

ZOIPER

softphone

API Documentation Version 2.04 (c)zoiper.com

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1. Introduction

Zoiper API interface is designed for use with programming languages, capable to connect to OLE Automation servers. Its purpose is to provide control over Zoiper functions and events, so it could be embedded in other applications.

The API provided with Zoiper consists of several enum types and two dual interfaces, which provides the functionality needed by the application for using Zoiper.

The Automation server is registered in the system by executing the Zoiper executable with the /regserver parameter and is unregistered with /unregserver.

2. Constants and enums

TLineState - this enum type is used to hold one of the line state constants and represents the current state of a given line. It is used with the information events and cannot be set to a line.

- ⇒ The **Line state** constants are as follows:
- lsDown there is no call on the line. Also indicates that a call has ended.
- lsDialing the line is dialing but still no meaningful state.
- **lsActive** the line is active i.e. has voice activity from the remote side (probably IVR or some server side ringing sounds), but the call is not answered.
- lsRinging there is an incoming call on the line.
- lsWaitForAnswer the outgoing call is ringing on the remote side.
- lsUp the call is on, both sides are hearing each other. This is the only call state that can be transferred and put on hold.
- lsResolvingPort resolving STUN port.

TRegistrationResult - this enum type holds information for a registration result, provided by the registration events

- \Rightarrow The **Registration result** constants are as follows:
- rrRejected registration is rejected by the server (unsuccessful).



- rrAknowledged registration is acknowledged by the server (successful).
- rrTimeout no reply from the server.

TCodecs - this enum type represents the codecs that can be used by the phone.

- \Rightarrow The **Codec** constants are as follows:
- cPCMU G.711 u-law codec
- **cGSM** GSM codec
- cG723 not used
- cDVI4_8K not used
- cDVI41_6K not used
- cLPC not used
- cPCMA G.711 a-law codec
- cG722 not used
- cL16_2CH not used
- cL16_1CH not used
- cQCELP not used
- cCN not used
- **cMPA** not used
- cG728 not used
- cDVI4_11k not used
- cDVI4_22k not used
- **cG729** not used
- cCELB not used
- cJPEG not used
- cNV not used
- **cH261** not used
- cMPV not used
- cMP2T not used
- cH263 not used
- cSpeex_Narrow SpeeX narrowband codec
- cSpeex_Wide SpeeX wideband codec
- cSpeex_Ultra SpeeX ultraband codec
- ciLBC_30 iLBC codec(30ms frame)
- ciLBC_20 iLBC codec(20ms frame)
- cTel_Event not used
- cG726 not used



TDTMFCodes - this enum type describes the DTMF tone codes used by the phone.

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⇒ DTMF codes are as follows:
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■ dtmfc0 - DTMF tone digit 0;
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- dtmfc1 DTMF tone digit 1;
- dtmfc2 DTMF tone digit 2;
- dtmfc3 DTMF tone digit 3;
- dtmfc4 DTMF tone digit 4;
- dtmfc5 DTMF tone digit 5;
- dtmfc6 DTMF tone digit 6;
- dtmfc7 DTMF tone digit 7;
- dtmfc8 DTMF tone digit 8;
- dtmfc9 DTMF tone digit 9;
- dtmfcAsterisk DTMF tone symbol asterisk(*);
- dtmfcPound DTMF tone symbol pound(#);
- dtmfcA DTMF tone letter A;
- dtmfcB DTMF tone letter B;
- dtmfcC DTMF tone letter C;
- dtmfcD DTMF tone letter D.

TAccountTech - this enum represents the technology used by an account.

- atSIP account uses SIP protocol technology;
- atIAX account uses IAX protocol technology.

TRegistrationState - this enum type describes the registration state of an account.

- rsNotRegistered the account is not registered;
- rsRegistering the account is in process of registering;
- rsRegistered the account is registered.

TTransportType - this enum represents the transport type used by a SIP technology account:

- ttUDP the default UDP transport;
- ttTCP TCP based SIP communication;
- ttTLS TLS/TCP based SIP communication.



TDTMF_Style - this enum represents the method of sending DTMF tones with SIP:

- dtmfs_RFC2833 (default)- for DTMF RFC-2833 method of outband DTMF sending using the RTP protocol;
- dtmfs_SIPINFO for DTMF SIP INFO method of outband DTMF sending using the SIP protocol. It is useful when the RTP traffic is not going through the proxy server or PBX.

TTransferRequestStyle - this enum represents the way Zoiper handles transfer requests (with SIP conversations only):

- trsAccept with each transfer attempt no pop-up window shall appear and all transfers will be accepted automatically;
- trsReject with each transfer attempt no pop-up window shall appear and all transfers will be rejected automatically.;
- trsAsk with each transfer attempt a pop-up window will be appearing, asking the user to confirm proceeding with the transfer;

3. IIdefiskAccount interface

The **IIdefiskAccount** interface represents a Zoiper account with all of its properties and the ability to register and unregister the account. This interface has no class factory so instances of this interface can be obtained only by methods and properties of the **IIdefiskAPI** interface.

- 1. The properties of IIdefiskAccount interface are as follows:
- AccountTech (Read Only) the technology used by an account.
- Name (Read Only) string the name of the account.
- Domain (Read/Write) string the domain used for registration and outbound proxy server. If a different outbound proxy server is specified in the
- OutboundProxy property and the UseOutboundProxy property is set, this property will be used for registrations only (SIP only).
- UseOutboundProxy (Read/Write) boolean indicates wheather to use the outbound proxy specified in the OutboundProxy property.(SIP only).



- OutboundProxy (Read/Write) string this property is used to specify a different outbound proxy for a SIP account.(SIP only).
- **Host** (Read/Write) string the host that an IAX account should connect to.(IAX only).
- Context (Read/Write) context used by an IAX account. (IAX only)
- Username (Read/Write) string the user name that is used by the account for registration and outgoing calls.
- Password (Read/Write) string the password that is used by the account for registration and outgoing calls.
- CallerID (Read/Write) string the caller name used by the account for outgoing calls.
- CallerNumber (Read/Write) string the caller number used by the account for outgoing calls.
- STUNHost (Read/Write) string STUN server host address used for port resolving.(SIP only).
- STUNPort (Read/Write) unsigned short STUN server port used for port resolving.(SIP only).
- STUNRefreshPeriod (Read/Write) long STUN server refresh period which is used to refresh the assigned port.(SIP only).
- UseCustomCodecs (Read/Write) boolean indicates that this account uses a custom set of codecs.
- CurrentCodecs (Read) array of TCodecs currently selected codecs for this account. If UseCustomCodecs property is not set to True, this property will contain the globally selected codecs.
- RegisterOnStartup (Read/Write) boolean if True, the account is registered when the Automation server starts.
- Registered (Read Only) boolean shows if the account is registered.



- RegistrationID (Read Only) integer registration unique identificator.
- NewMessages (Read Only) integer the amount of new voicemail messages for this account. 'Registered' should be True.
- OldMessages (Read Only) integer the amount of old voicemail messages for this account. 'Registered' should be True.
- DonotPlayRingBackTones (Read/Write) boolean toggles the automatic playing of ringback tones.
- VoiceMailCheckExtension (Read/Write) string specifies the extension that is used to check account voicemail.
- TransportType (Read/Write) TTransportType this property specifies the transport protocol used by the account. (SIP only)
- DTMFStyle (Read/Write) TDTMF_Style this property specifies the method of outband DTMF sending, used by the account. (SIP only)
- 2. The functions of **IIdefiskAccount** interface are as follows:
- RegisterAccount no return value registers the account to the host with the current user name and password.
- UnregisterAccount no return value unregisters the account and cancels the registration refresh.
- EnableCodec(Codec: TCodec; Priority: integer) no return value enables the codec given by Codec parameter with the given by the Priority parameter priority. The priority ranges from 1 to 65535. If UseCustomCodecs property is not set to true this function will not take effect.
- **DisableCodec**(Codec: TCodec) no return value disables the codec given by the Codec parameter.



4. IIdefiskAPI interface

The IIdefiskAPI interface represents the Zoiper softphone core properties, functions and events. The interface has a class factory and instances of this interface can be obtained from the client application directly. This is a multi-instance, single apartment automation server which means that many instances of the interface can be obtained and all will work on the same Zoiper instance. Zoiper doesn't work with two instances.

IIdefiskAPI provides an interface, which allows the client application to apply all the functions to the currently selected line and to switch between lines.

The interface is also used to select one account, which is used to dial without specifying an account. Usually this is calling from external applications like a browser or Microsoft Outlook.

- 1. The properties of IIdefiskAPI interface are as follows:
- NumberOfLines (Read Only) integer shows the number of lines supported by the Zoiper application.
- SelectedLine (Read/Write) integer read this property to see which line is currently selected. Write to this property to change the selected line.

 Indexing of the lines is zero based. Values above NumberOfLines 1 are not accepted.
- AccountNames (Read Only) array of strings this property contains an array of all the account names present in the softphone.
- MuteOnEarlyMedia (Read/Write) boolean boolean property used to mute early media sound stream.
- RingWhenTalking (Read/Write) boolean property indicating if Zoiper will make ringing sounds if another line is not in down state.
- RecordCalls (Read/Write) boolean this property toggles the recording of calls.
- RecordPath (Read/Write) string this property sets the general path used to record calls.



- InputDevNames (Read Only) array of strings this array provides the names of all available input devices.
- InputDevName (Read/Write) string read from this property to see which input device is currently used. Write to this property to select another input device.
- OutputDevNames (Read Only) array of strings this array provides the names of all available output devices.
- OutputDevName (Read/Write) string read from this property to see which output device is currently used. Write to this property to select another output device.
- RingDevNames (Read Only) array of strings this array provides the names of all available ringing devices.
- RingDevName (Read/Write) string read from this property to see which ringing device is currently used. Write to this property to select another ringing device.
- InputVolume (Read/Write) double this property controls the input volume. It ranges from 0 to 1.
- OutputVolume (Read/Write) double this property controls the output volume. It ranges from 0 to 1.
- MicBoost (Read/Write) boolean property used to toggle the microphone boost option.
- PCSpeakerRing (Read/Write) boolean property used to toggle the ringing on the PC speaker.
- SelectedAccount (Read/Write) string read from this property to see which account is currently selected. Write to this property to select another account.
- **DefaultAccountName** (Read/Write) string this property is used to specify the account that will be selected when the phone starts.



- Account[name: string] (Read Only) IIdefiskAccount this is an indexed property which will take the account name and will return an instance of the IIdefiskAcount interface to work with. The property will return NULL(Nothing) if this account does not exists.
- CurrentCodecs (Read) array of TCodecs currently selected codecs for this account. This property will contain the globally selected codecs.
- EchoCancellation (Read/Write) boolean this property controls the use of echo cancellation filter.
- CatchProtocolrequests (Read/Write) boolean this property enables and disables the ability of the phone to catch URL protcol(callto) requests.
- IntegrateIntoOutlook (Read/Write) boolean property used to control the phones integration into Microsoft Outlook. Microsoft Outlook must be restarted for this to take effect.
- MuteInput (Read/Write) boolean this property mutes and unmutes the input(microphone) for the selected line.
- MuteOutput (Read/Write) boolean this property mutes and unmutes the output(speakers) for the selected line.
- **DisableRingingSound** (Read/Write) boolean if this property is set to True it disables the ringing sounds in Zoiper
- **Version -** (Read only) string this property is used to determine the current Zoiper version.
- LibraryRevision (Read only) string this property is used to determine the current Zoiper library revision.
- IsInConference[Line : integer] (Read only) boolean this is an indexed property that is used to determine if the given Line is in conference.
- **Debug** (Read/Write) boolean this property toggles the library debugging file. The debugging text file is located in the folder where Zoiper is executed from and is called "debug.log".



- TransferRequestStyle (Read/Write) TransferRequestStyle this property defines the way that Zoiper handles incoming transfer requests.
- 2. The functions of **IIdefiskAPI** interface are as follows:
- Hangup no return value the function is used to hangup the selected line.
- HangupLine (line: integer) no return value the function is used to hangup the specified line by the 'line' parameter.
- Hold no return value this functions toggles hold/unhold on the selected line. The client should keep track of the state of the line. Selecting another line and than again the previous line will unhold the line.
- Transfer (extension : string) no return value this function makes blind transfer to the given extension. If there is no one to answer this extension, the call fails.
- EnableForwarding (extension : string; forwardafter: integer) no return value this function enables the call forwarding to the given extension after forwardafter seconds. If forwardafter is 0 then instant forwarding takes place.
- **DisableForwarding** no return value this function disables the call forwarding.
- AccountNew (Name: string; AccountTech : TAccountTech) IIdefiskAccount this function creates a new account using the specified by AccountTech parameter technology. In case of an account with the same name or the AccountTech is unknown, this function returns NULL(Nothing).
- AccountDelete (name: string) no return value this function deletes given account.
- Answer (line: integer) no return value this function answers the given line.
- Dial (extension: string; accountname: string) no return value this function dials the given extension on the selected line or on the first free line if the selected line is taken. The dialing is done using the account given or if an empty string is supplied; then it uses the current selected account to dial.



- SendDTMF (code: integer) no return value this function sends a DTMF sound to the selected line if it is not in state lsDown. The code has the following values:
 - (0 9) the digits from 0 to 9
 - (10) the asterisk digit
 - (11) the pound digit
 - (12 15) the letters form A to D
- EnableCodec (Codec: TCodec; Priority: integer) no return value enables the codec given by Codec parameter with the given by the Priority parameter priority. The priority ranges from 1 to 65535.
- **DisableCodec** (Codec: TCodec) no return value disables the codec, given by the Codec parameter.
- SaveOptions no return value this function save the current options. The options are saved automatically on server exit.
- AddToConference (line: integer) no return value this function adds the given from the parameter "line" line if the line itself is in lsActive or lsUp state.
- RemoveFromConference (line: integer) no return value this function removes line given from the parameter "line" from the conference.
- AttendedTransfer (firstline, secondline: integer) no return value this function attempts to make an attended transfer i.e. to connect directly the lines given by the "firstline" and "secondline" parameters. The first line is the leading line and OnTransferXXX events will report it state. Both lines have to be in state lsUp.
- 3. The events of IIdefiskAPI interface are as follows:
- OnRegistrationEvent (accountname: string; registrationresult:

 TRegistrationResult; newmessages: integer; oldmessages: integer) this event is fired when a registration result is received. It provides the account name and the registration result and if the registration result is rraknowledged, both the new and old voicemail messages count.



- OnUnregistrationEvent (accountname: string; registrationresult:

 TRegistrationResult) this event is fired when there is an unregistration result. It provides the account name and the registration result.
- OnLineStateEvent (line: integer; state: TLineState; accountname: string; format: TCodecs; callerid: string; releasecause: integer) the line state event is fired when a line changes its state and provides the line, the new state, the account used for the call, the format that is used for the call, the callerid and the release cause if the new state is lsDown.
- OnTextReceiveEvent (line: integer; text: string) this event is fired when a line receives a text message.
- OnURLReceiveEvent (line: integer; URL: string) this event is fired when a line receives a url.
- OnAudioLevelsEvent (inputlevel: integer; outputlevel: integer) this event is fired continuously to indicate the input and output levels.
- OnExternalDialEvent (extension : string) this event is fired when there is a dial request from a browser or command line. The extension provided by this event is not dialed automatically. This event is fired only when the phone is started in COM mode only i.e. it is invisible.

5. Release cause codes

- 1 unallocated
- 2 no route to transit net
- 3 no route to destination
- 6 channel unacceptable
- 7 call awarded delivered
- 16 normal clearing
- 17 user busy
- 18 no user response
- 19 no answer
- 21 call rejected
- 22 number changed
- 27 destination out of order
- 28 invalid number format



- 29 facility rejected
- 30 response to status enquiry
- 31 normal unspecified
- 34 normal circuit congestion
- 38 network out of order
- 41 normal temporary failure
- 42 switch congestion
- 43 access info discarded
- 44 requested channel unavailable
- 45 pre empted
- 47 resource unavailable
- 50 facility not subscribed
- 52 outgoing call barred
- 54 incoming call barred
- 57 bearercapability notauth
- 58 bearercapability notavail
- 63 service or option unavailable
- 65 bearercapability notimpl
- 66 channel not implemented
- 69 facility not implemented
- 79 service or option not implemented
- 81 invalid call reference
- 88 incompatible destination
- 95 invalid messsage unspecified
- 96 mandatory ie missing
- 97 message type nonexist
- 98 wrong message
- 99 ie nonexist
- 100 invalid ie contents
- 101 wrong call state
- 102 recovery on timer expire
- 103 mandatory ie length error
- 111 protocol error
- 127 interworking
- 200 local

