

BERKSHIRE IP5000



USER MANUAL

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1. PRODUCT OVERVIEW

The ATL Berkshire 5000 is a SIP-based IP telephone developed by ATL Telecom Limited. It supports a number of advanced call features and can be easily customized for enterprise or consumer use. New features can also be added to meet different customer's needs.

1.1 Key Features

The IP Phone provides the following features:

- 20 x 2 LCD Display
- 2 Call Lines
- Handsfree
- Headset
- Redial
- Speed Dialing
- Caller ID and Name Display
- Call History
- Call Hold
- Call Waiting
- Call Transfer (Attended / Unattended)
- Call Blocking
- Call Forward (Busy / No Answer / Unconditional)
- 3-Way Conferencing
- Direct IP Call
- Phonebook
- Message Waiting Indication (MWI)
- Firmware Upgrade through TFTP
- Multiple User Settings
- NAT Traversal (STUN)
- Web Interface Configuration

1.2 User Interface

1.2.1 Top & Side Views



1.2.2 LCD Screen

The LCD screen can display 20 characters. The character spaces are arranged in 2 rows, with 10 characters in each row.



1.2.2.1 Scroll Up and Down for Display

When displaying information with length more than 2 rows, a down arrow ' \downarrow ' will appear automatically at the bottom-right corner indicating that you can scroll down. (The last character at each row is reserved for displaying ' \uparrow ' & ' \downarrow ' arrow.)

1	>	U	S	e	r	n	a	m	e					
2		Р	a	S	S	w	0	r	d					\downarrow

You can press the 'Down' button to scroll down, and an up arrow ' \uparrow ' will appear automatically at the top-right corner indicating that you can scroll up. You can press the 'Up' button to scroll up to the top.

2	>	Р	a	S	S	w	0	r	d							1
3		D	i	S	р	1	a	У		Ν	a	m	e			\downarrow

If you scroll down to the last row, only the ' \uparrow ' arrow will be displayed.

8		0	b		•		Р	r	0	X	у		Р	0	r	t		1
9	٨	R	e	g	i	S	t	r	a	t	i	0	n					\downarrow

1.2.2.2 Item Selection

When an item can be selected on the screen, a selection cursor '>' is displayed on the left of the item to be selected. Press the 'Up' and 'Down' button can move the selection cursor to the targeted item. To confirm the selection, press the "OK" key.

1	>	U	S	e	r	n	a	m	e					
2		Р	a	S	S	W	0	r	d					\downarrow

1.2.2.3 Character Input

When information is to be entered, an input cursor '_' is displayed to indicate the position of character to be inserted on the screen. The cursor shifts to the right as characters are entered. To delete the last inputted character, press "Menu / Delete". After all information is entered, press "OK" key to confirm the input.

D	i	S	p	l	a	у	N	a	m	e				A
_														

1.2.2.4 Input Modes

There are in total four input modes: *Alphabet*, *Number*, *Hostname* and *IP Address*. Depending on the input contents, one or more input modes will be provided. Basically, these input methods are similar to those adopted in mobile phones. If the input mode can be changed by the user during information entry, a character, which indicates the current input mode, will be displayed at the top-right corner of the screen. The current input mode can be toggled by pressing the '#' key. The input modes represented by each character are depicted as follow:

- 1. A: Alphabet
- 2. N: Number
- 3. H: Hostname; similar to Alphabet except special characters include only '.' and '-'.
- 4. I: IP Address; similar to Number except the '*' key is mapped to '.'.

Some examples are provided here to illustrate the usage of input modes. E.g. the default input mode of Display Name is Alphabet.

D	i	S	р	1	a	у	N	a	m	e				A
R	F	С	Ι											

After pressing the "#" key, the mode is changed to Number.

D	i	S	р	1	a	у		N	a	m	e				N
R	F	С	3	2	6	1	Ι								

E.g. the default input mode of Outbound Proxy is Hostname.

0	u	t	b	0	u	n	d		P	r	0	X	у					Η
S	i	р	•	a	t	1	t	e	1	e	c	0	m	•	c	0	m	

After pressing the '#' key, the mode is changed to IP Address.

0	u	t	b	0	u	n	d		Р	r	0	X	у			Ι
1	0	•	5	•	1	•	1	١								

1.2.2.5 Scroll Right for Long Input

During information entry, if the cursor approaches the end of row, the display of inputted information will automatically scroll to the right side.

D	i	S	p	l	a	у		N	a	m	e						A
С	h	a	n		S	i	u		Μ	i	n	50	Р	a	t	-	

After the character 'r' is entered, a left arrow ' \leftarrow ' is displayed on the first space of the row to indicate the display has scrolled to the right side.

D	i	S	p	1	a	у	N	a	m	e						A
÷	a	n		S	i	u	M	Ι	n	g	Р	a	t	r	_	

1.2.2.6 Current Setting

When multiple options are displayed on the screen, the current setting is indicated by a dash ('-') on the left. E.g.

1	>	E	n	a	b	1	e							
2	I	D	i	S	a	b	1	e						

1.2.3 Keypad

The Keypad contains:

- 1) Regular Numeric Pad 12 keys
 - 1. Key '2'-'9' are also mapped to alphabetic characters 'a' to 'z'.
 - Key '*' is also mapped to special characters '*', '.', '@', '-', '^', '!', '&' and '→'.
 - 3. Key '#' is also mapped to input mode toggle.
 - 4. Key '0' is also mapped to *<space>*.
- 2) Simple Buttons 11 keys
 - 1. Message (Voice Mail)

- 2. Line1/Hold (select line 1 for held call or call waiting, or hold current call)
- 3. Line2/Hold (select line 2 for held call or call waiting, or hold current call)
- 4. Menu / Delete (Menu or backspace)
- 5. OK (Make call or Yes button)
- 6. Cancel (Hang up or No button)
- 7. "Handsfree" (Speaker Phone)
- 8. Up (for both volume increase and menu scroll up)
- 9. Down (for both volume decrease and menu scroll down)
- 10. Mute
- 11. Function (Function key selection)
- 3) Hook 1 key
- 4) Function Keys 4 keys
 - 1. 3-way Conference (Function, 1)
 - 2. Call Transfer (Function, 2)
 - 3. Redial (Function, 9)
 - 4. DND (Function, 7)

1.2.4 LEDs

The phone contains 5 LEDs

- Message (Message Waiting Indication)
- Line1 ("flashing" when ringing or call-waiting, "on" when line 1 in use)
- Line2 ("flashing" when ringing or call-waiting, "on" when line 2 in use)
- Handsfree
- Headset

1.2.5 Web Interface

The phone supports remote configuration from a web browser. The settings of the phone can be viewed and changed through a web-based and user-friendly graphical user interface. For details of web configuration interface, please refer to Section 7.

2. INSTALLATION & EASY SETUP

2.1 Phone Installation

Follow the steps below to install the phone.

Step 1: Unpack the ATL Berkshire 5000 telephone carton.

The phone package should contain the following components:

- 1. IP Telephone
- 2. Handset
- 3. Power Adapter
- 4. Ethernet Cable
- 5. Handset Cord

Step 2: Handset Connection

Connect the handset to the phone body with the handset cord.

Step 3: Network Connection

Connect the phone to a hub or switch with the Ethernet cable through the uplink port "LAN". A desktop or notebook, if any, can be connected to the network through the "PC" port in the phone.

Step 4: Power Connection

Connect one end of the AC to DC adapter to the power jack of the SIP phone. Plug the power socket converter into the power outlet if necessary. Then connect the other end of adapter to the converter.

Step 5: Booting Up

After the power is connected, the phone will boot up automatically. The following screen will be shown on the LCD:

P	h	0	n	e	S	t	a	r	t	i	n	g	•	•		

After booting up, the phone will enter the *Easy Setup* wizard and the following screen will be displayed on the phone. (If *Easy Setup* wizard does not start, it can be triggered by pressing the 'Menu / Delete' button and selecting Settings > Easy Setup.)

-	-	-		E	a	S	у		S	e	t	u	р		-	-	-	
		Р	r	e	S	S		a	n	у		k	e	У	•			

2.2 Easy Setup

Before you can use the phone to make calls, you should first configure the phone settings. The *Easy Setup* wizard will guide you to set up the phone easily.

0. First of all, the start screen of Easy Setup will be shown as follows:

-	-	-	E	a	S	у	S	e	t	u	р	-	-	-		
		Р	r	e	S	S	a	n	у		k	e	у	•		

1. Press any button to continue. The following screen will then be shown:

Ι	S		D	Η	С	Р		a	v	a	i	l	a	b	1	e	?	
1		=		Y	e	S	,		2		=		N	0				

2. If DHCP service is available on your network (i.e. you get your IP address dynamically), press 1 and jump to Step 7. If not, press 2 and the following screen will then be shown:

Ι	Р	A	d	d	r	e	S	S					
_													

3. Press the '**Menu** / **Delete**' button to delete the current setting if necessary. Input the IP address of the phone with the keypad. The "." character can be inputted by pressing the "*" button. Press the 'OK' button to confirm and save the setting. If the setting is saved, the following screen will then be shown:

S	u	b	n	e	t	M	a	S	k					
-														

4. Change the subnet mask if necessary. If the setting is confirmed and saved, the following screen will then be shown:

D	e	f	a	u	1	t	G	a	t	e	w	a	у			
_																

5. Change the IP address of default gateway if necessary. If the setting is confirmed and saved, the following screen will then be shown:

D	N	S	S	e	r	V	e	r					
_													

6. Change the IP address of DNS Server if necessary. If the setting is confirmed and saved, the following screen will then be shown:

D	e	f	i	n	e	Т	i	m	e	Z	0	n	e	•		

7. You will need to set your timezone. Select your timezone from the list:

2	7	>	G	M	Т	+	8	:	0	0				1
2	8		G	M	Т	+	9	:	0	0				\downarrow

8. Press the 'OK' button to confirm and save the setting. The following screen will then be shown:

U	S	e	Ν	e	t	w	0	r	k		Т	i	m	e	?		
1		=	Y	e	S	,		2		Ш		N	0				

9. Press 1 to enable the use of network time or 2 to disable it. If "No" is selected, jump to Step 10. Otherwise, if "Yes" is selected, the following screen will then be shown:

N	Т	Р		S	v	r		A	d	d	r	e	S	S			Η
р	0	0	1	•	n	t	р	•	0	r	g						

10. Change the NTP server address if necessary. Press the 'OK' button to confirm and save the setting. Jump to Step 12.

D	a	t	e	:		(Y	Y	Y	Y	M	M	D	D)		
2	0	0	5	0	6	1	0	I									

11. Input the date (E.g. 20050610). Press the 'OK' button to confirm and save the setting. The following screen will then be shown:

Т	i	m	e	:	(H	Η	M	M	S	S	,	2	4	Н	r)	
1	4	1	0	2	0	-												

12. Input the time in 24-hr format (E.g. 141020). Press the 'OK' button to confirm and save the setting. The STUN setup screen will then be shown:

S	Т	U	N		a	V	a	i	1	a	b	l	e	?			
1		=		Y	e	S	,		2		=		N	0			

13. If STUN service is available on your network, press 1 and the following screen will then be shown. If not, press 2 and jump to Step 16:

S	T	U	N		S	V	r	A	d	d	r	e	S	S		Η
0	•	0	•	0	•	0	Ι									

14. Delete the current setting and input the hostname or IP address of your STUN server. Press the 'OK' button to confirm and save the setting. The user setup screen will then be shown:

-	-	U	S	e	r	1	S	e	t	u	р	-	-	

15. After a moment, you will be asked to input your username.

Y	0	u	r	U	S	e	r	n	a	m	e				A
_															

16. Change your username if necessary. If the setting is confirmed and saved, you will be asked to input your password:

Y	0	u	r	Р	a	S	S	w	0	r	d				A
_															

17. Input your password. For privacy reason, the inputted character will be changed to '*'. Press the 'OK' button to confirm and save the setting. The following screen will then be shown:

Y	0	u	r	D	i	S	р	1	a	у	N	a	m	e		A
-																

18. Change your display name if necessary. If the setting is confirmed and saved, you will be asked to input the SIP domain:

Y	0	u	r	D	0	m	a	i	n					Η
_														

19. Change your domain if necessary. If the setting is confirmed and saved, you will be asked to set the SIP registrar:

S	e	t	R	e	g	i	S	t	r	a	r	?				
1		I	Y	e	S	,		2		II		Ν	0			

20. Press 1 if you would like to set the registrar. The registrar host screen will then be shown. Otherwise, press 2 to skip this configuration and jump to Step 20. (If skipped, the Domain setting will be used to resolve the registrar.)

R	e	g	i	S	t	r	a	r	H	0	S	t			Η

21. Change the registrar host if necessary. If the setting is confirmed and saved, you will be asked to input the registrar port:

R	e	g	i	S	t	r	a	r	P	0	r	t			
5	0	6	0	-											

22. The default registrar port number is 5060. Change it if necessary. If the setting is confirmed and saved, you will be asked to set the outbound SIP proxy:

S	e	t	0	u	t	b	0	u	n	d	Р	r	0	X	у	?	
1		=	Y	e	S	,		2		=		Ν	0				

23. Press 1 if you would like to set the outbound proxy. The proxy host screen will then be shown. Otherwise, press 2 to skip this configuration and jump to Step 23. (If skipped, the Domain setting will be used to resolve the outbound proxy.)

Р	r	0	X	у	H	0	S	t					Η
-													

24. Change the proxy host if necessary. If the setting is confirmed and saved, you will be asked to input the proxy port:

Р	r	0	X	у	Р	0	r	t					
5	0	6	0	١									

25. The default proxy port number is 5060. Change it if necessary. If the setting is confirmed and saved, the ending screen will then be shown:

E	a	S	у	S	e	t	u	р	e	n	d	e	d	!		

26. You have finished the phone setup. After a moment, the standby screen will be shown. Check the phone icon on the top-left corner of the screen. If the screen shows a normal phone icon a s below, it indicates that the phone has registered to the SIP server successfully. You can now start to make or receive calls. Refer to Section 3 for the details of basic operations.

		1	0	0	0		A	1	i	c	e		A	u						
--	--	---	---	---	---	--	---	---	---	---	---	--	---	---	--	--	--	--	--	--

_	_	_							_					_
Δ	7		1	0		1	Δ	D	ъл					
U	1	:	1	9	:		U	r	IVI					

27. Otherwise, if the screen shows the phone's IP address as below, it indicates that the phone has failed to register to the SIP server and may not be able to make or receive calls.

1	•	2	•	3	•	4								
0	7	••	1	9	••	1	0	Р	Μ					

28. Make sure the inputted settings are correct. You can run the *Easy Setup* wizard again to review and change the settings. To do so, press "Menu / Delete" to enter the phone menu.

1	>	P	h	0	n	e	b	0	0	k						
2		С	a	1	1		Н	i	S	t	0	r	У			\downarrow

29. Scroll down to "Settings".

4	>	S	e	t	t	i	n	g	S							\uparrow
5		S	W	i	t	c	h		U	S	e	r				\downarrow

30. Press the 'OK' button to enter "Settings". Various settings will be shown:

1	>	P	h	0	n	e								
2		Ν	e	t	w	0	r	k						\downarrow

31. Scroll down to "Easy Setup".

3		U	S	e	r									1
4	٨	E	a	S	У	S	e	t	u	р				

32. Press the 'OK' button to enter "Easy Setup". The setup wizard will start again.

3. BASIC FUNCTIONS

3.1 Dialing

3.1.1 Dialing a Phone Number, SIP URI or IP address

To make a call by dialing the number, SIP URI or IP address (The phone can make SIP calls directly to another SIP phone without passing through SIP Proxy via IP address). You could dial a number during off-hook or on-hook.

3.1.1.1 Off-hook dialing screen

1. Pick up the handset, press the **"Handsfree"** button or press **"Line1/Hold**" to use the speakerphone. A dial tone will be heard and the **"Line1/Hold**" LED will be turned on. The dialing screen will be shown:

E	n	t	e	r	n	u	m	b	e	r				Ν
-														

2. Enter the phone number (e.g. 2000), SIP URI (e.g. 2000@1.2.3.4) or IP address (e.g. 123.123.123.123.123). The maximum number of characters allowed is 60. You could toggle the input method to "A" using the "#" key.

Ε	n	t	e	r	n	u	m	b	e	r				Ν
2	0	0	0	_										

3. Press the 'OK' button. The phone will start calling the number. The calling screen will be shown as below:

С	a	l	1	i	n	g	2	0	0	0	•	•	•		

4. If the callee receives the call, a ring-back tone will be heard. After the callee answers the call, the following screen will be shown:

С	0	n	n	e	c	t	e	d	t	0				
2	0	0	0											

3.1.1.2 On-hook predialing screen

1. When the phone is on hook, press the number "1" to "9", "*" or "#" to enter the predialing screen like this:

E	n	t	e	r	n	u	m	b	e	r				N
_														

2. Enter the phone number (e.g. 2000), SIP URI (e.g. 2000@1.2.3.4) or IP address (e.g. 123.123.123.123.123). The maximum number of characters allowed is 60. You could toggle the input method to "A" using the "#" key.

Е	n	t	e	r	n	u	m	b	e	r				N
2	0	0	0	-										

3. Pick up the handset, press the **"Handsfree"** button, press "Line1/Hold" or press the 'OK' button to use the speakerphone. The "Line1/Hold" LED will be turned on. The phone will start calling the number. The calling screen will be shown as below:

С	a	l	l	i	n	g	2	0	0	0	•	•	•		

4. If the callee receives the call, a ring-back tone will be heard. After the callee answers the call, the following screen will be shown:

С	0	n	n	e	c	t	e	d	t	0				
2	0	0	0											

3.1.2 Dialing a Call from Call History

The phone numbers of the last 30 missed calls, dialed calls and received calls are saved in the call history. You can dial a call from the call history.

To dial a call from the call history:

- 1. Press "Menu / Delete" button to enter the phone menu.
- 2. Scroll up/down to 'Call History' as shown below:

1		Р	h	0	n	e	b	0	0	k						
2	٨	С	a	1	1		Η	i	S	t	0	r	У			\downarrow

3. Press the 'OK' button to enter "**Call History**". The call history types will be shown as below:

1	>	M	i	S	S	e	d		С	a	1	l	S				
2		R	e	c	e	i	V	e	d		С	a	l	1	S		\downarrow

4. Select 'Missed Calls', 'Dialed Calls' or 'Received Calls' and press the 'OK' button. A list of phone numbers will be shown as below:

1	>	2	0	0	0							
2		3	0	0	0							

- 5. Use the 'Up' or 'Down' buttons to navigate the list of numbers.
- 6. Press the 'OK' button to select the number, the following screen will be shown:

1	>	С	a	1	1							
2		S	a	v	e							

7. Press the 'OK' button to call the number. The phone will start calling the selected phone number and the speakerphone will be automatically turned on. Pick up the handset to switch the call to the handset if necessary.

С	a	1	1	i	n	g	2	0	0	0	•	•	•		

8. If the call is connected, the called number is shown on the screen as follows:

С	0	n	n	e	c	t	e	d	t	0				
2	0	0	0											

3.1.3 Redialing

To dial the phone number which has been most recently dialed:

- 1. Pick up the handset.
- 2. Press key '**Function**' > '**Redial**' (keypad '9'). The calling screen will be shown as below:

С	a	l	l	i	n	g	2	0	0	0	•	•	•		

If you are going to use speakerphone for the redial call, you can press the "**Redial**" button directly without pressing the "Handsfree" button.

3.2 Terminating a call

To end a call, do one of the followings:

- 1. Go on-hook.
- 2. Press the hook switch.
- 3. Press the 'Cancel' button.
- 4. Press the "Handsfree" button if the speakerphone is in use.

Otherwise, if the call is terminated by the other party, a busy tone will be heard. The following screen will also be shown:

С	a	l	l	t	e	r	m	i	n	a	t	e	d	b	у	
2	0	0	0													

Follow the methods of terminating a call to hang up the call completely.

3.3 Answering a call

(In order to receive calls properly, the screen should show a normal phone icon \square in standby mode. If the screen shows a phone icon with a cross \square , the phone may not be able to receive calls. Refer to Section 2.2 to fix the problem before continuing.)

When there is an incoming call, the phone will ring and the 'Line1/Hold' (or 'Line2/Hold') LED will flash. The caller number and name, if available, will be shown on the screen as below:

Ι	n	c	0	m	i	n	g		С	a	1	1	f	r	0	m	
2	0	0	0		J	0	h	n									

To answer the call, pick up the handset or press the "**Handsfree**" button if speakerphone is to be used.

3.4 Denying a call

If you do not want to answer an incoming call, you can deny the call.

To deny the call, press the 'Cancel' button. The phone will then stop alerting.

3.5 Adjusting the Volume

3.5.1 Changing the Ringer Volume

When the phone is in standby mode or is ringing, you can use the "Up" or "Down" keys to increase or decrease the volume of the ringer. The screen will be shown as below:

R	i	n	g	e	r	V	0	l	u	m	e				

3.5.2 Changing the Handset Volume

When the handset is off-hook, you can use the "Up" or "Down" keys to increase or decrease the volume of the handset. The screen will be shown as below:

Η	a	n	d	S	e	t	V	0	1	u	m	e			

3.5.3 Changing the Speaker Volume

When the speaker is turned on, you can use the "Up" or "Down" keys to increase or decrease the volume of the speaker. The screen will be shown as below:

S	р	e	a	k	e	r	V	0	1	u	m	e			

3.6 Muting a Call

The mute function prevents the party you are talking to from hearing your voice.

To mute a call, press button '**Mute**' during a call. A character '**M**' will then be shown on the bottom-right corner of the screen.

С	0	n	n	e	c	t	e	d	t	0				
2	0	0	0		J	0	h	n						Μ

To resume the 2-way conversation, press button 'Mute' again. The character 'M' will then disappear.

3.7 Handset / Speaker

The phone supports talking in a call by handset or speakerphone. During a call, it is also possible to easily switch between the two devices.

When the speakerphone is in use, pick up the handset will switch the call to the handset.

When the handset is off-hook, press the key "Handsfree" will toggle the connection between the two devices.

4. ADVANCED FUNCTIONS

4.1 Call Hold

To hold an active call, follow these steps:

1. Set up a call first and a screen will be shown as below:

С	0	n	n	e	c	t	e	d	t	0				
2	0	0	0											

2. Press the 'Line1/Hold' (or 'Line2/Hold') key to hold the call. The 'Line1/Hold' (or Line2/Hold) LED will then flash. The following screen will also be shown:

С	a	l	l	0	n	h	0	l	d	w	i	t	h		
2	0	0	0												

3. To unhold the call, press the Line1/Hold (or Line2/Hold) key again. The original screen will be shown again:

С	0	n	n	e	c	t	e	d	t	0				
2	0	0	0											

4.2 Call Waiting

4.2.1 To enable Call Waiting

Before the call waiting feature can be used, it must be enabled first. To enable call waiting, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Call Features'.

3	>	С	a	l	l		F	e	a	t	u	r	e	S			
4		A	b	0	u	t											

3. Scroll down and select "Call Waiting".

4	>	С	a	1	l		W	a	i	t	i	n	g				
5		D	0		Ν	0	t		D	i	S	t	u	r	b		

4. Enter "Use Call Waiting".

1	>	U	S	e		С	a	1	l		W	a	i	t	i	n	g	
2		С	a	l	1		W	a	i	t	i	n	g		D	u	r	

5. Select "Enable" to enable call waiting.

1	>	E	n	a	b	l	e							
2	I	D	i	S	a	b	1	e						

4.2.2 To answer Call Waiting

After call waiting is enabled, if you are engaged in an active call and a new call comes in, the call waiting feature will be triggered. You can follow these steps to answer the new incoming call:

1. If you are engaged in Line 1 (or Line 2) and a new call comes in (e.g. from 3000), you will hear continual tones and the LED of the line with new incoming call will flash. The caller ID of incoming call will also be shown on the screen periodically as below:

Ι	n	c	0	m	i	n	g	c	a	l	l	f	r	0	m	
L	2	:		3	0	0	0									

2. To answer the call, you can press the line button of the incoming call Line2/Hold (or Line1/Hold). The active call will then be held and the LED of the held line will flash. The LED of the active line will be also on. The screen will show the caller ID of the new incoming call:

С	0	n	n	e	c	t	e	d	t	0				
3	0	0	0											

3. You can now talk on the new line. To switch back to the original held line, press the button of the line. The screen will change back to show the original line:

С	0	n	n	e	c	t	e	d	t	0				
2	0	0	0											

If the incoming call from remote party is terminated during alerting or exceeds the no answer timeout duration, the missed call indicator will be displayed after the current call is terminated.

M	i	S	S	e	d	c	a	1	1	S	:	1			

1. Press "OK" to see the details of the missed call. The missed call history will then be shown.

1	>	2	0	0	0	J	0	h	n					
2		3	0	0	0	M	a	r	у					

4.2.3 To configure Call Waiting Duration

To configure the "Call Waiting Duration", follow these steps:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call Features".

2		С	a	1	1	Η	i	S	t	0	r	у				1
3	٨	С	a	l	1	F	e	a	t	u	r	e	S			\downarrow

3. Scroll down and enter "Call Waiting".

3		S	р	e	e	d		D	i	a	l	i	n	g			1
4	>	С	a	l	1		W	a	i	t	i	n	g				\downarrow

4. Scroll down and enter "Call Waiting Dur".

1		U	S	e		С	a	l	1		W	a	i	t	i	n	g	
2	٨	С	a	l	1		W	a	i	t	i	n	g		D	u	r	

5. The available durations are 5 seconds, 10 seconds and 15 seconds. Select the preferred setting and press "OK" key.

1	-	5		S	e	c	0	n	d	S					
2	٨	1	0		S	e	c	0	n	d	S				\downarrow

4.3 Call Transfer

The phone supports two call transfer modes: Unattended and Attended.

4.3.1 Unattended Call Transfer

If an active call is ongoing (e.g. with 2000) and the user wants to perform an unattended (blind) call transfer to 3000, follow these steps:

1. Press Function > Xsfer (keypad '2') to enter the call transfer screen as shown:

Т	r	a	n	S	f	e	r	t	0					
_														

2. Input the number to transfer to (E.g. 3000).

Т	r	a	n	S	f	e	r	t	0					
3	0	0	0	Ι										

3. Press 'OK' button to confirm. The call will be transferred to the inputted number.

С	a	l	1		t	r	a	n	S	f	e	r	r	e	d		
t	0		3	0	0	0											

4. After the call transfer operation is completed, the call termination screen will be shown:

С	a	1	1	t	e	r	m	i	n	a	t	e	d	b	у	
2	0	0	0													

5. Hang up the call. The phone will return to standby mode as shown:

K	1	0	0	0		A	1	i	c	e		A	u					
0	6	-	0	6	1	1	4		1	7	:	5	0	:	1	9		

4.3.2 Attended Call Transfer

If an active call is ongoing (e.g. with 2000) and the user wants to perform an attended call transfer to 3000, follow these steps:

1. Press the 'Line1/Hold' key to hold the call. The 'Line1/Hold' LED will then flash. The following screen will also be shown:

С	a	l	l	0	n	h	0	l	d	W	i	t	h		
2	0	0	0												

2. Press the 'Line2/Hold' key to place another call (e.g. 3000). The following screen will be shown:

С	0	n	n	e	c	t	e	d	t	0				
3	0	0	0											

3. Press **Function > Xsfer** (keypad 2) to transfer the call. The call transfer screen will be shown:

С	a	1	1	Τ	r	a	n	S	f	e	r	r	i	n	g		

4. If the call transfer is successful, the phone will return to standby mode.

Note: For attended call transfer to be successful, the transfer target must enable call waiting. Otherwise, the transfer target will not be able to receive the new call from the transferee (the party being transfer).

4.4 3-way Conferencing

The phone supports 3-way conferencing with two other parties. To start conferencing, follow these steps:

1. Set up a call first and a screen will be shown as below:

С	0	n	n	e	c	t	e	d	t	0				
2	0	0	0											

2. Press the 'Line1/Hold' key to hold the call. The 'Line1/Hold' LED will then flash. The following screen will also be shown:

С	a	l	l	0	n	h	0	l	d	W	i	t	h		
2	0	0	0												

3. Press the 'Line2/Hold' key to place another call. The following screen will be shown:

С	0	n	n	e	c	t	e	d	t	0				
3	0	0	0											

4. Press **Function > 3-way** (keypad '1') to bridge two calls. The conferencing call has now been set up. The phone number on each line will be shown as below ('*' indicates the line is active):

1	-	*	2	0	0	0						
2	-	*	3	0	0	0						

5. End the conferencing call by terminating any one of the lines.

If you would like to hold the conferencing call, follow these steps:

1. Press the 'Line1/Hold' or 'Line2/Hold' key. The following screen will be shown ('O' indicates the line is held):

1	-	0	2	0	0	0						
2	I	0	3	0	0	0						

2. Press the 'Line1/Hold' or 'Line2/Hold' key again to unhold the conferencing call.

If one of the parties holds your call (e.g. 2000), the following screen will be shown ('**H**' indicates the line is held by remote party):

1	-	Η	2	0	0	0						
2	I	*	3	0	0	0						

4.5 Call Forward (On Busy / On No-Answer / Unconditional)

4.5.1 Call Forward Activation

The phone supports three call-forward modes: **On-Busy**, **On No-Answer** and **Unconditional**. To enable the call-forward feature, follow these steps:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call Features".

3	>	С	a	1	1		F	e	a	t	u	r	e	S			1
4		S	e	t	t	i	n	g	S								\downarrow

3. Select "Call Forward".

1	>	С	a	1	1		F	0	r	W	a	r	d				
2		D	i	r	e	c	t		Ι	Р		С	a	1	1		\downarrow

4. Select the mode to be enabled.

1	>	U	n	c	0	n	d	i	t	i	0	n	a	1			
2		0	n		B	u	S	у									\downarrow

5. Select "Forward Address" to set the call-forward number.

1	>	F	0	r	W	a	r	d		A	d	d	r	e	S	S		
2		U	S	e		U	n	c	0	n	•							

6. Input the call-forward number (E.g. 3000) or SIP URI (e.g. 2000@1.2.3.4). The maximum number of characters allowed is 60. You could toggle the input method to "A" using the "#" key.

F	0	r	W	a	r	d	A	d	d	r	e	S	S			A
3	0	0	0	Ι												

7. Press 'OK' button to confirm. Then depending on the selected mode, scroll down to select "Use Uncon.", "Use Busy" or "Use No-Answer".

1		F	0	r	W	a	r	d		A	d	d	r	e	S	S		
2	>	U	S	e		U	n	c	0	n	•							

8. Select "**Yes**" to enable the call forward feature.

1	>	Y	e	S								
2	-	N	0									

9. If the call forward feature is enabled, a call forward icon will be shown on top-right corner of the screen.

1		F	0	r	w	a	r	d		A	d	d	r	e	S	S	Ę	
2	٨	U	S	e		U	n	c	0	n	•							

10. For **On No-Answer** mode, the timeout period of no answer can also be set. To change the timeout period, scroll up/down to select "**Timeout (secs)**".

2	>	Τ	i	m	e	0	u	t		(S	e	c	S)		1
3		U	S	e		Ν	0	I	A	n	S	W	e	r			

11. Select the no-answer timeout period. The available periods are 5, 10 and 15 seconds.

1	>	5		S	e	c	0	n	d	S					
2		1	0		S	e	c	0	n	d	S				\downarrow

Note: When multiple call forward modes are enabled concurrently, **Unconditional** mode will override both **On Busy** and **On No-Answer** modes. If **Unconditional** mode is disabled, the other enabled modes will become effective again.

4.5.2 Call Forward Status Query

The phone provides call forward status query function which allows user to check the status of each call forward mode in one shot.

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call Features".

2		С	a	1	1	H	i	S	t	0	r	у				1
3	٨	С	a	l	1	F	e	a	t	u	r	e	S			\downarrow

3. Enter "Call Forward".

1	>	С	a	1	1		F	0	r	W	a	r	d				
2		D	i	r	e	c	t		Ι	Р		С	a	1	1		\downarrow

4. Scroll down and enter "Status Query".

4	>	S	t	a	t	u	S	Q	u	e	r	у			1
5		С	a	n	c	e	1	A	l	1					

5. Below is an example of the result page, with CFU is on, target 2000, CFB is off, CFNA is off. "Down" button is pressed one at a time

С	a	l	l		F	0	r	w	a	r	d						
U	n	c	0	n	d	i	t	i	0	n	a	1	:	0	n		\downarrow

U	n	c	0	n	d	i	t	i	0	n	a	1	:	0	n		\uparrow
2	0	0	0														\downarrow

2	0	0	0											1
0	n		B	u	S	у	:	0	f	f				\downarrow

0	n	B	u	S	у	:		0	f	f						1
0	n	N	0	I	A	n	S	w	e	r	:	0	f	f		

4.5.3 Cancel All Call Forward

"Cancel All Call Forward" allows user to cancel the unconditional call forward, busy call forward and no answer call forward at one shot.

To cancel all the forward modes, follow these steps:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call Features".

2		С	a	1	1	H	i	S	t	0	r	у				1
3	>	С	a	1	1	F	e	a	t	u	r	e	S			\downarrow

3. Enter "Call Forward".

1	>	С	a	l	l		F	0	r	W	a	r	d				
2		D	i	r	e	c	t		Ι	Р		С	a	1	1		\downarrow

4. Scroll down and enter "Cancel All".

4		S	t	a	t	u	S	Q	u	e	r	у			1
5	>	С	a	n	c	e	1	A	l	1					

5. The following screen will be shown.

Р	1	e	a	S	e	W	a	i	t	•	•	•			

6. After completed, it will stay in the last page as below.

4		S	t	a	t	u	S	Q	u	e	r	у			1
5	٨	С	a	n	c	e	1	A	1	1					

4.6 Do Not Disturb

When **Do Not Disturb** mode is turned on, the phone does not receive calls. To enable **Do Not Disturb** function, follow these steps:

1. Press key '**Function**' > '**DND**' (keypad '7'). A character '**D**' will then be displayed on the bottom-right corner of the screen to indicate **Do Not Disturb** mode is on.

	1	0	0	0		A	l	i	c	e	A	u				
0	2	:	5	1	:	3	0		Р	Μ					D	

2. If **Do Not Disturb** mode is to be turned off, press key '**Function**' > '**DND**' again. The character '**D**' will then disappear.

Alternatively, the **Do Not Disturb** function can also be enabled / disabled from the phone menu. Follow these steps:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call Features".

3	>	С	a	l	1		F	e	a	t	u	r	e	S			
4		S	e	t	t	i	n	g	S								

3. Scroll down the menu and choose "Do Not Disturb".

4		С	a	1	1		W	a	i	t	i	n	g				
5	٨	D	0		Ν	0	t		D	i	S	t	u	r	b		

4. Select 'Enable' / 'Disable' to enable / disable Do Not Disturb function.

1	>	E	n	a	b	l	e							
2	I	D	i	S	a	b	1	e						

4.7 Call Blocking

4.7.1 Blocking a Set of Phone Numbers to be dialed out

Call blocking disallows a set of phone numbers to be dialed out. The blocking number patterns can only be configured on the web interface. (For details of call blocking configuration, refer to Section 7.4 "Call Blocking".)

If the caller dials a number which is blocked by the phone, the following screen will be shown:

N	0	t	a	l	1	0	W	e	d					

On the other hand, a key code can be inputted before the destination number to unblock the outgoing call. The key code can be set on the web interface.

For example, if phone number 2000 is blocked and you set "*173*" to be the key code, you can still place a call to 2000 without blocking by dialing "*173*2000".

4.7.2 Blocking Non-speed Dial Outgoing Calls

The Blocking Non-speed Dial Outgoing Calls feature allows the phone to prevent the user from making Non-Speed Dial Outgoing Calls. By default, this feature is disabled.

If this feature is enabled, a caller is only allowed to make calls with the speed dial buttons, while the keypad buttons 0 - 9, * and # will be disabled.

This feature can only be disabled or enabled from the web administration page (for security reasons).

4.8 Call History

The phone can store the last 30 missed calls, dialed calls and received calls in the call history. Each record contains the phone number and the calling date and time.

To view the call history, follow these steps:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call History".

2	>	С	a	1	1	H	i	S	t	0	r	у				
3		С	a	1	1	F	e	a	t	u	r	e	S			\downarrow
3. Select the type of call history.

1	>	Μ	i	S	S	e	d		С	a	l	1	S				
2		R	e	c	e	i	V	e	d		С	a	1	1	S		\downarrow

4. Scroll up/down to view the history of the selected call type. Select a particular entry will provide further options.

1	>	2	0	0	0	J	0	h	n					
2		3	0	0	0	Μ	a	r	у					\downarrow

5. Select "**Call**" to call the number. Speakerphone will be automatically turned on for calling. Pick up the handset to switch the call to the handset if necessary.

1	>	С	a	1	1							
2		S	a	v	e							\downarrow

6. Select "Save" to save the number.

1		С	a	l	l							
2	>	S	a	v	e							\leftarrow

7. You can choose to save the number to old contact or create it as new contact. For details, please refer to Section 4.10.4 and 4.10.5.

1	>	A	d	d		t	0	0	1	d	С	0	n	t	a	
2		С	r	e	a	t	e	n	e	W	С	0	n	t	a	\downarrow

8. Select "Date & Time" to view the date and time of the call record.

3	>	D	a	t	e	&		Т	i	m	e				1
4		V	i	e	W	D	e	t	a	i	1	S			\downarrow

9. The date and time will be shown on the screen. Press 'OK' button to go back to the list.

2	0	0	5	-	0	9	-	2	5					
1	7	••	0	3	••	2	7							

10. Select "View Details" to view the full number if it is too long to be shown on one line.

4	>	V	i	e	W		D	e	t	a	Ι	1	S			1
5		R	e	m	0	v	e									

11. The full number will be shown on the screen. Press 'OK' button to go back to the list

0	0	1	8	5	2	1	2	3	4	5	6	7	8	9	0	1	2	3	4
5	6	7	8	9	0		С	a	r	0	1								

12. Select "**Remove**" to remove the selected call record.

5	>	R	e	m	0	v	e							1
6		R	e	m	0	v	e	A	1	1				

13. Select "Remove All" to remove the whole history of the selected call type.

5		R	e	m	0	v	e							1
6	٨	R	e	m	0	v	e	A	1	1				

Missed calls will also be shown on the standby screen. Follow these steps to check the missed call information:

1. If an incoming call from remote party is terminated during alerting or exceeds the no answer timeout, the missed calls indicator will be displayed as follows. Press "OK" to go to see the details of the missed call.

Μ	i	S	S	e	d	c	a	l	l	S	:	1			

2. The missed call history will be shown.

.

1	>	2	0	0	0	J	0	h	n					
2		3	0	0	0	Μ	a	r	у					

4.9 Speed Dial

The speed dialing allows a user to dial a phone number by pressing shortcut keys.

The phone can store ten speed-dial phone numbers to be dialed through ten dedicated keys.

4.9.1 To enter Speed Dial menu

To configure the phone numbers for speed dialing, follow these steps:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call Features".

2		С	a	l	l	H	i	S	t	0	r	у				1
3	>	С	a	1	1	F	e	a	t	u	r	e	S			\downarrow

3. Scroll down and enter "Speed Dialing".

2		D	i	r	e	c	t		Ι	P		С	a	l	1		↑
3	>	S	р	e	e	d		D	i	a	1	i	n	g			\downarrow

4. It will show the current phone numbers for speed dialing (blank if not configured). Select the speed dial button number to configure and press 'OK'.

1	>	1	0	0	0							
2		2	0	0	0							

5. You could either set speed dial from direct input (Section 4.9.2) or 4.9.3 phonebook (Section 4.9.3):

1	>	D	i	r	e	c	t	Ι	n	р	u	t					
2		S	e	t		f	r	Р	h	0	n	e	b	0	0	k	

4.9.2 Direct Input

1. After selecting "Direct Input" as in section 4.9.1, input a number, SIP URI (e.g. 2000@1.2.3.4) or IP address (e.g. 123.123.123.123). The maximum number of characters allowed is 60. You could toggle the input method to "A" using the "#" key.

S	р	e	e	d	D	i	a	1	i	n	g	1			A
3	0	0	0	-											

2. Press "OK" to save the change.

4.9.3 Set from Phonebook

1. After selecting "Set fr Phonebook" as in section 4.9.1, you would see a list of entries of phonebook. Choose an entry using "OK".

>	a	р	р	l	e								
	0	r	a	n	g	e							

2 Then, the phone numbers of that contact will be shown. Choose the number you want by using "OK".

>	M	:	1	0	0	0							
	Н	:											\downarrow

4.10 Phone Book

The phone book feature allows a user to store and manage contacts easily. The phone book can hold up to 100 entries. Phone book management can be done through the phone menu only.

4.10.1 Adding Contact in Phone Book from "Adding" Menu

To add a contact into the phone book, follow these steps:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Enter "Phonebook".

1	>	Р	h	0	n	e	b	0	0	k						
2		С	a	1	1		Η	i	S	t	0	r	у			\downarrow

3. Scroll down and enter "Add Contact".

1		С	a	1	1		С	0	n	t	a	c	t			
2	>	A	d	d		С	0	n	t	a	c	t				\downarrow

4. Input the name of person (E.g. Susan).

Ν	a	m	e									A
S	u	S	a	n	-							

5. Press the "OK" button to save the setting and the following screen will show.

Μ	0	b	i	1	e							A
-												

6. Input the mobile number (E.g. 15102442445), SIP URI (e.g. 2000@1.2.3.4) or IP address (e.g. 123.123.123.123). The maximum number of characters allowed is 60. You could toggle the input method to "A" using the "#" key.

Μ	0	b	i	1	e										N
1	5	1	0	2	4	4	2	4	5	Ι					

7. Press the "OK" button to save the setting and the following screen will show.

Η	0	m	e								Ν

8. Input the home number (E.g. 15102442445), SIP URI (e.g. 2000@1.2.3.4) or IP address (e.g. 123.123.123.123). The maximum number of characters allowed is 60. You could toggle the input method to "A" using the "#" key.

Н	0	m	e												N
1	5	1	0	2	4	4	2	4	5						

9. Press the "OK" button to save the setting and the following screen will be shown..

0	f	f	i	c	e							N

Input the office number (E.g. 15102442445), SIP URI (e.g. 2000@1.2.3.4) or IP address (e.g. 123.123.123.123). The maximum number of characters allowed is 60. You could toggle the input method to "A" using the "#" key..

0	f	f	i	c	e										N
1	5	1	0	2	4	4	2	4	5	I					

11. Press the "OK" button to save the setting and the following screen will be shown.

1	>	S	a	v	e									
2		D	i	S	c	a	r	d						

12. Enter "Save" to save the setting.

4.10.2 Making Phone Call from Phone Book

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Enter "Phonebook".

1	>	Р	h	0	n	e	b	0	0	k						
2		С	a	1	1		Η	i	S	t	0	r	У			\downarrow

3. Enter "Call Contact".

1	>	С	a	l	1		С	0	n	t	a	c	t			
2		A	d	d		С	0	n	t	a	c	t				\downarrow

4. A name list like the following will be shown.

>	A	l	e	X								
	A	1	i	c	e							\downarrow

5. Scroll down and the rest of the name list will be displayed.

	A	m	a	r	e								1
>	A	n	d	r	e	w							\downarrow

6. Press an alphabet (e.g. "**c**"), and the cursor will jump to the first name which starts with the inputted letter (E.g. Chris).

>	С	h	r	i	S							
	D	i	c	k								\downarrow

7. Select a contact (e.g. **Chris**). The phone menu will display the details of the selected contact, which M represents mobile number, H represents home number and O represents office number.

>	Μ	:	1	5	1	0	9	5	4	3	2	1	0			
	Η	:	8	5	2	2	7	4	5	2	7	2	7			\downarrow

8. Select the phone number and press the "OK" button. The phone will dial out to the selected number.

С	a	l	l	i	n	g		С	h	r	i	S		[Μ]		
1	5	1	0	9	5	4	3	2	1	0		٠	•	•				

4.10.3 Modifying A Contact in Phone Book

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Enter "Phonebook".

1	>	Р	h	0	n	e	b	0	0	k						
2		С	a	1	1		Η	i	S	t	0	r	У			\downarrow

3. Scroll down and enter "Manage Contact".

2		A	d	d		С	0	n	t	a	c	t					1
3	>	Μ	a	n	a	g	e		С	0	n	t	a	c	t		

4. The contact list will be shown as below.

>	A	l	e	X								
	A	1	i	c	e							\downarrow

5. Press the keypad (e.g. "**5**") to jump to the contacts starting with the respective alphabet (e.g. "**J**").

>	J	a	S	0	n							
	J	e	r	r	у							

6. Select a contact (E.g. "Jason"). The following screen will be shown.

1	>	E	d	i	t		С	0	n	t	a	c	t				
2		R	e	m	0	v	e		С	0	n	t	a	c	t		

7. Enter "Edit Contact". Change the name of the contact if necessary.

Ν	a	m	e									Α
J	a	S	0	n	I							

8. Press the "OK" button. Change the mobile number if necessary.

Μ	0	b	i	1	e										A
1	5	1	0	2	4	4	2	4	5	-					

9. Press the "OK" button. Change the home number if necessary.

Н	0	m	e								A
_											

10. Press the "OK" button. Change the office number if necessary.

0	f	f	i	c	e								A
1	2	3	4	5	6	7	Ι						

11. Enter "Save" to save the setting.

1	>	S	a	v	e									
2		D	i	S	c	a	r	d						

4.10.4 Saving A Number to Old Contact from Call History

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call History".

1		Р	h	0	n	e	b	0	0	k						
2	٨	С	a	l	1		Η	i	S	t	0	r	У			\downarrow

3. Select the call type and enter it (e.g. "Missed Calls").

2	R	e	c	e	i	v	e	d	С	a	1	1	S		Ţ
_		-	-	-	-		-		-		-	_	~		¥

4. Select the phone number.

1	>	1	8	1	0	2	4	5	4	2	2	2				
2		8	5	2	4											\downarrow

5. Scroll down and enter "Save".

1		С	a	l	l							
2	>	S	a	v	e							\downarrow

6. If the number is to be added to an old contact, enter "Add to old Conta".

1	>	A	d	d		t	0	0	1	d	С	0	n	t	a	
2		С	r	e	a	t	e	n	e	W	С	0	n	t	a	

7. The contact list will be shown as below.

>	A	1	l	e	n							
	A	m	a	r	e							\downarrow

8 Press the keypad to select the contact to be updated. (E.g. "8" to jump to those contacts starting with "T".)

>	T	i	m									1
	Τ	0	n	у								\downarrow

9 Select the contact (e.g. "**Tim**"). The details of the contact will be shown.

>	M	:	1	5	1	0	9	5	4	3	2	1	0			
	Η	:														\downarrow

10 Select the phone number to be replaced.

	Η	:														1
>	0		8	5	2	2	7	4	5	2	7	2	7			

11 Press the "OK" button and the following screen will be displayed.

R	e	р	l	a	c	e	d		W	i	t	h				
1	8	1	0	2	4	5	4	2	2	2	?					

12 Press the "OK" button and save the setting.

S	a	v	e	d	•							

13 The phone menu returns to the call history list and a screen like the following will be shown.

1	>	T	i	m		[0]	1	8	1	0	2	4	5	4	2	2
2		J	a	S	0	n												\rightarrow

4.10.5 Saving A Number to New Contact from Call History

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Call History".

1		Р	h	0	n	e	b	0	0	k						
2	>	С	a	l	l		Η	i	S	t	0	r	у			\downarrow

3. Select the call type and enter it (e.g. "Missed Calls").

1	>	Μ	i	S	S	e	d		С	a	l	l	S				
2		R	e	c	e	i	V	e	d		С	a	1	1	S		\downarrow

4. Select the phone number.

1	>	1	8	1	0	2	4	5	4	2	2	2				
2		8	5	2	4											\downarrow

5. Scroll down and enter "Save".

1		С	a	1	l							
2	٧	S	a	v	e							\downarrow

6. Scroll down and enter "Create new Conta".

1		A	d	d		t	0	0	1	d	С	0	n	t	a	
2	٨	С	r	e	a	t	e	n	e	w	С	0	n	t	a	

7. Select the type of phone number to be saved (e.g. mobile).

>	M	:									
	Η	:									\downarrow

8. Input the name of the contact (E.g. Tim).

N	<u>a</u>	m	e								A
Т	i	m	١								

9. Press the "OK" button. Confirm the mobile number.

Μ	0	b	i	1	e										Α
1	8	1	0	2	4	5	4	2	2	2	Ι				

10. Press the "OK" button. Add the home number if necessary.

H	0	m	e										A
1	2	3	4	5	6	7	Ι						

11. Press the "OK" button. Add the office number if necessary.

0	f	f	i	c	e								A
7	6	5	4	3	2	1	Ι						

12. Press the "OK" button and the following screen will be displayed.

1	>	S	a	v	e									
2		D	i	S	c	a	r	d						

13. Press the "OK" button and save the setting.

S	a	V	e	d	•							

14. The phone menu returns to the call history list and a screen like the following will be shown.

1	>	<u>T</u>	i	m		[Μ]	1	8	1	0	2	4	5	4	2	2
2		J	a	S	0	n												\downarrow

4.11 Voice Mail

The "**Message**" button and LED are used for voice mail features. The "**Message**" LED functions as a message waiting indicator (MWI). It will be on if there are new voice messages. The "**Message**" button can be pressed for viewing the number of new and old voice messages.

4.11.1 To activate voice mail function

To activate voice mail function, follow these steps:

- 1 Press the "**Menu** / **Delete**" button to enter the phone menu.
- 2 Scroll down and enter "Settings".

3		С	a	1	l		F	e	a	t	u	r	e	S			↑
4	٧	S	e	t	t	i	n	g	S								\downarrow

3 Scroll down and enter "Users".

2		N	e	t	W	0	r	k						1
3	٨	U	S	e	r	S								\downarrow

4 Scroll down and select the current user (e.g. "User 1").

1	>	U	S	e	r	1						
2		U	S	e	r	2						\downarrow

5 Scroll down and enter "Voice Mail".

9		R	e	g	i	S	t	r	a	t	i	0	n			1
1	0	>	V	0	i	c	e		M	a	i	l				

6 Enter "Use MWI".

1	>	U	S	e		M	W	Ι									
2		U	n	S	0	1	i	c	i	t	e	d	Μ	W	Ι		\downarrow

7 Choose "Yes" and press "OK" button.

1	>	Y	e	S								
2		Ν	0									\downarrow

4.11.2 To listen to voice mail

After enabling the MWI function, the "**Message**" LED will be on if there are new voice messages.

If the "Message" button is pressed when the phone is on-hook, the number of new and old messages will be shown as below:

N	e	W	M	e	S	S	a	g	e	S	:	1			
0	1	d	Μ	e	S	S	a	g	e	S	:	4			

If the "**Message**" button is pressed before or after the phone is off-hook, the phone will call in to the voice mail system. Follow the instructions from the IVRS to retrieve your voice mails.

Before enabling the MWI function, you may need to check if the MWI settings are correct. For more details, please go to Section 6.6.9 Voice Mail Settings.

5. WORKING WITH NAT (STUN)

If the phone is connected to a private network behind NAT (Network Address Translation) device, it will not be able to call the phones on the public network. To overcome this problem, various NAT traversal technologies are developed. Currently, the ATL Berkshire 5000 supports the STUN approach.

5.1 STUN

Before STUN can be used to call the phones on the public network, a public STUN server must be available for connection. After a STUN server is identified for use, follow these steps to enable STUN:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Settings".

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Scroll down and enter "Network".

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter "STUN".

5	>	S	T	U	N											1
6		Μ	A	С		A	d	d	r	e	S	S				

5. Scroll down and select "Server Address".

2	>	S	e	r	v	e	r	A	d	d	r	e	S	S		1
3		S	e	r	v	e	r	Р	0	r	t					\downarrow

6. Input the hostname or IP address of STUN Server.

S	e	r	v	e	r	A	d	d	r	e	S	S			Η
-															

7. Press 'OK' button to confirm. Then select "Server Port".

3	>	S	e	r	V	e	r		P	0	r	t						1
4		R	e	f	r	e	S	h		Ι	n	t	e	r	v	a	1	

8. Input the port number of STUN Server.

S	e	r	v	e	r	P	0	r	t					
_														

9. Press 'OK' button to confirm. Then select "**Refresh Interval**" to change the refresh interval of STUN if necessary.

3		S	e	r	V	e	r		Р	0	r	t						1
4	>	R	e	f	r	e	S	h		Ι	n	t	e	r	V	a	1	

10. Input the refresh interval (0 - 300 seconds).

R	e	f	r	e	S	h	Ι	n	t	e	r	v	a	1		
_																

11. Press 'OK' button to confirm. Then scroll up to "Use STUN" and select it.

1	>	U	S	e		S	t	U	Ν								
2		S	e	r	V	e	r		A	d	d	r	e	S	S		\downarrow

12. Select "Yes" to enable STUN.

1	>	Y	e	S								
2	-	Ν	0									

Now you should be able to call the phones on the public network.

Note: After STUN is enabled, the phone is unable to call the phones on the same private network. If calling the phones on the same private network is needed, disable STUN first before calling.

6. CHANGING SETTINGS

6.1 Ringtone

A user can change the ringtone of the phone. There are 7 ringtones to be chosen, while 2 of which are user-defined and uploaded via the web interface.

The user could set ringtones for internal calls and external calls separately.

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	Р	h	0	n	e								
2		N	e	t	w	0	r	k						\downarrow

4. Enter 'Ringtone'.

1	>	R	i	n	g	t	0	n	e					
2		A	u	d	i	0								\downarrow

5. Enter 'Int Calls' or 'Ext Calls'.

1	<	Ι	n	t	С	a	1	1	S					
2		E	X	t	С	a	1	1	S					

6. Listen to the ringtone playback. Select the desired ringtone.

1	>	R	i	n	g	t	0	n	e	1				
2	-	R	i	n	g	t	0	n	e	2				

6.2 Audio Settings

6.2.1 Voice Codecs

The phone supports G.711 A-law, G.711 µ-Law, G.723.1 and G.729A voice codecs.

G.711 A & µ Law :

- 64 Kbps
- 8 kHz sample rate, with 8 bits per sample.
- A-law is the standard for international circuits.

G.723.1:

- 5.3 Kbps and 6.3 Kbps
- Voice Activity Detection and Comfort Noise Generation.
- Supports Multi-channel operation and Reentrancy.

G.729A:

- 8Kbps
- Voice Activity Detection and Comfort Noise Generation.
- Supports Multi-channel operation and Reentrancy.

To configure voice codecs, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	Р	h	0	n	e								
2		N	e	t	w	0	r	k						\downarrow

4. Enter 'Audio'.

1	>	A	u	d	i	0									
2		D	a	t	e	/	Т	i	m	e					\downarrow

5. Enter 'Voice Codec'.

1	>	V	0	i	c	e		С	0	d	e	c						
2		J	i	t	t	e	r		B	u	f	f	e	r	(m	S	\downarrow

6. Multiple voice codecs are available for selection as below. Scroll up/down to the preferred one:

1	>	G	•	7	1	1	A						
2		G	•	7	1	1	u						\downarrow

7. Press 'OK' to save the change.

6.2.2 VAD (Voice Activity Detection)

The VAD (Voice Activity Detection) function can reduce the bit rate of a call. To enable/disable VAD, follow these steps:

- 7. Press the 'Menu / Delete' button to enter the phone menu.
- 8. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\downarrow

9. Enter 'Phone'.

1	>	P	h	0	n	e								
2		N	e	t	W	0	r	k						\downarrow

10. Enter 'Audio'.

1	>	A	u	d	i	0									
2		D	a	t	e	/	Т	i	m	e					\downarrow

11. Scroll down and enter 'VAD'.

2		J	i	t	t	e	r	B	u	f	f	e	r	(m	S	1
3	٧	V	A	D													\downarrow

12. Select 'Enable' / 'Disable' to enable / disable VAD.

1	>	E	n	a	b	1	e							
2	I	D	i	S	a	b	1	e						

6.2.3 DTMF Generation

The phone supports DTMF generation using in-band and RFC2833. To set the DTMF generation method, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.
- 3. Scroll down and enter 'DTMF Generation'.

3		V	A	D													1
4	٨	D	Τ	Μ	F	G	e	n	e	r	a	t	i	0	n		

4. Three options, namely 'In Band', 'DTMF RFC2833' and 'Both' are available. Scroll up/down to select the preferred mode.

1	>	Ι	n	-	B	a	n	d								
2		D	Τ	Μ	F		R	F	С	2	8	3	3			\downarrow

5. Press 'OK' to save the change.

6.2.4 Adjustable Jitter Buffer

The adjustable jitter buffer helps to absorb the delay jitter caused in packet transmission and reorder the out-of-sequence RTP audio packets in Internet telephony (VoIP) network.

To adjust the jitter buffer size, follow these steps:

1. Press the 'Menu / Delete' button to enter the phone menu.

2. Scroll down and enter 'Settings'.

3		С	a	1	l		F	e	a	t	u	r	e	S			↑
4	>	S	e	t	t	i	n	g	S								\downarrow

3. Enter "Phone".

1	>	Р	h	0	n	e								
2		N	e	t	W	0	r	k						\downarrow

4. Enter "Audio".

1	>	A	u	d	i	0									
2		D	a	t	e	/	Т	i	m	e					\downarrow

5. Scroll down and enter "Jitter Buffer".

1	V	0	i	c	e		С	0	d	e	c						
2	J	i	t	t	e	r		B	u	f	f	e	r	(m	S	\downarrow

6. The current jitter buffer size (in ms) will be shown (e.g. 0 ms).

1	->	0		m	S								
2		2	0		m	S							\downarrow

7. Scroll down and choose the desired size (e.g. 60 ms).

3		4	0	m	S							1
4	<	6	0	m	S							\downarrow

8. Press the 'OK' button to confirm and save the setting.

6.3 Date/Time

The phone supports two types of time settings: NTP (network) and local.

6.3.1 To set time using NTP

To use NTP to set the time, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	Р	h	0	n	e								
2		N	e	t	w	0	r	k						\downarrow

4. Scroll down and enter 'Date/Time'.

3	>	D	a	t	e	/	Τ	i	m	e								1
4		Ι	n	t	e	r	-	D	g		Т	i	m	e	0	u	t	\downarrow

5. Scroll down and enter 'NTP Svr Address'.

2	>	N	Τ	Р		S	V	r		A	d	d	r	e	S	S		1
3		Т	i	m	e	Z	0	n	e									\downarrow

6. Change the NTP server address if necessary.

N	Т	Р		S	v	r		A	d	d	r	e	S	S			Η
р	0	0	l	•	n	t	р	•	0	r	g	Ι					

7. Press 'OK' to save the change. Scroll up to 'Use NTP'.

1	>	U	S	e	N	T	P									
2		Ν	Т	Р	S	v	r	A	d	d	r	e	S	S		\downarrow

8. Select 'Yes' to enable NTP.

1	>	Y	e	S								
2	I	N	0									

6.3.2 To set time locally

To set the time locally, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	Р	h	0	n	e								
2		N	e	t	W	0	r	k						\downarrow

4. Scroll down and enter 'Date/Time'.

3	>	D	a	t	e	/	Т	i	m	e								1
4		Ι	n	t	e	r	-	D	g		Т	i	m	e	0	u	t	\downarrow

5. Scroll down and enter 'Local Settings'.

4		D	a	у	1	i	g	h	t		S	a	v	i	n	g		1
5	٨	L	0	c	a	1		S	e	t	t	i	n	g	S			

6. Input the date and time.

1	>	D	a	t	e							
2		Т	i	m	e							

7. Input date in the format **YYYYMMDD** (E.g. 20050610).

D	a	t	e		(Y	Y	Y	Y	M	M	D	D)			
2	0	0	5	0	6	1	0	Ι									

8. Press the 'OK' button to confirm and save the setting. Input the time in 24-hr format (E.g. 145030).

Т	i	m	e		(Н	H	M	M	S	S	,	2	4	H	r)	
1	4	5	0	3	0	Ι												

9. Press the 'OK' button to confirm and save the setting.

1	>	D	a	t	e							
2		Т	i	m	e							

6.3.3 To change date format

To change the date format of the phone, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	Р	h	0	n	e									
2		Ν	e	t	w	0	r	k						\downarrow	

4. Scroll down and enter 'Date/Time'.

3	>	D	a	t	e	/	Т	i	m	e								1
4		Ι	n	t	e	r	-	D	g		Т	i	m	e	0	u	t	\downarrow

5. Scroll down and enter 'Local Settings'.

4		D	a	у	l	i	g	h	t		S	a	V	i	n	g		1
5	Ν	L	0	c	a	1		S	e	t	t	i	n	g	S			

6. Scroll down and enter "Date Format"

3	>	D	a	t	e		F	0	r	m	a	t				
4		1	2	/	2	4		Η	0	u	r	S				

7. There are three available date formats; namely, International, American and Western Europe. Select the preferred format and press "OK" key to confirm.

1	-	Ι	n	t	e	r	n	a	t	i	0	n	a	1			
2	٨	A	m	e	r	i	c	a	n								\downarrow

6.3.4 To change time format

To change the time format of the phone, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	P	h	0	n	e								
2		N	e	t	w	0	r	k						\downarrow

4. Scroll down and enter 'Date/Time'.

3	>	D	a	t	e	/	Т	i	m	e								1
4		Ι	n	t	e	r	I	D	g		Т	i	m	e	0	u	t	\rightarrow

5. Scroll down and enter 'Local Settings'.

4		D	a	у	1	i	g	h	t		S	a	V	i	n	g		1
5	>	L	0	c	a	1		S	e	t	t	i	n	g	S			

6. Scroll down and enter "12/24 Hours"

3		D	a	t	e		F	0	r	m	a	t				1
4	٧	1	2	/	2	4		Η	0	u	r	S				↓

7. Select the preferred the time format and press "OK" key to confirm.

1	-	1	2	H	0	u	r	S	f	0	r	m	a	t		
2	٨	2	4	Η	0	u	r	S	f	0	r	m	a	t		

6.3.5 To change timezone

To change the timezone of the phone, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

	4	>	S	e	t	t	i	n	g	S										1
--	---	---	---	---	---	---	---	---	---	---	--	--	--	--	--	--	--	--	--	---

5	S	W	i	t	c	h	U	S	e	r				↓
•	~		-	•	•		C	5	•	-				*

3. Enter 'Phone'.

1	>	Р	h	0	n	e								
2		Ν	e	t	w	0	r	k						\downarrow

4. Scroll down and enter 'Date/Time'.

3	>	D	a	t	e	/	Т	i	m	e								↑
4		Ι	n	t	e	r	-	D	g		Т	i	m	e	0	u	t	\downarrow

5. Scroll down and enter 'Timezone'.

3	>	T	i	m	e	Z	0	n	e								1
4		D	a	у	l	i	g	h	t	S	a	V	i	n	g		\downarrow

6. Scroll up/down to select your timezone from the list:

2	7	>	G	Μ	Τ	+	8	:	0	0				1
2	8		G	Μ	Т	+	9	:	0	0				\downarrow

7. Press the 'OK' button to confirm and save the setting.

6.3.6 To enable daylight saving

To enable daylight saving time of the phones, follow these steps:

- 1. Press the "**Menu** / **Delete**" button to enter the phone menu.
- 2. Scroll down and enter "Settings".

3		С	a	1	l		F	e	a	t	u	r	e	S			1
4	>	S	e	t	t	i	n	g	S								\downarrow

3. Enter "Phone".

1	>	Р	h	0	n	e								
2		N	e	t	w	0	r	k						\downarrow

4. Scroll down and enter "Date/Time".

1		A	u	d	i	0									
2	٨	D	a	t	e	/	Т	i	m	e					\downarrow

5. Scroll down and enter "Daylight Saving".

3		Т	i	m	e	Z	0	n	e								
4	٨	D	a	у	1	i	g	h	t	S	a	v	i	n	g		

6. Select daylight saving or standard time. Press "OK" key to confirm.

1	>	D	a	у	l	i	g	h	t	S	a	v	i	n	g		
2	-	S	t	a	n	d	a	r	d	Т	i	m	e				

Please note that:

Change from daylight saving to standard time would move the clock backward one hour (-1).

Change from standard time to daylight saving should move the clock forward one hour (+1).

6.4 Inter-digit Timeout

The inter-digit timeout feature simulates the usage of analog phone. When this feature is enabled, the user does not require to press the "OK" key after keying in the phone number. The phone will automatically call the phone number after the user stop entering digits for the configured timeout duration.

To configure the inter-digit timeout duration, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

3		С	a	l	l		F	e	a	t	u	r	e	S			1
4	٧	S	e	t	t	i	n	g	S								\downarrow

3. Enter "Phone".

1	>	Р	h	0	n	e								
2		Ν	e	t	w	0	r	k						\downarrow

4. Scroll down and enter "Inter Dg Timeout".

3		D	a	t	e	/	Т	i	m	e								1
4	٨	Ι	n	t	e	r	-	D	g		Т	i	m	e	0	u	t	\downarrow

5. User can select among "Disable", "1 second", "2 seconds", "3 seconds", "4 seconds", "5 seconds" and "8 seconds". Press "OK" key to save the timeout duration.

1	>	D	i	S	a	b	1	e						
2	-	1		S	e	c	0	n	d					\downarrow

6.5 Network Settings

The following network settings can be viewed or changed in the phone menu under 'Settings' > 'Network'.

- DHCP
- IP Address
- Subnet Mask
- DNS Server
- Default Gateway
- STUN
- MAC Address
- VLAN
- DSCP
- SIP Port

6.5.1 Enabling DHCP

To enable/disable DHCP for getting IP address and other network configurations, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\rightarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						Î
3		U	S	e	r									\downarrow

4. Enter 'DHCP Setting'.

1		Ι	Р		Ι	n	f	0	•							
2	>	D	Η	С	Р		S	e	t	t	i	n	g			\downarrow

5. Enter 'Use DHCP'.

1	>	U	S	e	D	Η	С	Р					

6. Select 'Yes' / 'No' to enable / disable DHCP.

1	>	Y	e	S								
2	I	N	0									

6.5.2 Static IP Address

To change the static IP address of the phone, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter "Manual IP Conf".

3	>	M	a	n	u	a	l	Ι	P	С	0	n	f	•		1
4		S	T	U	N											\downarrow

5. Enter "IP Address".

1	>	Ι	Р		A	d	d	r	e	S	S					
2		S	u	b	n	e	t		Μ	a	S	k				\downarrow

6. Input the IP address of the phone. E.g.

Ι	Р		A	d	d	r	e	S	S						
1	9	2	•	1	6	8	•	1	٠	2	Ι				

7. Press 'OK' to save the setting.

6.5.3 Subnet Mask

To change the subnet mask, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Scroll and enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter "Manual IP Conf.".

3	>	Μ	a	n	u	a	l	Ι	P	С	0	n	f	•		1
4		S	Τ	U	N											\downarrow

5. Scroll down and enter "Subnet Mask".

1		Ι	P		A	d	d	r	e	S	S					
2	٨	S	u	b	n	e	t		Μ	a	S	k				\downarrow

6. Input the subnet mask of the phone. E.g.

S	u	b	n	e	t		Μ	a	S	k						
2	5	5	•	2	5	5	•	2	5	5	•	0	I			

7. Press 'OK' to save the setting.

6.5.4 DNS Server

To change the DNS server, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Scroll down and enter 'Network'.

2	>	>	N	e	t	W	0	r	k						1
3			U	S	e	r									\downarrow

4. Scroll down and enter "Manual IP Conf.".

3	>	Μ	a	n	u	a	1	Ι	P	С	0	n	f	•		1
4		S	T	U	N											\downarrow

5. Scroll down and enter "DNS Server".

2		S	u	b	n	e	t		Μ	a	S	k				1
3	Ν	D	N	S		S	e	r	v	e	r					\downarrow

6. Input the DNS server of the phone. E.g.

D	N	S		S	e	r	v	e	r							
1	9	2	•	1	6	8	•	1	•	2	5	0	١			

7. Press 'OK' to save the setting.

6.5.5 Default gateway

To change the default gateway, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Scroll down and enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter "Manual IP Conf.".

3	>	Μ	a	n	u	a	1	Ι	P	С	0	n	f	•		1
4		S	Τ	U	N											\downarrow

5. Scroll down and enter "Default Gateway".

3		D	N	S		S	e	r	V	e	r							1
4	>	D	e	f	a	u	l	t		G	a	t	e	W	a	у		

6. Input the default gateway of the phone. E.g.

D	e	f	a	u	l	t		G	a	t	e	W	a	у			
1	9	2	•	1	6	8	•	1	•	2	5	4	I				

7. Press 'OK' to save the setting.

6.5.6 STUN

The STUN settings are used for NAT traversal support. For details, refer to Section 5 "Working with NAT".

6.5.6.1 To enable/disable STUN

To enable/disable STUN, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.
| 2 | > | N | e | t | W | 0 | r | k | | | | | | 1 |
|---|---|---|---|---|---|---|---|---|--|--|--|--|--|--------------|
| 3 | | U | S | e | r | | | | | | | | | \downarrow |

4. Scroll down and enter 'STUN'.

3	>	S	Τ	U	N											1
4		Μ	A	С		A	d	d	r	e	S	S				\downarrow

5. Enter 'Use STUN'.

1	>	U	S	e		S	Т	U	N								
2		S	e	r	v	e	r		A	d	d	r	e	S	S		\downarrow

6. Select 'Yes/ No' to enable or disable STUN.

1	>	Y	e	S		(S	Τ	U	N	0	n	l	у)		
2		Ν	0		(N	0	n	e)							\downarrow

6.5.6.2 To change the server address of STUN server

To change the server address of the STUN server, follow these steps:

- 1. Press the '**Menu** / **Delete**' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'STUN'.

3	>	S	Τ	U	N											\uparrow
4		Μ	A	С		A	d	d	r	e	S	s				\downarrow

5. Scroll down and enter 'Server Address'.

1		U	S	e		S	t	u	n								
2	>	S	e	r	v	e	r		A	d	d	r	e	S	S		\downarrow

6. Input the address of the STUN server. E.g.

S	e	r	V	e	r		A	d	d	r	e	S	S			Η
1	9	2	•	1	•	1	•	1	9	2						

7. Press 'OK' to save the setting.

6.5.6.3 To change the server port of the STUN server

To change the server port of the STUN server, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'STUN'.

3	>	S	Т	U	N											1
4		Μ	A	С		A	d	d	r	e	S	S				\downarrow

5. Scroll down and enter 'Server Port'.

2		S	e	r	v	e	r	A	d	d	r	e	S	S		1
3	>	S	e	r	v	e	r	Р	0	r	t					\downarrow

6. Input the server port of STUN. E.g.

S	e	r	v	e	r	Р	0	r	t					
3	4	7	8	_										

7. Press 'OK' to save the setting.

6.5.6.4 To change the Refresh Interval of the STUN server

To change the Refresh Interval of the STUN server, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'STUN'.

3	>	S	Τ	U	N											1
4		Μ	A	С		A	d	d	r	e	S	S				\downarrow

5. Scroll down and enter 'Refresh Interval'.

3		S	e	r	V	e	r		P	0	r	t						1
4	٨	R	e	f	r	e	S	h		Ι	n	t	e	r	v	a	1	

6. Input the refresh interval of STUN. E.g.

R	e	f	r	e	S	h	Ι	n	t	e	r	V	a	1		
1	0	_														

7. Press 'OK' to save the setting.

6.5.7 MAC Address

MAC address is the hardware address that uniquely identifies the phone in the network. To view the MAC address of the phone, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'MAC Address'.

4	>	Μ	A	С		A	d	d	r	e	S	S				\uparrow
5		V	L	A	Ν											\downarrow

5. The MAC address of the phone will be shown. E.g.

M	A	С		A	d	d	r	e	S	S								
0	0	:	0	1	:	0	2	:	0	3	:	0	4	••	0	5		

6.5.8 VLAN

The VLAN feature allows the phone to transmit Ethernet frames with VLAN ID tagged. The VLAN tagging method and format is compliant with IEEE 802.1q.

The VLAN tagging feature can be configured from both phone menu and web administration page.

6.5.8.1 Turn on/off VLAN tagging feature

By default, the VLAN tagging feature is disabled.

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							↑
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'VLAN'.

5	>	V	L	A	N							1
6		D	S	С	Р							\downarrow

5. Enter 'Use VLAN'.

1	>	U	S	e		V	L	A	N					1
2		V	L	A	Ν		Ι	D						\downarrow

6. Select 'Yes' or 'No'.

1	>	Y	e	S								1
2		Ν	0									\downarrow

6.5.8.2 Set VLAN ID

By default, the VLAN ID is 0.

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							Î
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'VLAN'.

5	>	V	L	A	N							1
6		D	S	С	Р							\downarrow

5. Enter 'VLAN ID'.

2	>	V	L	A	N		Ι	D								1
3		С	0	S		Р	r	i	0	r	i	t	у			

6. The current VLAN ID will be shown. E.g.

V	L	A	N	Ι	D	(0	-	4	0	9	5)		
0	-														

7. Press the 'Menu / Delete' button to erase the current VLAN ID.

V	L	A	N	Ι	D	(0	-	4	0	9	5)		
_															

8. Input the desired VLAN ID. E.g.

V	L	A	N	Ι	D	(0	-	4	0	9	5)		
1	-														

9. Press 'OK' to save the setting.

6.5.8.3 Set CoS Priority

The CoS feature allows the phone to transmit VLAN tagged Ethernet frames with marked priority. The CoS marking method and format is compliant with IEEE 802.1p. By default, the CoS priority value is 0.

The CoS priority value can be configured from both phone menu and web interface.

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'VLAN'.

5	>	V	L	A	N							1
6		D	S	С	Р							\downarrow

5. Enter 'CoS Priority'.

2		V	L	A	N		Ι	D								1
3	٧	С	0	S		Р	r	i	0	r	i	t	у			

6. The current CoS Priority will be shown. E.g.

С	0	S	Р	r	i	0	r	i	t	у	(0	-	7)	
<u>0</u>																

7. Press the 'Menu / Delete' button to erase the current CoS Priority.

С	0	S	Р	r	i	0	r	i	t	у	(0	-	7)	
1																

8. Input the desired CoS Priority. E.g.

С	0	S	Р	r	i	0	r	i	t	у	(0	-	7)	
<u>1</u>																

9. Press 'OK' to save the setting.

6.5.9 DSCP

The DSCP feature allows the IP packets transmitted by the phone marked with a priority. The DSCP/TOS format is compliant with RFC2474. All SIP and voice traffic sent by the phone will be marked with the same DSCP value.

The DSCP value can be configured from both phone menu and web administration page.

The range of DSCP value is from 0 to 63. By default, the DSCP value is 0.

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							\uparrow
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'DSCP'.

6	>	D	S	С	Р									1
7		S	Ι	Р		Р	0	r	t					

5. The current DSCP value will be shown. E.g.

D	S	С	P	V	a	1	u	e	(0	-	6	3)		
0																

6. Press the 'Menu / Delete' button to erase the current DSCP value.

D	S	С	P	V	a	l	u	e	(0	-	6	3)		
_																

7. Input the desired VLAN ID. E.g.

D	S	С	P	V	a	1	u	e	(0	-	6	3)		
1	Ι															

8. Press 'OK' to save the setting.

6.5.10 SIP Port

SIP Port is the signaling port used by the SIP protocol. The default port number occupied by SIP is 5060. To change the SIP Port of the phone, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							Î
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Network'.

2	>	N	e	t	W	0	r	k						1
3		U	S	e	r									\downarrow

4. Scroll down and enter 'SIP Port'.

6		D	S	С	P									1
7	٧	S	Ι	Р		Р	0	r	t					

5. The current SIP Port number will be shown. E.g.

S	5	Ι	Р		Р	0	r	t						
5	5	0	6	0	-									

6. Press the 'Menu / Delete' button to erase the current port number.

S	Ι	Р	Р	0	r	t						
_												

7. Input the desired SIP port number. E.g.

S	Ι	Р		Р	0	r	t						
5	0	8	0	-									

8. Press 'OK' to save the setting.

6.6 User Settings

The phone allows you to store multiple VoIP (SIP) user settings. You can activate the user account of the calling service you would like to use. Each user account consists of the following settings.

- Username
- Password
- Display Name
- Domain
- Registrar Host
- Registrar Port
- Outbound Proxy Host
- Outbound Proxy Port
- Registration
- Voice Mail

6.6.1 Selecting User Account for Configuration

To select a user account for configuration, follow these steps:

1. Press the 'Menu / Delete' button to enter the phone menu.

2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Scroll down and enter 'User'.

3	>	U	S	e	r	S									1
4		E	a	S	у		S	e	t	u	р				

4. Select the user account you would like to configure. E.g. User 1:

1	>	U	S	e	r	1						
2		U	S	e	r	2						\downarrow

5. You may now change the settings for the selected user as described below.

6.6.2 Username

To change username, follow these steps:

1. Enter 'Username'.

1	>	U	S	e	r	n	a	m	e					
2		Р	a	S	S	W	0	r	d					\downarrow

2. Input the username. E.g.

U	S	e	r	n	a	m	e						
1	0	0	0	-									

3. Press the 'OK' button to save the setting.

6.6.3 Password

To change password, follow these steps:

1. Scroll down and enter 'Password'.

2	>	Р	a	S	S	W	0	r	d							1
3		D	i	S	р	1	a	у		N	a	m	e			\downarrow

2. Input the password. For privacy, the inputted character will change to '*'. E.g.

Р	a	S	S	W	0	r	d						
*	*	*	*	-									

3. Press the 'OK' button to save the setting.

6.6.4 Display Name

To change display name, follow these steps:

1. Scroll down and enter 'Display Name'.

3	>	D	i	S	р	1	a	у	N	a	m	e			1
4		D	0	m	a	i	n								\downarrow

2. Input the display name. E.g.

D	i	S	p	l	a	у		N	a	m	e				
A	1	i	c	e		A	u	-							

3. Press the 'OK' button to save the setting.

6.6.5 Domain

To change the SIP domain, follow these steps:

1. Scroll down and enter 'Domain'.

4	>	D	0	m	a	i	n								1
5		R	e	g	i	S	t	r	a	r					\downarrow

2. Input the domain. E.g.

D	0	m	a	i	n													Η
S	i	р	•	a	t	1	t	e	1	e	c	0	m	•	c	0	m	

3. Press the 'OK' button to save the setting.

6.6.6 Registrar

The hostname and port number of SIP registrar can be changed.

6.6.6.1 Registrar Host

To change the hostname or IP address of SIP registrar, follow these steps:

1. Scroll down and enter 'Registrar Host'.

5	>	R	e	g	i	S	t	r	a	r					1
6		R	e	g	•		Р	0	r	t					\downarrow

2. Input the hostname of IP address of registrar host. E.g.

R	e	g	•		P	0	r	t								Η
1	9	2	•	1	6	8	•	1	•	1	0	0				

3. Press the 'OK' button to save the setting.

6.6.6.2 Registrar Port

To change the port number of SIP registrar, follow these steps:

1. Scroll down and enter 'Registrar Port'.

6	>	R	e	g	•		P	0	r	t							1
7		0	u	t	b	0	u	n	d		Р	r	0	X	у		\downarrow

2. Input the port number of registrar host. E.g.

R	e	g	•		P	0	r	t						
5	0	6	0											

3. Press the 'OK' button to save the setting.

6.6.7 Outbound Proxy

The hostname and port number of SIP outbound proxy can be changed.

6.6.7.1 Outbound Proxy Host

To change the hostname or IP address of SIP outbound proxy, follow these steps:

1. Scroll down and enter 'Outbound Proxy'.

7	>	0	u	t	b	0	u	n	d		Р	r	0	X	у		1
8		0	b	•		Р	r	0	X	у		P	0	r	t		\downarrow

2. Input the hostname of IP address of outbound proxy. E.g.

0	u	t	b	0	u	n	d		P	r	0	X	у			Η
1	9	2	•	1	6	8	•	1	•	1	0	1				

3. Press the 'OK' button to save the setting.

6.6.7.2 Outbound Proxy Port

To change the port number of SIP outbound proxy, follow these steps:

1. Scroll down and enter 'Ob. Proxy Port'.

8	>	0	b	•		P	r	0	X	у		P	0	r	t		1
9		R	e	g	i	S	t	r	a	t	i	0	n				\downarrow

2. Input the port number of outbound proxy. E.g.

0	b	•		Р	r	0	X	у	P	r	0	X	у			
5	0	6	0													

3. Press the 'OK' button to save the setting.

6.6.8 Registration

After the user settings are changed, the phone should register to the SIP registrar to update the change.

6.6.8.1 Register/unregister to the registrar

To register to the registrar, follow these steps:

1. Scroll down and enter 'Registration'.

8		0	b	•		P	r	0	X	у		Р	0	r	t		1
9	>	R	e	g	i	S	t	r	a	t	i	0	n				

2. Scroll down and select 'Register Now'. E.g.

2	>	R	e	g	i	S	t	e	r		N	0	W				1
3		U	n	r	e	g	i	S	t	e	r		Ν	0	w		

3. Press the 'OK' button to register. If registration is successful, the following screen should be shown:

U	S	e	r		i	S									
r	e	g	i	S	t	e	r	e	d	•					

6.6.8.2 Check status

You may also want to know the current status of SIP registration. To check the current status of SIP registration, follow these steps:

1. Scroll down and enter 'Registration'.

8		0	b	•		Р	r	0	X	у		Р	0	r	t		1
9	٧	R	e	g	i	S	t	r	a	t	i	0	n				

2. Select 'Check Status'.

:

1	>	С	h	e	c	k		S	t	a	t	u	S			
2		R	e	g	i	S	t	e	r		Ν	0	W			\downarrow

3. Press the 'OK' button to view the status. If the phone has been registered, the following screen will be shown:

U	S	e	r		i	S		r	e	g	i	S	t	e	r	e	d	
Р	r	e	S	S		0	K	/	С	a	n	c	e	1		t	0	\downarrow

4. Otherwise, if the phone currently has not registered, the following screen will be shown:

U	S	e	r		i	S										
u	n	-	r	e	g	i	S	t	e	r	e	d				\downarrow

5. Press the 'OK' button to return to the previous screen.

Besides, to facilitate users to troubleshoot the problem, more meaningful error message will be displayed based on the registration response code.

"User is registered" will be displayed if the response code is 200.

"User is un-registered manually" will be displayed if the expired time of the registration request is zero and the response code is 200.

"User's registration is expired" will be displayed if no re-registration request is sent after registration expired.

"401 Registrar authentication error" will be displayed if the response code is 401.

"407 User is un-registered manually" will be displayed if the response code is 407.

"408 Registrar timeout" will be displayed if the response code is 408.

"423 Registration duration too short" will be displayed if the response code is 423.

"504 Proxy server timeout" will be displayed if the response code is 504.

6.6.8.3 Re-registration if Failed

The "Re-registration if Failed" option defines the duration after which the phone should re-register if the previous registration failed. User can choose among never, 5 minutes, 10 minutes, 30 minutes, 60 minutes and 5 hours. The default value is 5 minutes.

To set the re-registration duration, follow these steps:

- 1. Press "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Settings".

3		С	a	1	1		F	e	a	t	u	r	e	S			1
4	٨	S	e	t	t	i	n	g	S								\downarrow

3. Scroll down and enter "Users".

2		N	e	t	W	0	r	k						Î
3	٨	U	S	e	r	S								\downarrow

4. Enter "User 1".

1	>	U	S	e	r	1						1
2		U	S	e	r	2						\downarrow

5. Scroll down and enter "Registration".

8		0	b	•		Р	r	0	X	у		Р	0	r	t		1
9	>	R	e	g	i	S	t	r	a	t	i	0	n				\downarrow

6. Scroll down and enter "Re-Reg If Fail".

3		U	n	r	e	g	i	S	t	e	r	N	0	W			1
4	>	R	e	-	r	e	g	•		Ι	f	F	a	i	l		\downarrow

7. User can select among Never, 5 minutes, 10 minutes, 30 minutes, 60 minutes, 5 hours. Press "OK" key will store the value input.

1	>	N	e	v	e	r									
2	I	5		m	i	n	u	t	e	S					\downarrow

6.6.8.4 Registration Duration

The "Registration Duration" option defines the duration after which the phone should update the registration. User can choose among 15 seconds, 1 minutes, 5 minutes, 10 minutes, 20 minutes, 40 minutes, 1 hour, 2 hours, 3 hours and 5 hours for the registration duration. The default value is 1 hour.

To set the registration duration, follow these steps:

- 1. Press "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Settings".

3		С	a	l	1		F	e	a	t	u	r	e	S			1
4	>	S	e	t	t	i	n	g	S								\downarrow

3. Scroll down and enter "Users".

2		N	e	t	W	0	r	k						1
3	٨	U	S	e	r	S								\downarrow

4. Enter "User 1".

1	>	U	S	e	r	1						1
2		U	S	e	r	2						\downarrow

5. Scroll down and enter "Registration".

8		0	b	•		Р	r	0	X	у		P	0	r	t		1
9	>	R	e	g	i	S	t	r	a	t	i	0	n				\downarrow

6. Scroll down and enter "Reg Duration".

4		R	e	-	r	e	g	•		Ι	f		F	a	i	l	e	d	1
5	>	R	e	g		D	u	r	a	t	i	0	n						\downarrow

7. User should be able to choose among 15 seconds, 1 minute, 5 minutes, 10 minutes, 20 minutes, 40 minutes, 1 hour, 2 hours, 3 hours and 5 hours. Press "OK" key to save the setting.

1	>	1	5		S	e	c	0	n	d	S				
2		1		m	i	n	u	t	e						\downarrow

6.6.8.5 Auto Renew

Under normal situation, the registration will be expired according to the registration duration. In order to extend the registration, the phone needs to send registration request before the registration expired. The "Auto Renew" option defines whether the phone should renew the registration automatically.

To change the "Auto Renew" settings, follow these steps:

- 1. Press "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Settings".

3		С	a	1	1		F	e	a	t	u	r	e	S			1
4	٨	S	e	t	t	i	n	g	S								\downarrow

3. Scroll down and enter "Users".

2		N	e	t	W	0	r	k						1
3	٨	U	S	e	r	S								\downarrow

4. Enter "User 1".

1	>	U	S	e	r	1						1
2		U	S	e	r	2						\downarrow

5. Scroll down and enter "Registration".

8		0	b	•		Р	r	0	X	у		P	0	r	t		1
9	>	R	e	g	i	S	t	r	a	t	i	0	n				↓

6. Scroll down and enter "Auto Renew".

5		R	e	g		D	u	r	a	t	i	0	n			1
6	٨	A	u	t	0		R	e	n	e	w					\downarrow

7. Enter "Yes" (or "No") to save the settings.

1	-	>	Y	e	S							1
2			N	0								\downarrow

6.6.9 Voice Mail Settings

To configure the voice mail settings, follow these steps:

- 1. Press "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Settings".

3		С	a	1	l		F	e	a	t	u	r	e	S			1
4	>	S	e	t	t	i	n	g	S								\downarrow

3. Enter "Users".

2		N	e	t	W	0	r	k						1
3	٨	U	S	e	r	S								\downarrow

4. Enter "User 1".

1	>	U	S	e	r	1						
2		U	S	e	r	2						\downarrow

5. Scroll down and enter "Voice Mail".

9		R	e	g	i	S	t	r	a	t	i	0	n			1
1	0	7	V	0	i	c	e		M	a	i	1				

6. Scroll down and enter "Unsolicited MWI".

2	>	U	n	S	0	l	i	c	i	t	e	d		M	W	Ι		
3		Μ	W	Ι		S	e	r	V	e	r		Η	0	S	t		\downarrow

7. Select to "Enable" or "Disable" "Unsolicited MWI" (depending on server's MWI mode).

1		Ε	n	a	b	1	e							
2	^	D	i	S	a	b	1	e						\downarrow

8. Scroll down and enter "MWI Server Host".

3	>	M	W	Ι	S	e	r	v	e	r	Η	0	S	t		1
4		Μ	W	Ι	S	e	r	V	e	r	Р	0	r	t		\downarrow

9. Input MWI server host and press "OK" button.

Μ	W	Ι		S	e	r	v	e	r		H	0	S	t			H
1	0	•	6	•	1	0	0	•	5	1							

10. Scroll down and enter "MWI Server Port".

4	>	M	W	Ι		S	e	r	v	e	r	Р	0	r	t		1
5		V	0	i	c	e		M	a	i	1	E	X	•			

11. Input MWI server port and press "OK" button.

M	W	Ι		S	e	r	v	e	r	Р	0	r	t			
5	0	8	0	1												

12. Scroll down and enter "Voice Mail Ex.".

4		M	W	Ι		S	e	r	V	e	r	Р	0	r	t		1
5	٨	V	0	i	c	e		M	a	i	1	E	X	•			

13. Input the voice mail extension number. Press "OK" button to save it.

V	0	i	c	e	M	a	i	1	E	X	•			
1	2	3	Ι											

6.6.10 Switch User

After the multiple user settings are configured, you can conveniently switch between different user accounts in order to use different VoIP services.

To switch to another user account, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Switch User'

5	>	S	W	i	t	c	h		U	S	e	r				1
6		D	i	a	g	n	0	S	t	i	c	S				\downarrow

3. Select the user account whose Registration Status is desired to be shown on the phone idle screen and whose settings are to be used for making outgoing calls.

1		1	0	0	0							
2	-\ \	1	2	3	4							

4. Press the 'OK' button to confirm. The phone will switch to the selected user account as shown below:

	1	2	3	4		A	1	i	c	e		A	u					
0	6	i	0	6	-	1	3		1	5	:	3	5	:	5	6		

6.7 Restore Default Settings

To restore all settings of the phone to the default values, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	Р	h	0	n	e								
2		N	e	t	W	0	r	k						\downarrow

4. Enter 'Restore Defaults'.

5	>	R	e	S	t	0	r	e		D	e	f	a	u	1	t	S	1
6		Η	Т	T	Р		A	u	t	0		С	0	n	f	i	g	\downarrow

5. Scroll up to 'Reset Defaults'. The following screen will be shown:

1	>	R	e	S	e	t		D	e	f	a	u	l	t	S		
2		С	a	n	c	e	1										

6. Press the 'OK' button to confirm. The phone will reboot with default settings and launch the *Easy Setup* wizard.

6.8 HTTP Auto Configuration

The phone will try to download the configuration file from the configured HTTP server on boot-up. If the file is downloaded successfully, the configurations specified in the file will be adopted. Otherwise, the configurations stored will be used.

This feature can be enabled or disabled from the phone menu or web administration page.

6.8.1 To save server address for HTTP Auto Config

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter '**Phone**'.

1	>	Р	h	0	n	e								
2		N	e	t	w	0	r	k						\downarrow

4. Enter 'HTTP Auto Config'.

6	>	H	Т	T	Р		A	u	t	0		С	0	n	f	i	g	1
7		F	i	r	m	w	a	r	e		U	р	g	r	d			

5. Enter 'Server Address'.

1		U	S	e		A	u	t	0		С	0	n	f	i	g		
2	٨	S	e	r	V	e	r		A	d	d	r	e	5	S			

6. Press the 'OK' button to save the server address entered.

S	e	r	v	e	r	A	d	d	r	e	S	S			Η
-															

6.8.2 To enable or disable HTTP Auto Config

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter '**Phone**'.

1	>	Р	h	0	n	e								
2		Ν	e	t	W	0	r	k						\downarrow

4. Enter 'HTTP Auto Config'.

6	>	H	Τ	Т	Р		A	u	t	0		С	0	n	f	i	g	1
7		F	i	r	m	W	a	r	e		U	р	g	r	d			

5. Enter 'Use Auto Config'.

1	>	U	S	e		A	u	t	0		С	0	n	f	i	g		
2		S	e	r	V	e	r		A	d	d	r	e	S	S			

6. Select 'Yes' or 'No'.

1	>	Y	e	S								
2		N	0									

6.9 Firmware Upgrade

The firmware of the phone can be upgraded by HTTP or TFTP. To upgrade the phone firmware, follow these steps:

- 1. Ensure the new firmware has been placed in the HTTP or TFTP server.
- 2. Set the HTTP or TFTP server address. Refer to Section 6.9.16.9.1 "**HTTP Server**" and 6.9.2 "**TFTP Server**" for details.
- 3. Select the "**Protocol**" for firmware upgrade. Refer to Section 6.9.3.
- 4. Choose to "Upgrade Now". Refer to Section 6.9.4.

6.9.1 HTTP Server

You could use HTTP server for upgrading the firmware of the phone. To change the IP address of the HTTP server, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	Р	h	0	n	e								
2		N	e	t	w	0	r	k						\downarrow

4. Enter 'Firmware Upgrade'.

5		H	Т	Т	Р		A	u	t	0		С	0	n	f	i	g	1
6	٨	F	i	r	m	W	a	r	e		U	р	g	r	d			

5. Enter 'HTTP Server'.

2	>	Η	Τ	Т	Р	S	e	r	v	e	r				1
3		Т	F	Т	Р	S	e	r	v	e	r				\downarrow

6. Input the IP address of the HTTP Server as shown in the example below. The input method is by default, H (hostname) or, after toggling with "#" key, I (IP address).

Т	F	Τ	Р		S	e	r	V	e	r					H
1	0	•	6	•	9	6	•	1	8						

7. Press 'OK' to save the setting. The following screen would be the output.

Р	1	e	a	S	e	W	a	i	t	•	•	•			

8. Then, you would return to:

2	>	H	T	Т	P	S	e	r	V	e	r				1
3		Т	F	Т	P	S	e	r	v	e	r				\downarrow

6.9.2 TFTP Server

You could also use TFTP server for upgrading the firmware of the phone. To change the IP address of the TFTP server, follow these steps:

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter 'Phone'.

1	>	Р	h	0	n	e								
2		N	e	t	W	0	r	k						\downarrow

4. Enter 'Firmware Upgrade'.

5		Η	Т	Τ	Р		A	u	t	0		С	0	n	f	i	g	1
6	>	F	i	r	m	W	a	r	e		U	р	g	r	d			

5. Enter '**TFTP Server**'.

3	>	T	F	T	P		S	e	r	V	e	r				1
4		Р	r	0	t	0	c	0	1							

6. Input the IP address of the TFTP Server, say:

Т	F	Т	Р		S	e	r	V	e	r					
1	9	2	•	1	6	8	•	1	•	1					

7. Press 'OK' to save the setting. The following screen would be the output.

Р	1	e	a	S	e	W	a	i	t	•	•	•			

8. Then, you would return to:

3	>	Т	F	Τ	Р		S	e	r	V	e	r				1
4		Р	r	0	t	0	c	0	l							

6.9.3 Protocol

Apart from setting the HTTP or TFTP server address, you need to choose the protocol you prefer in order to start the firmware upgrade.

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	W	i	t	c	h		U	S	e	r				\downarrow

3. Enter '**Phone**'.

1	>	Р	h	0	n	e								
2		Ν	e	t	W	0	r	k						\downarrow

4. Enter 'Firmware Upgrade'.

5		H	T	T	Р		A	u	t	0		С	0	n	f	i	g	1
6	>	F	i	r	m	W	a	r	e		U	р	g	r	d			

5. Enter '**Protocol**'.

3		T	F	T	P		S	e	r	V	e	r				1
4	٨	Р	r	0	t	0	c	0	1							

6. Choose either HTTP or TFTP.

1	-	H	T	T	Р							
2	٧	Т	F	Т	Р							

7. Press 'OK' to save the setting. The following screen would be the output.

Р	l	e	a	S	e	W	a	i	t	•	•	•			

8. Then, you would return to:

3		T	F	Τ	P		S	e	r	V	e	r				1
4	>	Р	r	0	t	0	c	0	1							

6.9.4 Upgrade Now

After the HTTP or TFTP server address and the desired protocol are selected, you could upgrade the firmware using the "**Upgrade Now**" option.

6.9.4.1 Initiate Upgrade Process

- 1. Press the 'Menu / Delete' button to enter the phone menu.
- 2. Scroll down and enter 'Settings'.

4	>	S	e	t	t	i	n	g	S							1
5		S	w	i	t	c	h		U	S	e	r				\downarrow

3. Enter '**Phone**'.

1	>	Р	h	0	n	e								
2		N	e	t	w	0	r	k						\downarrow

4. Enter 'Firmware Upgrade'.

5		H	Τ	Т	Р		A	u	t	0		С	0	n	f	i	g	1
6	٨	F	i	r	m	W	a	r	e		U	р	g	r	d			

5. Select '**Upgrade Now**'. Refer to 6.9.4.2 for HTTP Firmware Upgrade and 6.9.4.3 for TFTP Firmware Upgrade.

1	>	U	р	g	r	a	d	e		N	0	W				
2		Η	T	Т	Р		S	e	r	v	e	r				\downarrow

6.9.4.2 Confirmation to have HTTP Firmware Upgrade

1. If you have chosen the HTTP protocol for firmware upgrade, there will be a version check of the phone's current software against the server's latest software. The checking version screen will be displayed:

С	h	e	c	k	i	n	g		V	e	r	S	i	0	n		
Р	l	e	a	S	e		W	a	i	t	•	•	•				

2. After the version check, the result screen is as shown below. There may be no updates or there may be a newer version with version number displayed.

In both cases, you could either press "OK" key to reinstall/ upgrade or 'Cancel' to cancel reinstall/ upgrade.

Case 1. No newer version.

Ν	0		u	р	d	a	t	e	S							
R	e	i	n	S	t	a	1	1		n	0	w	?			

Case 2. Newer version exists.

Ν	e	w	e	r		v	e	r	:	2	2	1	0	1	5	5	9	
U	р	g	r	a	d	e		n	0	w	?							

For the upcoming screens for firmware upgrade, refer to Section 6.9.4.4.

6.9.4.3 Confirmation to have TFTP Firmware Upgrade

1. If you have chosen the TFTP protocol for firmware upgrade, there will be no version check but the confirmation screen will be displayed. You could select "Confirm Upgrd" to upgrade. You could select "Cancel Upgrd" or press 'Cancel' key to cancel reinstall/ upgrade.

1		С	0	n	f	i	r	m		U	р	g	r	d			
2	٨	С	a	n	c	e	1		U	р	g	r	d				

For the upcoming screens for firmware upgrade, refer to Section 6.9.4.4.

6.9.4.4 HTTP or TFTP Firmware Upgrade process

After the upgrade process starts, the upgrade status screens will be shown..

The upgrade status screen will show '#' if the phone can connect to the HTTP or TFTP server and download the firmware. In case the status screen shows ' \mathbf{X} ', it indicates that the phone cannot connect to the HTTP or TFTP server or download the firmware.

NOTE: The upgrade process may take several minutes depending on the system and network environment. Do not power off the phone during the process as it may introduce detrimental effect to the flash of the phone.

The phone will be rebooted after firmware upgrade.

7. WEB INTERFACE CONFIGURATION

Apart from phone menu, the phone can also be administrated and configured remotely from a web browser. The web interface supports both Internet Explorer and Mozilla Firefox browsers.

7.1 Accessing the Web Interface

To access the web interface, the user should visit the URL *http://<IP address of phone>/* from the browser. Before the user can view and configure the phone settings, he must first login the system. The login name is "*admin*". The default password is "*1111*".

7.2 Phone Users

You can configure the user settings from the web interface. Two user accounts are supported. For each user account, you can set the Username, Password, Display Name, Domain, Registrar Host, Registrar Port, Outbound Proxy, Outbound Proxy Port, Registration, Voice Mail.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

Configuration of iSip-phone by Administrator - Microsoft Internet Explorer File Edit View Favorites Tools Help ^ Phone Users <<|top|bottom|>> Menu Active User OUser 1 OUser 2 [+] Global Configuration The User Currently Used [O] Change Password Username - User 1 4000 [O] Configuration Archive The Name of User 1 [O] Firmware Upgrade Password - User 1 •••• The Password of User 1 [+] Phone Log Display Name - User 1 mo The DisplayName of User 1 Domain - User 1 10.6.100.119 The Domain of User 1 This website is powered by webserver Registrar Host - User 1 (optional) The Registrar Host of User 1 Registrar Port - User 1 (optional) 5060 The Registrar Port of User 1 Outbound Proxy - User 1 (optional) The Outbound Proxy of User 1 Outbound Proxy Port - User 1 (optional) 5060 The Outbound Proxy Port of User 1 Internet

For details of user settings, please refer to Section 6.6 "User Settings".

Configuration of iSip-phone by Adm	inistrator - Microsoft Internet Explorer		
File Edit View Favorites Tools Hel	p		
Menu	Re-Registration if Fail - User 1	5 minutes Re-Registration Time Interval When Response is Failed	•
[+] <u>Global Configuration</u>	Registration Duration - User 1	1 hour	
[0] <u>Configuration Archive</u>	Auto Renew - User 1	⊙ Yes ○ No Auto Renew the Registration	
[0] <u>Firmware Upgrade</u>	Enable WMI - User 1?	● ENABLE ● DISABLE Enable/Disable the MWI Service	
<u>[+] Phone Log</u>	The MWI Server Address - User 1	Set the MWI Server Address	
This website is powered by	The MWI Server Port - User 1	5080 Set the MWI Server Port	
webserver	The Voice Mail Server Address - User 1	The Voice Mail Server address can be either extension number or full SIP URI	
	Username - User 2	4000 The Name of User 2	
	Password - User 2	The Password of User 2	
	Display Name - User 2	No Name The DisplayName of User 2	/
🙆 👘 Internet			
7.3 Call Forward Feature

You can configure call forward feature from the web interface. Three call forward modes can be configured, namely **Unconditional**, **On Busy** and **On No-Answer**. You can enable / disable each mode by clicking the radio button. You can input the forward number for each mode. For **On No-Answer** mode, the no answer timeout can also be configured. The three options are "5 **Seconds**", "10 **Seconds**" and "15 **Seconds**".

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

For details of the call forward feature, please refer to Section 4.5 "Call Forward".

Configuration of iSip-phone by Adm	inistrator - Microsoft Internet Explorer	
File Edit View Favorites Tools Hel	p	and the second
Menu	Call Feature (Call Forward)	< <itop bottom >></itop bottom >
[-] <u>Global Configuration</u>	Unconditional Call Forward Enabled	○ ENABLE ○ DISABLE Set the Unconditional Call Forward Enable Flag
<u>Call Feature (Call</u>	Unconditional Call Forward Target	
Forward)		Set the Unconditional Call Forward Target (e.g. 1000)
<u>Call Feature (Speed</u>		Sectile Ontonialional Cannol Ward Parget (e.g. 1966)
Dialing Mapping)	On Busy Call Forward Enabled	O ENABLE O DISABLE
<u>Call Feature (Call</u>		Set the On Busy Call Forward Enable Flag
Blocking)		
<u>Call Feature (Others)</u>	On Busy Call Forward Target	
Phone Media		Set the On Busy Call Forward Target (e.g. 1000)
Phone Date Time	On No Answer Call Forward Enabled	
Inter-Digit Timeout		Set the No Answer Call Forward Enable Flag
Phone Network		
NAT Configuration	On No Answer Call Forward Target	
(STUN)		Set the No Answer Call Forward Target (e.g. 1000)
Phone Users		
PPPoE Login Settings	On No Answer Call Forward Timeout Value	● 5 Seconds ● 10 Seconds ● 15 Seconds
• Easy Setup Trigger 📃		Set the No Answer Call Forward Timeout Value
<u>About Phone</u>	Call Feature (Speed Dialing Mapping)	<< top bottom >>
O Change Password		
[O] <u>Configuration Archive</u>	Speed Dialing 1 Mapping	
		Set the Speed Dialing Button Mapping
[0] <u>Firmware Upgrade</u>		
	Speed Dialing 2 Mapping	· · · · · · · · · · · · · · · · · · ·
<u>ا</u>		🥏 Internet

7.4 Call Blocking Feature

You could choose to block non-speed dial outgoing calls or just a few number patterns on the web interface.

10 **Call Blocking Number** patterns can be set on the web interface. Also, the **Key Code** can be set to unblock a call when needed.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

For details of the usage of call blocking feature, please refer to Section 4.7 "Call Blocking".

Configuration of iSip-phone t	y Admi	nistrator - Microsoft Internet Explorer		
File Edit View Favorites Tools	Help			1
		Call Feature (Call Blocking)	<< <u> top bottom >></u>	
au	=	Block Non-Speed Dial Outgoing Calls	OYes ⊙No Block Non-Speed Dial Outgoing Calls	(
[+] <u>Global Configuration</u>		Key Code to Unblock The Dialing Number	*173*	
[+] <u>Tone / Music</u>			ne Super Key Code that Unblock the Dialing Number	
[0] <u>Change Password</u>		The Call Blocking User Number 1	1.2.3.4	
[0] <u>Configuration Archive</u>			Set the Call Blocking User Number	
e E		The Call Blocking User Number 2	🔹 🔹 🔮 Internet	<u> </u>

7.5 Other Call Features

Call Waiting and **Do Not Disturb** features can be set on the web interface. You can enable/disable the features by clicking the respective radio button.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

For details of call waiting and Do Not Disturb feature, please refer to Section 4.2 "Call Waiting" and Section 4.6 "Do Not Disturb".



7.6 Ringtone

Ringtone could be uploaded and selected on the web interface. The maximum number of ringtones is 7, including 2 user-defined ones. You could refer to section 6.1 for setting ringtone in phone menu.

To upload ringtone, the page is "Tone / Music > Ringtones". The user could upload 2 user-defined ringtones. The filetype allowed for the ringtone file to be uploaded is RTTTL (RingTone Text Transfer Language) (".txt").

Configuration of iSip-phone by Ac	lministrator - Microsoft Internet Ex	xplorer .	
File Edit View Favorites Tools He	lp		2
atl	Ringtones	replace the current one(if any).	^
	Name of Current Ringtone 6	Ringtone 1 Name of Current Ringtone 6	
[+] <u>Global Configuration</u>	Name of New Ringtone 6		3
[<u>-</u>] <u>Tone / Music</u>		Name of New Ringtone 6 to be uploaded. (Optional. If not entered, the name specified in the ringtone file	
<u>Ringtones</u> <u>Music On Hold</u>		will be used. Maximum number of characters for the name is 16. Alphanumeric characters, space and underscore are allowed.)	
[O] <u>Change Password</u>	File of New Ringtone 6	Browse	
[0] <u>Configuration Archive</u>		Please choose your New Ringtone 6 file here. (File type allowed: .txt)	
[0] <u>Firmware Upgrade</u>		(Note that the Current Ringtone 6 file (if any) will be replaced	
[0] <u>Registration</u>			
[+] <u>Diagnosis</u>		Upload Ringtone Upload New Ringtone 6 file.	
v	Upload a New Ringtone 7 and	replace the current one(if any).	~
🕘 Done		🌍 Internet	

To choose ringtone, the page is "Global Configuration". The user could choose the ringtone under the section "Phone Ringtone" for internal calls and external calls separately.

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🕘 Co	onfi	guration	of iSip-pl	none by	y Admi	nistrator - Microsof	t Internet Explorer	
File	Edi	t View	Favorites	Tools	Help			at 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 19
	/	0			~	Phone Ringtone	<< top bottom >>	
	/	a	t				Internal Call's Ringtone Fingtone 5 Set the Ringtone for Internal Calls, i.e. calls from Internal Domain	
[±] <u>G</u>	lobal Co	onfiqurati	<u>on</u>			External Call's Ringtone Ringtone 2 Set the Ringtone for External Calls, i.e. calls from	
	<u>1</u> • <u>1</u>	Rington Music O	<u>es</u> n Hold		>	Phone Media	External Domain << top bottom >>	
ど Do	one						🔮 Internel	

7.7 Music On Hold

User-defined call-hold music could be uploaded on the web interface. The music uploaded will be played when a call is being held at the phone. The user could also remove the user-defined one and use the default tone.

To upload the music clip for call hold, the page is "Tone / Music > Music On Hold". The file format is raw with attributes: u-law, mono, 64kbps (.snd).

🗈 Configuration of iSip-phone by Administrator - Microsoft Internet Explorer 📃 🗖 🔀			
File Edit View Favorites Tools Help			
atl	Music On Hold	place the current one(if any).	<
(+) <u>Global Configuration</u>	Current Music on Hold	User-defined Music Current Music On Hold. (Music On Hold could be either User-defined Music or Default	
[<u>-</u>] <u>Tone / Music</u> • <u>Rinatones</u> • Music On Hold	File of New Music On Hold	Tone.) Browse Please choose your New Music On Hold file here. (File type allowed: .snd)	
[O] <u>Change Password</u>		Upload Music Upload the New Music On Hold file.	
[0] <u>Configuration Archive</u> [0] <u>Firmware Upgrade</u>		(Note that the current Music On Hold file (if any) will be replaced after upload.)	
[O] <u>Registration</u>	Remove User-defined Music	Remove User-defined Music Remove the previously uploaded Music On Hold file (if any), so that Default Tone will be used for Music On Hold	
[<u>+</u>] <u>Diagnosis</u>		(Note that the current Music On Hold file (if any) will be replaced after pressing this button.)	
🖉 Done		a Internet	M

7.8 Phone Audio

Four types of audio settings can be configured on the web interface, namely Voice Codec, VAD, CNG and DTMF Generation.

For Voice Codec, there are 4 options: G.711A, G.711u, G.723.1 and G.729A.

Both VAD and CNG can be enabled or disabled.

For DTMF Generation, there are three options: In-band, RFC2833 and Both.

You can select the appropriate option by clicking the respective radio button.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

For details of audio settings, please refer to Section 6.1 "Audio Settings".

Configuration of iSip-phone by Adm	ninistrator - Microsoft Internet Explorer		. 🗗 🗙		
File Edit View Favorites Tools He	lp		.		
		Set the Call Waiting Mode			
Menu					
	Do Not Disturb enable?	O ENABLE O DISABLE			
		Set the Do Not Disturb Mode			
<u>[-]</u> <u>Global Configuration</u>	Dhone Media	< <tr><<td><<td>></td></td></tr>	< <td>></td>	>	
< <td>></td>	>				
	Phone Heald				
<u>Call Feature (Call</u>					
Forward)					
<u>Call Feature (Speed</u>	Voice Codec	G.711A G.7110 G.723.1 G.729A			
<u>Dialing Mapping)</u>		The preferred voice codec			
<u>Call Feature (Call</u>	VAD Enable?		=		
Blocking)	no cristio.	The preferred VAD setting			
<u>Call Feature (Others)</u>					
Phone Media	DTMF Generation Mode	○ DTMF In Band ○ DTMF RFC2833 ⊙ Both			
<u>Phone Date Time</u>		The DTMF Generation Mode			
Inter-Digit Timeout					
Phone Network	Jitter Buffer Time	50			
NAT Configuration		The preferred jitter buffer time (0 - 999 ms)			
(STUN)	Rhana Data Tiwa	a altern Hantham Is a			
Phone Users	Phone Date Time				
<u>PPPoE Login Settings</u>					
• Easy Setup Trigger		0			
<u>About Phone</u>	Auto time update (NTP) Enable?	ENABLE O DISABLE The Auto time undets (NTD) actions			
		The Auto unie update (NTP) secong			
O <u>Change Password</u>	NTP Server Address	poolintp.org			
		The NTP Server setting			
[O] <u>Configuration Archive</u>		The Win Server Setting			
[O] Firmware Ungrade	Update Real Time Clock?	• TRUE • FALSE			
L - J - miniare opgrade		Update The Real Time Clock If Kernel Clock Changed			
			×		
		🧭 Internet			

7.9 Phone Date Time

You can choose "**NTP**" or "**Local**" or "**Daylight Saving**" for setting the date & time of the phone on the web interface.

If NTP is enabled, you are also needed to fill in the NTP Server Address.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

For details of Date and Time, please refer to Section 6.3 "Date/Time".

Configuration of iSip-phone by Adm	inistrator - Microsoft Internet Explorer		_ 7 🛛
File Edit View Favorites Tools Hel	P		
		The DTMF Generation Mode	^
Menu	litter Buffer Time	0	
	Sicci Barler Hille	The proferred litter buffer time (0 , 000 ms)	
-] Global Configuration		The preferred Jucer burier diffe (0 - 999 fits)	
	Phone Date Time	<< top bottom >>	
<u>Call Feature (Call</u>			
Forward)			
<u>Call Feature (Speed</u>	Auto time update (NTP) Enable?	● ENABLE ● DISABLE	
Dialing Mapping)		The Auto time update (NTP) setting	
<u>Call Feature (Call</u>			
Blocking)	NTP Server Address	pool.ntp.org	=
<u>Call Feature (Others)</u>		The NTP Server setting	
Phone Media	Undate Real Time Clock?		
Phone Date Time		Update The Real Time Clock If Kernel Clock Changed	
Inter-Digit Timeout			
Phone Network	Timezone	GMT +8:00 🔽	
<u>NAT Configuration</u>		The Local timezone	
(STUN)	Deuliske Geuies	O Standard Time O Daulight Course Time	
<u>NAT Configuration</u>	Daylight Saving	Daylight Saving Time Setting	
(TURN)			
Phone Users PRD 5 Locid Contribution	Date Format	● International ○ America ○ British	
<u>PPPOE Login Settings</u> Satur Triagen		Date Formatting	
<u>Easy Setup Trigger About Doops </u>	10/04 Harma		
About Phone	12/24 Hours	12/24 Hours Setting	
[0] <u>Change Password</u>			
	Inter-Digit Timeout	<< top bottom >>	
[0] <u>Configuration Archive</u>			
F _1_,			
	Inter-Diait Timeout?	3 Seconds 🗸	~

7.10 Firmware Upgrade

You can upgrade the phone firmware by clicking "**Firmware Upgrade**" button from the web interface. You could choose to use HTTP or TFTP server for firmware upgrade. Remember to enter the respective server address.

For details of firmware upgrade, please refer to Section 6.9 "Firmware Upgrade".

Configuration of iSip-phone by Adm	ninistrator - Microsoft Internet Explorer		
File Edit View Favorites Tools Help			A
atl	Firmware Upgrade	ograde the phone's firmware.	
	HTTP (Use HTTP server to upgrade the phone's firmware.)	<< top bottom >>	
[+] <u>Global Configuration</u>			=
[<u>-</u>] <u>Tone / Music</u>	HTTP Server Address	10.6.96.18	
● <u>Ringtones</u> ● <u>Music On Hold</u>	Firmware Upgrade using HTTP	The Dornain/ IP Address of the HTTP Server. HTTP Firmware Upgrade	
[0] <u>Change Password</u>		upgrade the phone's intriware using HTTP Server. (Please refresh this page after around 5 minutes to	
[0] <u>Configuration Archive</u>		check for the new firmware version. If you are unable to refresh this page, please go to the phone directly for	
[0] <u>Firmware Upgrade</u>		further investigation.)	
[0] <u>Registration</u>	Firmware Version at the phone	22101607 Firmware Version at the phone.	
[±] <u>Diagnosis</u>	TFTP (Use TFTP server to upgrade the phone's firmware.)	<< top bottom >>	
✓		and a second sec	✓

File Edit View Favorites Tools Help	
TFTP (Use TFTP server to upgrade	<u>^</u>
the phone's firmware.)	
7 ati	
TFTP Server Address 192.168.1.1	
The IP Address of the TFTP Server.	
(+) <u>Global Configuration</u> Firmware Upgrade using TFTP TFTP Firmware Upgrade	
Upgrade the phone's firmware using TFTP Server.	
[_] Tone / Music (Please refresh this page after around 5 minutes to	
check for the new firmware version. If you are unable	
<u>Ringtones</u> to refresh this page, please go to the phone directly for	
<u>Music On Hold</u> <i>further investigation.)</i>	
[0] Change Bassword	
Firmware Version at the phone 22101607	
O Configuration Archive	
[O] <u>Firmware Upgrade</u>	в
O Registration Save Configuration	
	~

7.11 Phone Network

You can the network settings of the phone from the web interface. There are a number of settings: DHCP, IP Address, SIP Port, Subnet Mask, DNS Server, Default Gateway, VLAN, VLAN ID, CoS Priority, DSCP Value, Broadcast Address and MAC Address.

You can enable or disable DHCP and VLAN. For IP Address, SIP Port, Subnet Mask, DNS Server, Default Gateway, VLAN ID, CoS Priority and DSCP Value, you can input their appropriate settings. For Broadcast Address and MAC address, they can be viewed only.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

- -

For details of network settings, please refer to Section 6.5 "Network Settings".



Configuration of iSip-phone by A	ministrator - Microsoft Internet Explorer		
File Edit View Favorites Tools He	P		
	Default Gateway	10.6.96.1 The Default Gateway of the network adaptor	
∕ati	VLAN enabled?	● ENABLE ● DISABLE	
[+] <u>Global Configuration</u>	VLAN ID	0 VLAN ID (0-4095)	
[_] <u>Tone / Music</u>	CoS Priority	0	
<u>Ringtones</u> <u>Music On Hold</u>	DSCP Value	0	
[0] <u>Change Password</u>		DSCP Value (0-63)	
[0] <u>Configuration Archive</u>	Broadcast Address	10.6.111.255 The Broadcast Address of the phone	
[0] <u>Firmware Upgrade</u>	MAC Address	00:15:F8:00:01:00	
[0] <u>Registration</u>		The MAC Address of the phone	
[±] <u>Diagnosis</u>	NAT Configuration (STUN)	<< top bottom >>	~
🕘 Done		Internet	.:

7.12 NAT Configuration (STUN)

You can configure STUN from the web interface. There are four NAT (STUN) settings: Use STUN, Server Host, Server Port and Refresh Interval.

You can enable / disable STUN by clicking the radio button.

For STUN Server Host and Port Number, you can input the appropriate settings.

For **Refresh Interval**, the range is 0 - 300 seconds.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

For details of STUN, please refer to Section 5 "Working with NAT (STUN)".

Configuration of iSip-phone by Add	nistrator - Microsoft Internet Explorer	
File Edit View Favorites Tools Hel;		1
• <u>Call Feature (Speed</u>	NAT Configuration (STUN) <pre><< top bottom >></pre>	<u>^</u>
<u>Dialing Mapping)</u>		
• <u>Call Feature (Call</u>		
<u>Blocking)</u>	Enable STUN? OENABLE ODISABLE	
<u>Call Feature (Volume</u>	Enable or Disable STUN	
Control)		
<u>Call Feature (Others)</u>	STUN Server Host 0.0.0.0	
<u>Phone Ringtone</u>	The Host of the STUN Server (1.0.0.0 -	
• <u>Phone Media</u>	254.254.253)	
<u>Phone Date Time</u>		
• Inter-Digit Timeout	STUN Server Port Number 3478	
Phone Network	The Port Number of the STUN Server (1-65536)	
NAT Configuration		
(STUN)	Refresh Interval (0-300 Seconds) 10	
Phone Users	The Port Binding Refresh Interval (0-300	
HTTP Auto	Seconds)	≡
Configuration		
Easy Setup Trigger	Phone Users << <pre><</pre>	
Phone Log		
About Phone		
	Active User Olser 1	
[+] Tone / Music	O User 2	
	The User whose Registration Status would be	
[O] <u>Change Password</u>	shown on the phone idle screen and whose	
•	settings would be used for making outgoing calls	~
The second se	ession9 🥥 Internet	

After setting the STUN Server Host, you could use the Diagnostic tool NAT Discovery. For more details of NAT Discovery, also refer to Section 5 "Working with NAT (STUN)".



7.13 HTTP Auto Config

You can enable / disable the **HTTP** Auto Config from web interface. Also, you could enter the HTTP Auto Config Server address there.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

For details of HTTP Auto Config, please refer to Section 2.2 "HTTP Auto Configuration".

Configuration of iSip-phone by Add	inistrator - Microsoft Internet Explorer	
File Edit View Favorites Tools Help		
• Inter-Digit Timeout	HTTP Auto Configuration <<< top bottom >>	^
Phone Network		
 <u>NAT Configuration</u> 		
(STUN)	Enable HTTP Auto Configuration? O ENABLE O DISABLE	
<u>Phone Users</u>	- HTTP Auto Configuration	
HTTP Auto		
<u>Configuration</u>	HTTP Auto Config Server Address	
• Easy Setup Trigger	The Domain/IP address of the HTTP Server for	
• Phone Log	HTTP Auto Configuration	
<u>About Phone</u>		
	Easy Setup Trigger << top bottom >>	
[+] Tone / Music		
[O] <u>Change Password</u>	Enable Wizard On Startup? OENABLE ODISABLE	
falls a second	Easy Setup Trigger	
[O] <u>Configuration Archive</u>		
	Phone Log <<(top bottom >>	
[0] <u>Firmware Opgraue</u>		
[O] Registration		
L - J	Enable Phone Log? O ENABLE O DISABLE	
[_] <u>Diagnosis</u>	Phone Log	
• Phone Log	About Phone <<<[top bottom >>	=
• SIP Message Log		
• NAT Discovery		×
ど Done	🔮 Internet	

7.14 Easy Setup Trigger

You can enable / disable the **Easy Setup** from web interface. If it is enabled, the wizard will launch on phone startup. Please note that the network settings will remain unchanged.

After finishing the configuration, click "**bottom**" to jump to bottom of page. Save the settings by clicking "**Save configuration**".

For details of Easy Setup, please refer to Section 2.2 "Easy Setup".

Configuration of Berkshire 5000 by A	Administrator - Microsoft Internet Explorer				
<u>File E</u> dit <u>V</u> iew F <u>a</u> vorites <u>T</u> ools <u>H</u> elp					1
🌀 Back 👻 🕥 - 💌 🗟 🏠 🔎 Sea	rch 👷 Favorites 🚱 🔗 🛛 🖗 🗾 💽	•		Links	
	Easy Setup Trigger		<< top bottom >>		~
∕ati	Enable Wizard On Startup?	OENABLE ODISABLE Easy Setup Trigger			
[<u>-]</u> <u>Global Configuration</u>	Phone Log		<< top bottom >>		
• <u>Call Feature (Call</u> <u>Forward)</u> • Call Feature (Speed	Enable Phone Log?	O ENABLE O DISABLE Phone Log			
Dialing Mapping) • Call Feature (Call	About Phone		<< top bottom >>		
<u>Blocking)</u> <u>Call Feature (Volume</u> Control) 	Phone Hardware Version	00010001			
<u>Call Feature (Others)</u> <u>Phone Ringtone</u>	Phone Software Version	22101770			
<u>Phone Media</u> <u>Phone Date Time</u> <u>Inter-Digit Timeout</u>		Phone Software Version			
Phone Network NAT Configuration (STUN)	Save Configuration				
Phone Users					~
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7.15 About Phone

You can check the hardware and software versions of the phone from the web interface.

Configuration of Berkshire 5000	by Administrator - Microsoft Internet Explorer	
<u>File E</u> dit <u>V</u> iew F <u>a</u> vorites <u>T</u> ools <u>H</u> el		N
🌀 Back 🝷 🕥 - 📓 📓 🏠 🔎	Search 👷 Favorites 🤣 😥 👻 🥃 🔟 - 🧫 💽 -	Links "
atl	Enable Wizard On Startup? O ENABLE O DISABLE Easy Setup Trigger Phone Log	
[-] <u>Global Configuration</u>	Enable Phone Log? O ENABLE O DISABLE	
<u>Call Feature (Call</u> <u>Forward)</u> <u>Call Feature (Speed</u>	About Phone Log About Phone	
Dialing Mapping) Call Feature (Call Blocking) 	Phone Hardware Version 00010001	
<u>Call Feature (Volume</u> <u>Control)</u>	Phone Hardware Version	
<u>Call Feature (Others)</u> <u>Phone Ringtone</u> <u>Phone Media</u>	Phone Software Version	
<u>Phone Date Time</u> <u>Inter-Digit Timeout</u>		
<u>Phone Network</u> <u>NAT Configuration</u> <u>(STUN)</u>	Save Configuration	
Phone Users		~
E	💓 Intern	et .

7.16 Password Change

You can change the login password of the web interface. Click "**Change Password**" on the left. Input the new password twice and confirm the change by clicking "**Save**".

Configuration of iSip-phone by Adr	ninistrator - Microsoft Internet Explorer	_ 7 🗙
File Edit View Favorites Tools He	elp	
<u>Call Feature (Call</u> Forward)	Change of Administrator's Password	
<u>Call Feature (Speed</u> Dialing Manning)	Specify your new password below and then press the "Save" button to set your new password.	
<u>Call Feature (Call</u> Blocking)	New Password	
<u>Call Feature (Others)</u> Phone Media	New Password again	
Phone Date Time Inter-Digit Timeout	Save	
Phone Network NAT Configuration		
(STUN)		
PPPoE Login Settings Easy Setup Trigger		
<u>About Phone</u>		
[O] <u>Change Password</u>		
[O] <u>Configuration Archive</u>		
[0] <u>Firmware Upgrade</u>		
~		
E Done	🔹 🔹 🖉 Internet	

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7.17 Configuration Archive

Configuration archive allows users to backup and restore phone configuration. It assumes the user has a broadband router with DHCP enabled in the intranet. The PC, broadband router and the phone are located on the same subnet.

Click the "Configuration Archive" will display the configuration archive page on the right frame.

Configuration of iSip-phone by Adn	ninistrator - Microsoft Internet Explorer	_ 7 🛛
File Edit View Favorites Tools He	alp	.
Menu	Configuration Archive	
[<u>-</u>] <u>Global Configuration</u>	Configuration Backup Config Backup	
<u>Call Feature (Call</u> <u>Forward)</u> <u>Call Feature (Speed</u>	Backup the phone configuration to a text file. (Please do not modify the configuration file. Otherwise the configuration restoration will be failed.)	
<u>Call Feature (Call</u> <u>Blocking) Call Feature (Others)</u>	Choose your config backup file and Press "Config Restore" to recover the phone config.	
Phone Media Phone Date Time Inter-Digit Timeout	Configuration Restore Please choose your config backup file here.	
 <u>Phone Network</u> <u>NAT Configuration</u> (STUN) 	Config Restore Restore the previous stored phone configuration.	
Phone Users PPPoE Login Settings Easy Setup Trigger		
About Phone [0] Change Password		
 [0] <u>Configuration Archive</u> [0] <u>Firmware Upgrade</u> 		
< >	🔹 👘 Internet	

To back up the phone configurations, press the "Config Backup" button to download the phone configuration file to the PC. The browser will prompt the user to save the file on PC.

File Dowr	nload X									
?	Some files can harm your computer. If the file information below ooks suspicious, or you do not fully trust the source, do not open or save this file.									
	File name: backupConfig.cfg									
	File type:									
	From: 10.6.104.15									
	Would you like to open the file or save it to your computer?									
	<u>Open</u> <u>Save</u> Cancel <u>M</u> ore Info									
	Always ask before opening this type of file									

To restore the phone configurations, click the "Browse" button. The browser will prompt the user to select the configuration file from the PC.

Choose file						? ×
Look <u>i</u> n	: 🛅 Specification	ı	-	(† 🖻 🖻	* 🎫 🕇	
My Recent Documents Desktop My Documents My Computer	i draft i RFC i UMA i fxs_fxo					
My Network	File <u>n</u> ame:			•	-	<u>O</u> pen
Places	Files of type:	All Files (*.*)		•	-	Cancel

After selecting the file, press the "Config Restore" button to restore the saved configurations to the phone.

7.18 Inter-Digit Timeout

The web interface of inter-digit timeout is shown below. User can select among Disable, 1 second, 2 seconds, 3 seconds, 4 seconds, 5 seconds, 8 seconds.

Configuration of iSip-phone by Admi	inistrator - Microsoft Internet Explorer		_ @ 🛛
File Edit View Favorites Tools Help)		
Menu	Auto time update (NTP) Enable?	● ENABLE ○ DISABLE	
[_] <u>Global Configuration</u>	NTP Server Address	The Auto time update (NTP) setting	
• <u>Call Feature (Call</u> <u>Forward)</u>		The NTP Server setting	
• <u>Call Feature (Speed</u> <u>Dialing Mapping)</u>	Update Real Time Clock?	• TRUE • FALSE Update The Real Time Clock If Kernel Clock Changed	
<u>Call Feature (Call</u> <u>Blocking)</u> <u>Call Feature (Others)</u>	Timezone	GMT +8:00 V The Local timezone	
Phone Media Phone Date Time Inter Digit Timeout	Inter-Digit Timeout	<< top bottom >>	
Phone Network NAT Configuration	Inter-Digit-Timeout?	3 Seconds V	
(STUN) Phone Users PPPoE Login Settings	Phone Network	1 Seconds 2 Seconds 3 Seconds 4 Seconds	
Easy Setup Pringer About Phone	Phone DHCP enable?	5 Seconds 8 Seconds DISABLE Phone DHCP	
[0] <u>Configuration Archive</u>	Phone IP Address	10.6.104.27 The IP Address of the network adaptor	
[0] <u>Firmware Upgrade</u>	UDP Listening Port (Transport 1)	5060 The Natwork Listening Port	✓
e		🥶 Internet	

7.19 Speed Dialing Button Mapping

You can modify the speed dialing button mapping from the web interface. The mapping can be an extension number or phone number.

The layout of the web interface is shown below.

Configuration of iSip-phone by	, Admi	inistrator - Microsoft Internet Explorer		
File Edit View Favorites Tools	Help			*
	<u> </u>	Call Feature (Speed Dialing Mapping)	<< top bottom >>	
atl		Speed Dialing 1 Mapping	4600 Set the Speed Dialing Button Mapping	
[+] <u>Global Configuration</u>		Speed Dialing 2 Mapping	4600@10.5.5.6 Set the Speed Dialing Button Mapping	
[<u>-</u>] <u>Tone / Music</u>	≡	Speed Dialing 3 Mapping		
 <u>Ringtones</u> <u>Music On Hold</u> 		Speed Dialing 4 Mapping	Sec the Speed Dialing Bacton Mapping	
[O] <u>Change Password</u>		Speed Dialing 5 Mapping	Set the Speed Dialing Button Mapping	
[0] <u>Configuration Archive</u>			Set the Speed Dialing Button Mapping	
[0] <u>Registration</u>		Speed Dialing 6 Mapping	Set the Speed Dialing Button Mapping	
[±] <u>Diagnosis</u>		Speed Dialing 7 Mapping	Set the Speed Dialing Button Mapping	
	~			~
é			🍎 Internet	.::

7.20 Re-Registration Time

7.20.1 Re-registration If Fail

You can choose the re-registration period among never, 5 minutes, 10 minutes, 30 minutes, 60 minutes and 5 hours.



7.20.2 Registration Duration

You can choose the registration duration among 15 seconds, 1 minutes, 5 minutes, 10 minutes, 20 minutes, 40 minutes, 1 hour, 2 hours, 3 hours and 5 hours.

Configuration of iSip-phone by Adu	ministrator - Microsoft Internet Explorer		. 🗗 🗙
File Edit View Favorites Tools H	elp		
<u> </u>		The Outboaria Proxy of Oser 1	^
Menu	Outhound Proxy Port - User 1 (ontional)	5060	
		The Outhound Provy Port of Licer 1	
[-] Clobal Configuration			
	Re-Registration if Fail - User 1	5 minutes 👻	
<u>Call Feature (Call</u>		Re-Registration Time Interval When Response is Failed	
<u>Forward)</u>			
<u>Call Feature (Speed</u>	Registration Duration - User 1	1 hour V	
Dialing Mapping)		15 seconds Registration	
<u>Call Feature (Call</u>	Auto Renew - User 1	5 minutes	
<u>Blocking)</u>		10 minutes 20 minutes = Registration	
<u>Call Feature (Others)</u>		40 minutes	
Phone Media	Enable WMI - User 1?	1 hour DISABLE	
Phone Date Time		2 hours the MWI Service	
• Inter-Digit Timeout	The MWI Server Address - User 1	5 hours	
Phone Network	THE MWI SERVER Address - Oser I	Cet the MMAIL Convert Address	
NAT Configuration		Set the MWI Server Address	
(STUN)	The MWI Server Port - User 1	5080	
Phone Users		Set the MWI Server Port	
PPPoE Login Settings			=
• Easy Setup Trigger	The Voice Mail Server Address - User 1		
<u>About Phone</u>		The Voice Mail Server address can be either extension	
		number or full SIP URI	
[0] Change Password			
[0] <u>Configuration Archive</u>	Username - User 2	4000	
		The Name of User 2	
[0] <u>Firmware Upgrade</u>	Dessword - Hoor 2		
< >	Passwold - Osei 2	The Backword of Liser 2	~
e		🥥 Internet	

7.20.3 Auto Renew

You can choose between "Yes" or "No" for auto renewing the registration.

Configuration of iSip-phone by Admin	nistrator - Microsoft Internet Explorer		PX
File Edit View Favorites Tools Help			_
		THE Outbound Proxy of Oser 1	^
Menu	Outbound Proxy Port - User 1 (optional)	5060	
		The Outbound Proxy Port of User 1	
[<u>-</u>] <u>Global Configuration</u>	Re-Registration if Fail - User 1	5 minutes 👻	
<u>Call Feature (Call</u>		Re-Registration Time Interval When Response is Failed	
Forward)	Desistration Duration - User 1	1 hour .	
<u>Call Feature (Speed</u>	Registration Duration - Oser 1	Duration of the Depictration	
<u>Dialing Mapping)</u>		Duration of the Registration	
<u>Call Feature (Call</u>	Auto Renew - User 1	⊙ Yes ○No	
<u>Blocking)</u>		Auto Renew the Registration	
<u>Call Feature (Others)</u>			
• <u>Phone Media</u>	Enable WMI - User 1?	● ENABLE ○ DISABLE	
Phone Date Time		Enable/Disable the MWI Service	
• Inter-Digit Timeout	The MMU Conjer Address - User 1		
Phone Network	The MWI Server Address - Oser I		
NAT Configuration		Set the MWI Server Address	
(STUN)	The MWI Server Port - User 1	5080	
Phone Users		Set the MWI Server Dort	
PPPoE Login Settings			
• Easy Setup Trigger	The Voice Mail Server Address - User 1		
<u>About Phone</u>		The Voice Mail Server address can be either extension	
		number or full SIP URI	
O <u>Change Password</u>			
	Username - User 2	4000	
		The Name of User 2	
O Firmware Upgrade			
	Password - User 2	••••	
		The Deceward of Hear 2	⊻
۲. Constant and a second se		🥑 Internet	

7.21 Phone Log

The phone log is provided through the web administration interface. It is useful for troubleshooting. To view the phone log, the user first needs to **ENABLE** the phone log. Remember to **Save Configuration** after the selection. (The user could **DISABLE** phone log on the same interface.)

Configuration of iSip-phone by Adr	ninistrator - Micros	soft Internet Explorer			
File Edit View Favorites Tools Help)				1
[0] <u>Firmware Upgrade</u>	Phone Log			<< top bottom >>	
[0] <u>Registration</u>		Enable Phone Log?	O ENABLE O DISABLE		
[<u>-</u>] <u>Diagnosis</u>			Phone Log		
Phone Log STP Message Log	About Phone			<< top bottom >>	
• SIF Message Log					
		Phone Hardware Version	00010001		
ê				🌍 Internet	

To view the log messages, click the link for **Phone Log** on the left column of the web interface ("**Diagnosis** > **Phone Log**") as shown above. The web interface of the phone can display the most recent 500 log entries.

The user could clear the phone log using the Clear Phone Log button.



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7.22 SIP Trace

The SIP trace is provided through the web administration interface. It can show the SIP traffic exchanged in the phone. The web interface can display the most recent 200 SIP messages.



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8. DIAGNOSTICS

8.1 PING

The PING utility is provided in phone menu to perform connectivity check for troubleshooting. Users can ping the specific IP address and hostname.

To ping an IP address or hostname, follow these steps:

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Diagnostics".

6	>	D	i	a	g	n	0	S	t	i	c	S				1
7		A	b	0	u	t										

3. Enter "Ping".

1	>	Р	i	n	g										
2		Т	r	a	c	e	r	0	u	t	e				

4. Input the Ping destination and press the 'OK' button

Р	Ι	N	G		D	e	S	t	i	n	a	t	i	0	n		Ι
2	0	2	•	2	0	2	•	2	0	2	•	2	0	2	-		

5. Press 'OK' button to confirm, then the result should display as follows:

P	Ι	N	G		R	e	S	u	1	t							
0	%	L	0	S	S		R	Т	Т	I	2	0	m	S			

8.2 Traceroute

The TRACEROUTE utility is provided in phone menu to print the route taken by the packets. Users can trace the route of packets destined for the specific IP address and hostname.

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Diagnostics".

6	>	D	i	a	g	n	0	S	t	i	c	S				1
7		A	b	0	u	t										

3. Enter "Traceroute".

1		Р	i	n	g										
2	٧	Т	r	a	c	e	r	0	u	t	e				

4 Input the destination" and press the 'OK' button

Т	r	a	c	e	r	0	u	t	e		D	e	S	t	•		Ι
2	0	2	•	2	0	2	•	2	0	2	•	2	0	2	_		

5. The result should be displayed as follows:

Τ	r	a	c	e	r	0	u	t	e		R	e	S	u	l	t		
1	•	1	9	2	•	1	6	8	٠	1	1	1	٠	2	5	4		\downarrow

6. Scroll down and the rest of the result will be displayed as follows:

1	•	1	9	2	•	1	6	8	•	1	1	1	•	2	5	4		
		4	0	0	0		3	0	0	0		2	0	0	0	m	S	\downarrow

8.3 Phone Log

The phone log is provided through the web administration interface of for troubleshooting. For the details of Phone Log, please refer to Section 7.21 Phone Log.

8.4 SIP Trace

The SIP trace is provided through the web administration interface for tracing the SIP messages. exchanged in the phone. For the details of Phone Log, please refer to Section 7.22 SIP Trace.

8.5 NAT Discovery

The NAT Discovery utility is provided in phone menu to enable the user to check the NAT type and external IP address using the configured STUN server. The NAT type results include No NAT, Symmetric UDP Firewall, Full cone NAT, Restricted cone NAT and Port restricted cone NAT.

NAT type results	Short forms used in phone menu and web interface
No NAT	No NAT
Symmetric UDP Firewall	Symmetric UDP FW
Full cone NAT	Full cone NAT
Restricted cone NAT	Restricted NAT
Port restricted cone NAT	Port restrict NAT

The diagnostic tool can be used even when STUN is disabled.

You could also check the NAT type using the web administration interface.

Note that the STUN server address should be entered at a different location beforehand, which is, 'Menu / Delete' > '4 Settings' > '2 Network' > '6 STUN' > '2 Server Address', prior to the diagnosis.

To check the NAT type on the phone menu,

- 1. Press the "Menu / Delete" button to enter the phone menu.
- 2. Scroll down and enter "Diagnostics".

6	>	D	i	a	g	n	0	S	t	i	c	S				1
7		A	b	0	u	t										

3. Scroll down and enter "NAT Discovery".

2		Τ	r	a	c	e	r	0	u	t	e						
3	٨	N	A	Т		D	i	S	c	0	v	e	r	у			

4. The result should be displayed as follows:

Р	0	r	t		R	e	S	t	r	i	c	t		N	A	Т		
Ι	Р	••	1	2	3	•	1	2	3	٠	1	2	3	•	1	2	3	

APPENDIX A PHONE MENU STRUCTURE

1. Phone Book

Call Contact Add Contact Manage Contact 2. Call History Missed Calls (max 30 entries) Call Save Date and Time View Details Remove Remove All **Received Calls** (max 30 entries) Call Save Date and Time View Details Remove Remove All **Dialed** Calls (max 30 entries) Call Save Date and Time View Details Remove Remove All **3.** Call Features Call Forward Unconditional Forward Address Use Uncon. Yes No On Busy Forward Address Use Busy Yes

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No On No-Answer Forward Address Timeout (secs) 5 seconds 10 seconds 15 seconds Use No-Answer Yes No Status Query Cancel All Speed Dialing Call Waiting Use Call Waiting Enable Disable Call Waiting Dur 5 Seconds 10 Seconds 15 Seconds Do Not Disturb Enable Disable Phone Ringtone Int Calls Ext Calls Audio Voice Codec G.711A G.711u G.723.1 G.729A Jitter Buffer (ms 0 ms 20 ms 40 ms 60 ms 80 ms 100 ms 120 ms

4. Settings

140 ms VAD Enable Disable **DTMF** Generation In-Band DTMF RFC2833 Both Date/Time Use NTP Yes No NTP Svr Address Timezone (33 options) Daylight Saving Daylight Saving Standard Time Local Settings Date Time Date Format International American Western Europe 12/24 Hours 12 Hours format 24 Hours format Show/Hide Second Show Seconds Hide Seconds Inter-Dg Timeout Disable 1 second 2 seconds 3 seconds 4 seconds 5 seconds 8 seconds **Restore Defaults Reset Defaults** Cancel HTTP Auto Config

Use Auto Config Yes No Server Address Firmware Upgrd Upgrade Now HTTP Server **TFTP Server** Protocol HTTP TFTP Network IP Info. **DHCP** Setting Use DHCP Yes No Manual IP Conf. **IP** Address Subnet Mask **DNS** Server Default Gateway **STUN** Use STUN Yes (STUN Only) No (None) Server Address Server Port **Refresh Interval** MAC Address VLAN Use VLAN Yes No VLAN ID **CoS** Priority DSCP SIP Port Users User 1 Username Password Display Name

Domain Registrar Reg. Port **Outbound Proxy** Ob. Proxy Port Registration Check Status Register Now Unregister Now Re-reg. If Fail Never 5 minutes 10 minutes 30 minutes 1 hour 5 hours **Reg Duration** 15 seconds 1 minute 5 minutes 10 minutes 20 minutes 40 minutes 1 hour 2 hours 3 hours 5 hours Auto Renew Yes No Voice Mail Use MWI Yes No Unsolicited MWI Enable Disable **MWI Server Host** MWI Server Port Voice Mail Ex.

Username Password

- -

User 2

Display Name Domain Registrar Reg. Port Outbound Proxy Ob. Proxy Port Registration Check Status Register Now Unregister Now Re-reg. If Fail Never 5 minutes 10 minutes 30 minutes 1 hour 5 hours **Reg Duration** 15 seconds 1 minute 5 minutes 10 minutes 20 minutes 40 minutes 1 hour 2 hours 3 hours 5 hours Auto Renew Yes No Voice Mail Use MWI Yes No Unsolicited MWI Enable Disable **MWI Server Host** MWI Server Port Voice Mail Ex.

Easy Setup

5. Switch User
6. Diagnostics

Ping Traceroute NAT Discovery

7. About

IP Address HW Version SW Version

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APPENDIX B SUMMARY OF IP PHONE FUNCTION KEY MAPPING

Function – 1	3-way Conferencing
Function – 2	Call Transfer
Function – 3	No Mapping
Function – 4	No Mapping
Function – 5	No Mapping
Function – 6	No Mapping
Function – 7	Do Not Disturb
Function – 8	No Mapping
Function – 9	Redial
Function – 0	No Mapping

APPENDIX C WORKING WITH 3RD PARTY SIP SERVERS

- 1. SIPphone.com
- Configure "IP Address" (e.g. 171.198.202.41). Please make sure it is an free address and not behind NAT (Network Address Translation)
- Configure "Subnet Mask" (e.g. 255.255.255.0)
- Change the "Outbound Proxy" to **198.65.166.131**
- Change the "Outbound Proxy Port" to 5060
- Change the "Registrar Host" to **198.65.166.131**
- Change the "Registrar Port" to **5060**
- Configure "Default Gateway" (e.g. 171.198.202.62)
- Make sure "DHCP" is disabled
- Set the "Domain" to be proxy01.sipphone.com

After the above configuration, you may then select "Register Now" to login to sipphone.com proxy.

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