatic', and the handset's time is an hour earlier.
- 11. Fixed the issue that 'Account Label' is 11 characters, but it displays incompletely in WEB.
- 12. Fixed the issue that 'Start Date' is greater than 'End Date' when using 'Daylight Saving Time'.
- 13. Fixed the failure issue of 'Anonymous Call' and 'Anonymous Call Rejection' features.
- 14. Fixed the issue that 'Blind Transfer' fails sometimes.
- 15. Fixed the issue that after registering 5 handsets, registration will fail when handset is making a call.
- 16. Fixed the issue that 'Remote Phonebook' can't work well.
- 17. Fixed the issue that when 4 handsets have been registered to a base, the handsets always drop.
- 18. Fixed the issue that when 'Refer' message needs authentication, but the phone will not send out the 'Refer' message with authentication.
- 19. Fixed the issue that the handset will be in 'Busy' state after several conferences.
- 20. Fixed the issue that Blind Transfer fails sometimes when using PortaWwitch.
- 21. Fixed the issue that 'offset' of Daylight Saving Time can't be set to 60.
- 22. Fixed the issue that the Date will go wrong when network disconnects, when reconnected to network it can't connect to VPN.
- 23. Fixed the issue that when phone sends AutoP request, User-agent needs to be with MAC address.
- 24. Fixed the issue that Intercom Warning Tone can't be played from speaker.
- 25. Fixed the issue that when Blind Transfer with the contacts in Directory, the handset will crash.
- 26. Fixed the issue that when Transfer with contacts in XML Phonebook, the handset



will crash.

- 27. Fixed the issue that the Password of Account can't support the '!' character.
- 28. Fixed the issue that the state of Hold is wrong.
- 29. Fixed the issue about the Offset of Daylight Saving Time.
- 30. Fixed the Attended Transfer issue.
- 31. Fixed the issue that when searching in Directory, GUI will crash.
- 32. Fixed the issue that DUT stays in 'Held' even call is disconnected.

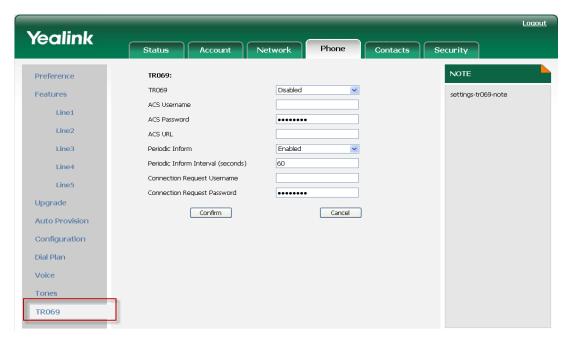
5. New Features instructions:

1. Added TR069 feature.

Description: The usage is consistent with V71. The server manages and configures the Phone, then the Phone downloads the configuration.

Detail: For more details, please refer to Yealink TR-069 Technote.pdf.

Path: Phone->TR069



Here is the configuration of Auto Provisioning(M7):

#!version:1.0.0.1

managementserver.enable =

managementserver.username =

managementserver.password =

managementserver.url =

managementserver.periodic inform enable =



managementserver.periodic_inform_interval =
managementserver.connection_request_username =
managementserver.connection_request_password =

2. Added AutoP Code feature.

Description: The usage is consistent with V71, the handset dials out AutoP Code to trigger AutoP. The handset dials out the AutoP Code without registered account, and then the phone will download the AutoP configuration files.

Note: Please do not perform any action in handset when Auto Provisioning.

Detail: For more details, please refer to

Yealink_Auto_Provisioning_User_Guide_V71_110.pdf

Here is the configuration of Auto Provisioning(M7):

#!version:1.0.0.1

autoprovision.x.name =

autoprovision.x.code =

autoprovision.x.url =

autoprovision.x.user =

autoprovision.x.password =

autoprovision.x.com aes =

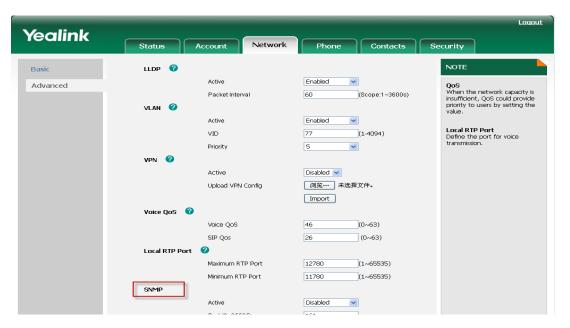
autoprovision.x.mac_aes =

3. Added SNMP feature.

Description: The usage is consistent with V71, the manager gets phone's information through SNMP for operating the phone.

Path: Network->Advanced->SNMP.





4. Added Auto Intercom feature.

Description:

Before you use this feature, please ensure that only one account and only one handset are registered with one Base.

When you enable this feature in handset (Settings->Telephony->Auto Intercom), if other phones call to this account using format of "Intercom Code + desired Extension", the handset will automatically answer with a beep and this Intercom call will be in hands-free status.

5. Added Repeater feature. (Note: Only if the Base and Handset have upgraded to the latest software 25.30.0.40 and 26.30.0.40 can use the Repeater feature.)

Description: Yealink DECT repeater can be deployed to extend the range of the W52P IP DECT phone to cover areas where reception was previously not available. This feature is especially useful to users who require mobility in large dwellings.

Detail: Enable Repeater mode with the handset (Settings->System Settings->Repeater Mode), and register the Repeater into the Base. Then find a good location to achieve optimum performance for the repeater.

For more details, please refer to 'Yealink DECT Repeater RT10 User Guide.pdf'.

6. Optimization instructions:

1. Optimized code of software.



Description: Optimized code of software.

2. Optimized the logic when performing a factory reset.

Description: Handset path: Menu->Setting->System Settings->Base Reset, WEB

Path: Phone->Upgrade->Reset to Factory.

When resetting to factory through above methods, only PIN Code and all the Handset's registration information will be kept; other configuration will be reset to factory.

3. Modified the logic of Anonymous Call feature.

Description: WEB path: Account->Basic, options 'Anonymous Call' and 'Anonymous Call Rejection' are used to enable/disable Anonymous Call and Anonymous Call Rejection features in the phone side.

WEB path: Phone->Features, option 'Anonymous Call code' and 'Anonymous Call Rejection Code' are used to enable/disable Anonymous Call and Anonymous Call Rejection features in the server side.

4. Optimized Warning tone.

Description:

- 1. There is no warning tone when the handset is out of the range of Base.
- 2. When Low Battery Alert Tone is on, if the battery is below 10% or 5%, comfortable voice will be sent out as well as prompt 'Battery Low' and note that press any key can stop the warning tone. Without being charged, warning tone will only be displayed two times (10% and 5%) from 100% to 0%, the interval of the two warning tones should be not less than 5 minutes.

5. Optimized the base's flash in the hardware, modified the software to be compatible with.

Description: Optimized the base's flash in the hardware, modified the software to be compatible with.



Yealink W52P Release Notes Of Version 25.30.0.20

1. Introduction

■ Product Name: DECT-W52P

■ Firmware Version: Base version: 25.20.0.10 upgrade to 25.30.0.20

■ Release Date: April 25th, 2013

2. New Features

 Added Broadworks Personal Network Directory, this feature is supported by Broadworks Xtended Services Interface.

3. Optimization

1. Optimized the speed of loading Contacts and Call History.

- 1. Fixed DTMF function doesn't work in some special situations.
- Fixed the issue that the W52P only tries to download the first batch of contacts even if the server inform the W52P the whole contact list is not fully downloaded.
- 3. Fixed the issue that in DHCP VLAN environment, W52P fails booting up because of the LLDP message sent by the phone is not correct.
- 4. Fixed the issue that in some cases, when the call ended, on the screen, the second section of call time is displayed as 7-digit number.
- 5. Fixed the issue that when W52P dials some numbers, there is no ring back tone.
- 6. Fixed the issue when the Caller ID is too long, it is not fully displayed; the fix is display the Caller ID in two lines.
- 7. Fixed the issue when the Auto Provisioning URL is from the RPS server, W52P doesn't try to get the configuration files although the Handset LCD displays



'updating....'

- 8. Fixed the issue that in some cases the configuration changes and requires reboot, the phone doesn't display a prompt as 'The setting needs a reboot, Reboot now or not'.
- 9. Fixed handset does not work well when the handset sets After Ring Time of Broadsoft no answer forward to some certain value no answer forward.
- 10. Fixed W52P can't accept the PBX Auto Intercom or Paging issue.
- 11. Fixed the issue that when making the screensaver and backlight settings as enabled, no matter which key you press, the first action is only activating the screen while not affect directly.
- 12. Fixed the language issue that after importing CSV phonebook, the screen display 'The operator is done' change to display 'The operation is complete'.
- 13. Fixed the issue that when long press '*', it will not disable all the ring tones.
- 14. Fixed the issue that when Screen Saver and the Keyboard Lock are enabled, long press hook off, screen will display that the clock interface and the idle interface are overlapping.
- 15. Fixed the issue that sometimes when you are making a transfer of a call from handset1 to handset2 (same base), you can hear both the MOH and the ring-back tone.
- 16. Fixed the issue that when the codec is G722 or G711A while doing the call transfer, the MOH (music on hold) will be distorted.
- 17. Fixed the issue that when handset is busy and reject a call, the second caller doesn't display Busy Here (486).
- 18. Fixed the issue that after a reboot of the base or the handset, sometimes the system information like 'Initializing data' remains in the display of the handset.

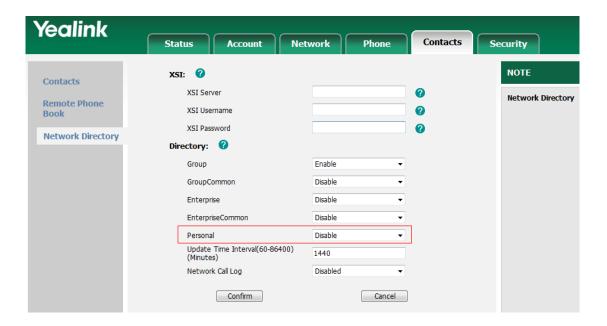
5. New Features instructions:

1. Added Broadworks Personal Network Directory, this feature is supported by Broadworks Xtended Services Interface.

Description: In Network Directory, add personal specific group, could Add, Adit, Delete.

Path: Contracts Network Directory





6. Optimization instructions:

1. Optimized the speed of loading contacts and call history.