MizuPhone Customization

About

MizuPhone –the SIP softphone for Windows PC- can be customized for your company's needs in the way that users have to type only their username and password on the login form to start the usage. All settings can be preconfigured (server address, sip settings, language, color scheme, icons, etc).

Customization process

Common customization will be done for free of charge. If you wish a new feature that is not already implemented then MizuTech may apply a customization fee. Delivery time is usually a few days after MizuTech received all the data and materials needed for the customization work. You can just send us an email with all your requirements and we will take care of the details. If you wish to know more details, please read below.
The configuration can be hardcoded in the software and / or we can use external configuration files to control the softphone.
Most of the settings can be changed later runtime (from a config file stored on your server)

Basic settings

-Your company name
-Brand name (name of the softphone)
-Company web site URL
-Your SIP server address (domainname or ipaddress)
-Logo file (preferably a 180x80 gif or png with transparent background)

Other settings

1. Preconfigured server settings (domainname or ipaddress, port, proxy, dialplan, etc). Can the user change the predefined account setting?
2. List of ip addresses and/or domain names where the software can connect to. (Otherwise a separate activation key will be required for all copies). If the user cannot change their predefined settings, then this point is irrelevant.
3.Other company logo (displayed on the login screen and on the dial page if requested) (size: 180x80, format: cur,pcx,gif,ani,jpg,bmp,ico,wmf,emf)
4. Any other direct URL that should be integrated in main menu (for example direct links to your customer service or credit recharge pages, etc)
5. Service support email address
6. A link to credit charge and a link to user information page, if any
7. Describe if you would like any other modification on the GUI
8. Other default settings changes (archive preferences, etc)
9. You can send us 2 test accounts if you would like us to further test the softphone compatibility with your sip server
10. Specify if MizuTech account option needs to be included (profile storage, routing helper, search for users). Yes or No
11. Credit balance display method (describe, if your server has such feature)
12. Application (default) language (may need some time to translate if already not supported or we can send a text file in English that has to be translated into your language)
13. Other?

Skin editor

You can fine-tune your configuration in the softphone Appearance settings (File Menu -> Advanced Settings ->Appearance)
When you are done, export your setting (File menu -> Export current customized settings) and send the exported files to MizuTech.

Default settings

There are several "levels" where the default settings can be stored.
Levels with higher priority always overwrite the default settings with lover priority.
There are a few settings that cannot be overwritten by higher levels (for example the list of provider ip address which ensures proper licensing)

The following levels are defined (in priority order):

1. Default values (usually empty strings and 0 values)
2. Hardcoded basic settings (basic settings embedded with the executable)
3. Company specific default settings (embedded in the software based on your requirements sent to MizuTech)
4. Company configuration file   (can be created by the skin editor)
5. Default config file (can be downloaded from your web server)
6. User preferences (Customer settings)

This means that user preferences will always overwrite default company settings (but in the default company settings you can define values that cannot be overwritten. For example you can specify 60 sec for the registration interval and set it as read-only, thus it cannot be overwritten by higher level settings)

Default values

All config setting has a meaningful default value by default. These are usually empty strings and 0 for integer values. If there are no changes on upper levels, than you get a full featured softphone with no customization (the same with the default softphone that can be downloaded from the MizuTech website)

Hardcoded basic settings

For custom softphones there are some basic configuration settings that can be embedded in the source code. The following settings have been defined:

-Force one account (CUSTOM_COMPANY_FORCE)
0: not customized
1: no (the user can add other sip domains and use the softphone with multiple providers)
2: only one account (the user cannot add new accounts. The softphone can be used only with company voip server)

-Company name (CUSTOM_COMPANY_NAME_DEF):
Your company full name (Example: Best Voip LTD)
This is displayed on several location across the softphone (About box, account setting, etc)

-Company SIP domain or IP address (CUSTOM_COMPANY_DOMAIN_DEF)
This is the preconfigured sip server where the softphone will register and initiate calls. If you use not standard port, then it can be specified like domainname:port.

-Logo (CUSTOM_COMPANY_LOGO_DEF)
Company logo file name displayed on several place (Dial pad, About box, etc)
Usually a transparent bitmap (BMP) file. (or cur,pcx,gif,ani,jpg,bmp,ico,wmf,emf). Common size is: 180x80
(Will be stored in image subdirectory)

-Start Page (CUSTOM_COMPANY_STARTPAGE_DEF):
you can display a start page for the users
-**Start page URL** (webpage as the default page)
-"General Info": a default start page will be displayed with general info
-empty: no start page will be displayed (default)

-Main web URL (CUSTOM_COMPANY_WEBLINK_DEF)
Company website URL, displayed in several location (menu, about box, etc)

-Support page (CUSTOM_COMPANY_WEBLINK_SUPPORT_DEF)
Direct link to support page displayed on the help menu. Example: www.mycompany/support

-Credit recharge page (CUSTOM_COMPANY_WEBCREDIT_DEF)
Link to a page where the user can recharge its credit Example: www.mycompany/recharge
This can be displayed automatically when the user runs out of credit and/or can be put I the help menu

-Low credit value (CUSTOM_COMPANY_LOWCREDIT_DEF)
When the credit is below this limit, then it will appear in red with the link mentioned above

-Company email address (CUSTOM_COMPANY_MAIL_DEF)
For example: support@mycompany.com

-Default language (CUSTOM_COMPANY_DEFLANGUAGE_DEF)
Application (default) language (may need some time to translate if already not supported or we can send a text file in
English that has to be translated into your language)

-New user link (CUSTOM_COMPANY_NEWUSERLINK_DEF)
A direct link to your webpage where new accounts can be created
On the login form the "New User" link will open a browser window to this address. In this way users can register for your service and quickly create a new sip account

-Advertisement file (CUSTOM_COMPANY_AD_FILE_DEF)
Mizuphone can display the following types of files as advertisement:
  -(animated) image files using the image control: gif, jpg, bmp
  -video files using the local video player: avi, wmv, div, mpg, mpeg
  -any html content using the embedded internet explorer engine
-Advertisement link (CUSTOM_COMPANY_AD_LINK_DEF)
If the advertisement is not html, than this is the action what happens when the user click on the ad image (will open a browser and navigate to this site).
If the advertisement is html, than it has no meaning, because html can contain links.
-Credit check link (CUSTOM_COMPANY_CREDIT_CHECK_DEF)
-Link to call forward settings (CUSTOM_COMPANY_CALLFORWARDLINK_DEF)
-Link to callback settings (CUSTOM_COMPANY_CALLBACKLINK_DEF)
-STUN server address (CUSTOM_COMPANY_STUN_DEF)
-Profile storage location FTP (CUSTOM_STORAGE_FTP_DEF)
-Profile storage location http (CUSTOM_STORAGE_HTTP_DEF)
-Profile storage directory 1 (CUSTOM_STORAGE_DIR1_DEF)
-Profile storage directory 2 (CUSTOM_STORAGE_DIR2_DEF)
-Profile username (CUSTOM_STORAGE_USERNAME_DEF)
-Profile password (CUSTOM_STORAGE_PASSWORD_DEF)
-Allowed codec's (CUSTOM_COMPANY_CODECS_DEF)
List of the allowed codec's. For example: PPCMU,PPCMA,PG729,PG723, PH263, PH264
-DTMF mode (define CUSTOM_COMPANY_DTMF_DEF)
The following values have been defined:
        0: In-Band
        1: INFO
        2: In-Band and INFO
        3: RFC 2833
        4: RFC 2833 and INFO
        5: In-Band+RFC 2833+INFO
        6: RFC 2833 or INFO
-Brand name (CUSTOM_COMPANY_BRANDNAME_DEF)
The name of the softphone (instead of "MizuPhone" we can rename the software after your wishes)
This is the title and also the executable file will reflect this name.
For example: "SuperPhone"
-SMTP server address (CUSTOM_COMPANY_EMAIL_ADDR_DEF)
You can give your customers the opurtunity to send emails directly from the softphone via your email server. For this you just have to setup a common SMTP account that will be used for this purpose.
-SMTP username (CUSTOM_COMPANY_EMAIL_USER_DEF)
-SMTP password (CUSTOM_COMPANY_EMAIL_PWD_DEF)
-List of allowed server (ALLOWEDSERVERS)
A list of domain names and ip addresses where the softphone is allowed to connect.
This value cannot be modified by upper level settings. (Otherwise a separate activation key would be required for all copies)
-Encrypted peers (ENCRYPTEDPEERLIST)
List of ip address or domain names. With these peers (usually sip servers) the communication is always done encrypted
-Auto update (AUTOUPDATEURL)
Link to auto update file (zip file)
-Central config file (CUSTOM_COMPANY_CONFIG)
The softphone can download configurations from this location

Company specific default settings

We can modify the softphone based on your requirements. These are programming works done based on your request.
Example: protocol stack changes, GUI behavior changes, integration with your CRM, etc

Company configuration file

Usually the file exported by File menu -> Export current customized settings.

Default config file

The functionality of the softphone can be controlled by more than 200 configuration value.
These are usually technical settings.
Every aspect can be changed anytime later by placing a config file on the server which will be downloaded automatically by the clients.

User preferences

These are the settings changed by the users from the File Menu -> Settings panels. We can control the values that can be hided or disabled (not allowing the user to change the default value)

Configuration file hierarchy

By default the configuration is stored in a file named BRANDNAME.ini
This is an encrypted file (ini or xml format)
There are more than 200 settings that can be controlled with this configuration files (gui settings, sip stack configurations, etc)

Configuration files are checked in the following order:
1. in the install (bin) path (usually C:\Program Files\Mizu) –usually empty
2. in the data path (usually C:\Documents and Settings\WINUSER\Application Data\MizuPhone)
3. in the user profile path (usually C:\Documents and Settings\ WINUSER\Application Data\MizuPhone\usrprofiles\MIZUUSER)
4. defconfig.ini –this can be downloaded from your server and overwrite any user config. In this way you can easily force new configuration settings for your users.

The config value stored in the user profile will overwrite the data path config, which will also overwrite the install path config values.
This means that you can ship some default configuration with the install package (stored in the bin path), but some of them can be overwritten explicitly by the user (if you allow it on the GUI).

Configuration settings

Category	Setting	Default Value	Comment
Setting	Alertonlowdiskspace	5	alert on low disc space
settings	allownumbersendback	false	allow to route back the call to the caller
settings	autodetectlocalip	true	automatically overwrite the localip value if set to true
settings	bindip		bind to this ip (for multihomed servers or if we run multiple serers on the same maschine)
settings	boostonfirstcall	false	if to start with low priority and boost it when the first call arrives
settings	canaoutosaveinifiles	false	if we can save unsaved items
settings	cfg_block711	0	if g711 (PCMU,PCMA) calls are not allowed
settings	checkcputime	false	if cputime is constantly high, will restart
settings	checknumlen	99	if to check the incoming number len //14
settings	checkqueryrelease	false	check query object consistence on release
settings	connectondisccode	1	if to connect the sipcall before to play the disconnect reason. 0=not connect,1=connect only local users,2=connect all
settings	country		used in number normalizations
settings	countryprefix		used on routing
settings	cpuaffinity	0	0=dont change,1=cpu1,2=cpu2,3=cpu 1,other=number of cpu cores
settings	currency		local currency
settings	dailymainttaskhour	1	when to perform daily maintenance tasks
settings	dbloglevel	1	db server loglevel (0=only errors to monitor, 1=only all to monitor, 2=only errors to db, 3=no protocoll and events, 4= no protocoll and filtering, 5=no duplicates,6=log all)
settings	dbmaint_backuplevel	1	0=no backups,1=daily,2=daily,monthly,weekly,3=hourly,daily,monthly,weekly,4=full,5=keep lots of files
settings	dbtimeout	40	database query timeout
settings	deldbbackup	-1	delete old backup files after this day elapsed
settings	deleteoldlogfiles	14	delete older logfiles than the specified day (set to 0 to disable)
settings	emailfromaddr	sipserver@your compa	default email config
settings	emailfromname	sipserver	default email config
settings	emailhost	127.0.0.1	smtp server used for alerting
settings	emailsubject	SIP Notification	default email config
settings	emailuser	sipserver	smtp username used for alerting
settings	emergencydir	unknown	route emergency calls to this gateway (user id)
settings	enablefirewall	true	enable/disable builtin firewall and dos attack filtering
settings	enforcestrongauth	true	enforce authorization and strong passwords
settings	faxfromaddr	sipserver@your compa	fax sender configuration (email to fax server)

Category	Setting	Value	Description
settings	faxfromname	sipserver	fax sender configuration (email to fax server)
settings	faxhost	127.0.0.1	fax sender configuration (email to fax server)
settings	faxnormalize	0	fax sender configuration (email to fax server)
settings	faxsubject	SIPServer Notification	fax sender configuration (email to fax server)
settings	faxsuffix		fax sender configuration (email to fax server)
settings	faxuser	sipserver	fax sender configuration (email to fax server)
settings	fileloglevel	3	file server loglevel (0=only errors to monitor, 1=only all to monitor, 2=only errors to db, 3=no protocoll and events, 4= no protocoll and filtering, 5=no duplicates,6=log all)
settings	filetransferbufflen	10000	fileserver buffer length
settings	filetransfertick	500	fileserver speed
settings	gmtdiff		the difference to gmt (useful for sip date header)
settings	InternalIP		sipserver internal ip (interface to clinets)
settings	keepbackuprecorded	310	days to keep voice records in the backup directory
settings	keeprecorded	92	days to keep voice records
settings	LocalIP		sipserver external ip
settings	loglevel	3	other server loglevel (0=only errors to monitor, 1=only all to monitor, 2=only errors to db, 3=no protocoll and events, 4= no protocoll and filtering, 5=no duplicates,6=log all)
settings	lognofreecardc	0	list free card data when no route found
settings	logtodb	true	trace to database (log)
settings	logtofile	true	trace to file (log)
settings	maxloglisten	600	max log message quee length
settings	maxmemoryutilization	590000	max memory utilization in KB (restart if exceed)
settings	maxudpselect	-2	max socket on select (set to -2 to autoconfigure. -1 means no limits)
settings	minlogdelay	15	minimum delay between writing two log messages in msec
settings	minmemoryutilization	250000	will restart on offpeak if exceed
settings	normalizenumbers	3	0=not at all,1=medium,2=all but no endusers,3=check endusers too, 4=full
settings	ppriority	2	0=low,1=below,2=normal,3=abowe,4=high,5=realtime
settings	removetrailinghash	1	remove # when routing
settings	rotatelogfile	true	create separate logfiles for every day
settings	sendmailalert		if to send alerts on critial errors (pease configure the emailalertX settings)
settings	servername	SIPServer	server name (will appear in reports, alerts, etc)
settings	usedefaultdisccodes	false	don't use customized disconnect codes
settings	usedelayedupdate	true	sql upfates in separate thread
SIPSettings	ABSOLUTETIMEOUT	11100	max session time (call duration setup time clearing time) in seconds
SipSettings	addcontentdispozition	2	0=no
SIPSettings	allowcallunregistered		allow to call before registered (terminals)
SIPSettings	allowdiscmessage	true	allow disconect reason voice playback
SipSettings	allowlist	REGISTER,INVITE,CANC	sip methods
SIPSettings	CanAcceptLocalIp	false	Can call from 127.0.0.1
SipSettings	cancutsipnumbers	true	packet dialplan for sipnumbers
SipSettings	canmove	1	0=not allowed,1=callednumber change allowed,2=domain change allowed
SIPSettings	checknomedia	true	disconnect calls on rtp disconnect
SipSettings	COMPANYNAME		this will apear in the sip signaling
SipSettings	def_max_sessiontimer	3600	sip session-timer config
SipSettings	def_mid_sessiontimer	1800	sip session-timer config
SipSettings	def_min_sessiontimer	90	sip session-timer config
SipSettings	domainnames		registrar domainnames (used for inter-domain rerouting)
SipSettings	eventlist		refer,telephone-event,keep-alive
SipSettings	forwardretrytimer	8	ivr forward retry interval
SIPSettings	fwdtootherdomains	0	0=no,1=check,2=notelnumbers,3=all,4=reroute all,5 unconditioanal
SIPSettings	fwdunknownheaders	true	forward unknown sip headers
SIPSettings	HasInternalAccess	true	accept from 192.168... or 10.0...., etc
SIPSettings	identityrwmode	2	0=no rewrite, 1=basic, 2=conform sip specification (identity)
SIPSettings	IDLETIMEOUT	120	used for various session timers
SIPSettings	im_parentid	-1	used for instant messaging
SIPSettings	lastlocalsdpport	-1	used in terminals
SIPSettings	loadcallednumberfromto	true	load the called number from sip to instead from the sip first line (URI)
SIPSettings	localclientport		useful for 2 port configurations
SIPSettings	LocalDomain		sipserver domainname
SIPSettings	localinternaldomain		sipserver internal domainname
SIPSettings	LocalPort	5060	sipserver listen port
SIPSettings	logsipmsgexchange	true	store the sip message headers in the cdr comment
SIPSettings	MAINTIMERIVAL	2000	sip background process timer. used for garbage collections mainly
SIPSettings	MAXEPCOUNTTRESHOL	1000	maximum number of registered endpoint (it may be limited by license too)

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SIPSettings	MAXH323GKCDRCACHE	300	this must at least the maximum h323-h323 simultaneous call number
SipSettings	maxreroute	3	max number rute retry
SipSettings	MaxRTP	44000	rtp port range begin for sip
SipSettings	MAXSPEACHLEN	10800	max allowed call duration in sec
SipSettings	maxstatchangepermin	80	max allowed enduser status changes/60 sec (slower if exceed)
SIPSettings	MAXSUBSMSGCOUNT	12000	max subsequent messages before block
SIPSettings	MAXSUBSMSGPERIOD	180	max subsequent messages before block are checked for this interval (sec)
SIPSettings	MAXWRONGMSGALLOWED	6000	dos attack protection
SIPSettings	MEDIATIMEOUT	300	will disconnect if the media disappears
SIPSettings	MEDIATIMEOUTSTART	150	will disconnect if no media detected
SIPSettings	MINRESENDIVAL	500	sip udp resend timer (T1) in msec
SIPSettings	MinRTP	24001	rtp port range begin for sip
SIPSettings	MINSPEACHLENONROUTE	40	minimum remained speechlength for the caller when the router will still route the call
SIPSettings	MINUSERCREDITONROUTE	1	minimum credit for the caller when the router will still route the call
SIPSettings	MINUSERNAMELEN	3	minimum accepted username length
SIPSettings	PRODUCTNAME		the name of the product. this will apear in the sip signaling
SIPSettings	propersipcomments	false	set to true if you want personalized sip headers
SIPSettings	REBUILDREGCLIENTS	10800	usually the same as maxspeachlen
SIPSettings	REENABLEDOSBLOCKED	43200	reenable blocked endpoints after this interval. defaults to 12 hour
SIPSettings	REGISTERIVAL	40	upper registration interval in msec. defaults to 40 min
SIPSettings	RELOADPROXYLISTIVAL	7200	reload proxies from the config
SIPSettings	repeatdisc	false	if to send sip disconnect more than once
SIPSettings	REPOPUPULATEFDSETIVAL	13	used for rtp routing
SipSettings	REPOPUPULATEFDSETIVAL_MAIN	1300	used for main routing
SipSettings	resolvedns	true	resolve uri domain names
SIPSettings	ringtimeout	120	sip calls ring timeout
SIPSettings	routepresence	false	route subscribes
SIPSettings	rtpsendonlytorec	1	send the rtp only to the rec address
SIPSettings	rtpwritefirst	true	send a rtp packet after connect (to open NAT)
SIPSettings	sdpalthandling	1	0=not handled,1=check,2=delete,3=replace
SIPSettings	sendfakesms	0	if to send fake sms
SipSettings	sessiontimer	2	0: don't use,1: load from global config ,2: autodetect (and using the sesskeepalive interval from the global configuration),Other: use with the specified timeout (minutes)
SipSettings	sipmsgresendcount	3	resend sip message count
SipSettings	sipmsgresendival	1500	sip message resend timer
SipSettings	sipsendtodefport	true	try to send to port 5060 too
SipSettings	statussaveival	2	minutes. used when predective is active
SipSettings	supportlist	timer,privacy,replaces	sip supported
SIPSettings	traceep1		user id, callednumber or callerip. all messages related to this ep will be written in logfile. set to negative to disable (log)
SIPSettings	traceep2		user id, callednumber or callerip. user id. all messages related to this ep will be written in logfile. set to negative to disable (log)
SIPSettings	udpkeepalive	25	send keepalive messages
SIPSettings	udppriority	3	rtp thread priority: 1=normal, 2 = higher, 3 = highest
SIPSettings	upperexpire	31	register expire
SIPSettings	usedateheader	true	send date to user agents
SIPSettings	userofflinemin	360	enduser will be considered offline if no register or invite for this period

Other config values

[contacts]
contact0=we can set default contacts here
[dnscache]
c0=predefined dns cache values here
c1=
[epsettings_1]
accountname=bt_basic
aec=false
agc=false

```
audiodevicein=default
audiodeviceout=default
audiodevicering=default
autoaccept_all=false
autoaccept_fax=true
cc_password=encrypted
cc_username=9991234570
delayedack=0
denoise=true
dtmftype=1
enablephototransfer=true
faxmode=1
forkallowed=true
hidecli=false
isfax=1
keepalivenotifynotsupported=
last_rtpresendin=0
maxjittersize=300
mediaaddressmode=2
minjittersize=31
missingnotify_calls=false
missingnotify_emails=false
noanswer_timeout=30
onlyencryptedsessions=false
plc=true
presence=1200000
proxyport=-1
registerival=120
retrywithallcodec=true
rtpsendonlytorec=0
sendcomposingnotification=true
sendmail_missedcalls=false
sendmail_missedmsg=false
serverip=
serverport=-1
sessiontimer=0
signalingaddressmode=2
smsinsertname=true
startjittersize=41
startmediawithsessionp=0
transportprotocol=0
udpkeepalive=25
udpkeepalivetype=0
useaccount=false
useencryption=false
usefaxecm=false
usepublish=false
usesrtp=false
videobitrate=2
voicemail_address2=
voicemail_always=false
voicemail_busystatus=false
voicemail_dontdisturb=false
voicemail_forward=false
voicemail_incall=false
voicemail_noanswer=false
voicemail_reject=false
voicemail_subscribe=4
volume_in=50
volume_out=50
volume_ring=50
webcamdevide=default

[events]
ev1=we can put some default event in the history
ev2=
firsteventid=1
```

lasteventid=2
[inifilehandling]
inifilesavedt=2008/10/20 15:06:48
inifilesavetick=16696437
[lastcallist]
num0=some predefine phone numbers here
num1=
[privacy]
blockcalls=0
blockchats=0
blockvideo=0
newpeople=1
transfer=0
[settings]
advancedsettingclicked=true
allcallcount=55
allowmultipleinstance=false
allphonecallcount=28
allspeechlength=885
allsucccallcount=29
alluseminutes=993
alwaysontop=false
autodetectlocalip=true
autogain_out_default=0.773807942867279
autogain_out_sb live! audio [ac00]=0.356729373335838
autoqos=true
autostart=true
cc_password=
cc_recording2=1
cc_username=fenesiistvan
checkfornewversions=true
configtest=0
cpu_optimizations=true
defaultpage=0
dialidentity=false
dialvideo=false
displayfriendlydt=true
dnscache=3000
enableice=true
enablescripts=false
enablestun=false
enableupnp=false
faxdirectory=C:\Documents and Settings\root\Application Data\Mizu\usrprofiles\u\incomingfax
fileloglevel=0
formstate=normal
history_chat=3
history_events=3
history_video=0
history_voice=3
keeprecorded=31
language=English
lastlocalip=
lastlocaliplist=10.0.0.1
lastpage=Dial
lastsavedepacc=1
lastselaccount=2
lastusedaccount=2
loglevelex=0
logsipmsgexchange=1
logsqlcommands=1
logtofile=false
msgtype=0
mydetails=user details here
nearpeople_allow=1
needbigbuttons=true
needtoolbar=false
ondoubleclick=2

```
periodicbackup=3
personid=pi2064526267
profilestorage=0
profilestoragepwd=$ppassword
profilestorageurl=http://$pdomain/webdav/$pusername/
profilestorageusername=$pusername
scriptcallconnect=mscript_call_connect.exe -D inout -A caller -B called  -S account
scriptcalldisc=mscript_call_disc.exe -D inout -A caller -B called  -S account -D duration -R reason
scriptcallring=mscript_call_ring.exe -D inout -A caller -B called  -S account
scriptcallstart=mscript_call_start.exe -D inout -A caller -B called  -S account
scriptcontactpresence=mscript_presence.exe  -A contact -S account -P status
scriptdtmfrec=mscript_dtmf.exe  -A from  -S account -T dtmf
scriptfaxrec=mscript_fax.exe -A from  -S account -F file
scriptim=mscript_im.exe -D inout -A from -B to  -S account
scriptmystatus=mscript_status.exe -S status
scriptuserlogin=mscript_ulogin.exe -U username
scriptuserlogoff=mscript_ulogoff.exe -U username
sendemailvia=0
sendmessageaction=0
separatechatwindow=false
showadvancedsettings=true
showadverisments=true
showcontactgroups=true
showcontacts=true
showdialpad=true
showhistory=true
showofflinecontacts=true
sielncesupress=false
silentmode=false
sortby_name=true
sortby_online=false
sortby_provider=false
sortby_ussage=false
stundomain=
use_rport=true
useridletime=10
[sipsettings]
blockselfcall=true
canmove=2
lastlocaladdresslist=
lastlocaliplist=10.0.0.100,
lastlocalsdpport=10100
localport=10001
maxrtp=10200
minrtp=10100
stunmap0=19616:19616
stunmap1=23974:23974
stunmap2=23976:23976
stunmap3=23978:23978
stunmap4=23980:23980
usetcp=false
usetls=false
useudp=true
[statuslist]
0=New custom status message
1=-
2=Call Me
3=Available
4=Do Not Disturb
5=Invisible
6=Away
7=Offline
```

Resources

Mizu SoftPhone homepage: http://www.mizu-voip.com/Products/Softphone.aspx

For help, contact support@mizu-voip.com