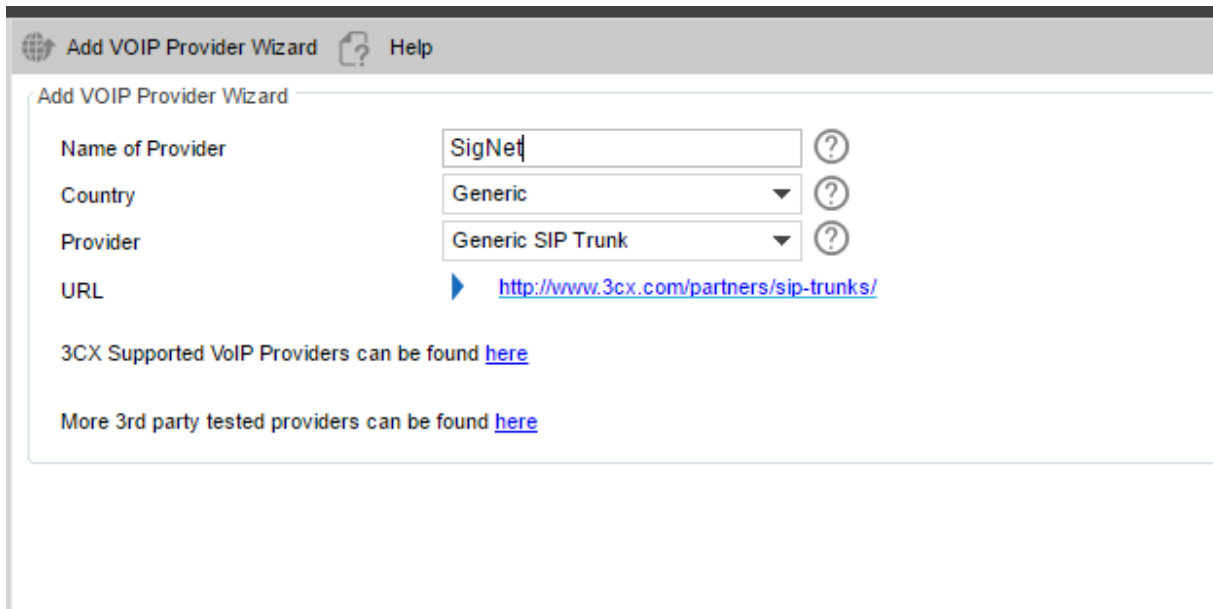


3cx omzetten SIP trunk Signet.

Start de management applicatie.

Maak een nieuwe Trunk aan. Kies bij country en provider voor 2x Generic



Add VOIP Provider Wizard Help

Add VOIP Provider Wizard

Name of Provider ?

Country ?

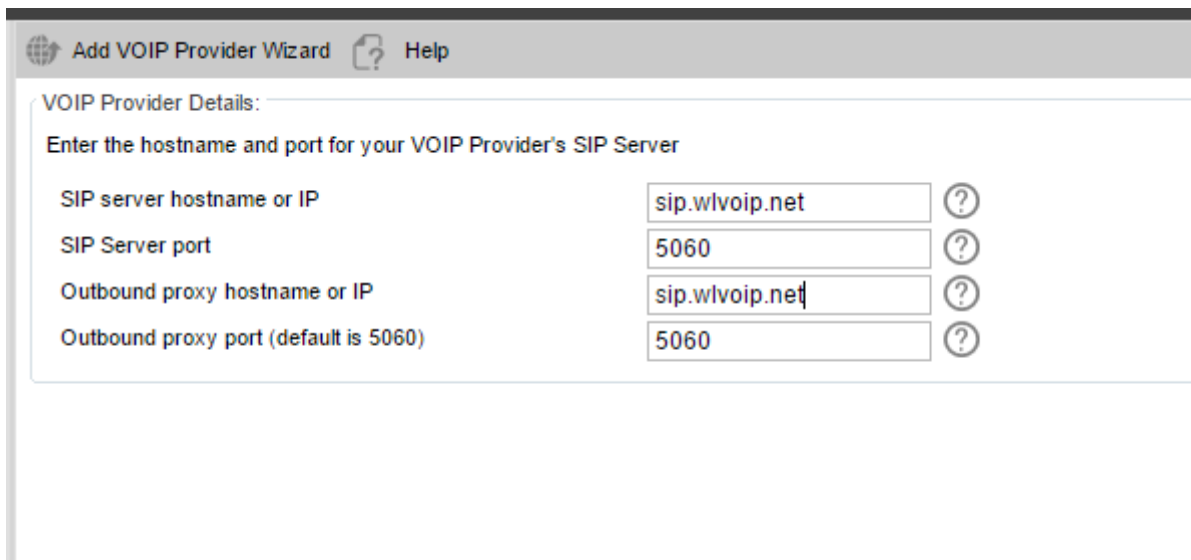
Provider ?

URL <http://www.3cx.com/partners/sip-trunks/>

3CX Supported VoIP Providers can be found [here](#)

More 3rd party tested providers can be found [here](#)

Kies Next



Add VOIP Provider Wizard Help

VOIP Provider Details:

Enter the hostname and port for your VOIP Provider's SIP Server

SIP server hostname or IP ?


SIP Server port ?

Outbound proxy hostname or IP ?

Outbound proxy port (default is 5060) ?





Voer 2 x sip.wlvoip.net in

En weer next


 Add VOIP Provider Wizard

Account Details

Enter the Authentication ID or SIP User, Password and number of your account


External Number	<input type="text" value="123456789"/>	
Authentication ID (aka SIP User ID)	<input type="text" value="123456789"/>	
Authentication Password	<input type="password"/>	
3 Way Authentication ID	<input type="checkbox"/> <input type="text"/>	

Simultaneous Calls

Maximum simultaneous calls	<input type="text" value="2"/>	
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Voer telefoonnummer in en aantal simulatneous calls.

Authenticatie gegevens kan je hier nu nog niet invullen. Dan weer Next

 Add VOIP Provider Wizard

Office Hours

Configure where calls should be routed during office hours.

<input type="radio"/> End Call		
<input checked="" type="radio"/> Connect to Extension	<input type="text" value="58 Algemeen/helpdesk"/>	
<input type="radio"/> Connect to Queue / Ring Group	<input type="text" value="80 Algemeen nummer"/>	
<input type="radio"/> Connect to Digital Receptionist	<input type="text"/>	
<input type="radio"/> Voicemail box for Extension	<input type="text" value="58 Algemeen/helpdesk"/>	
<input type="radio"/> Forward to Outside Number	<input type="text"/>	
<input type="radio"/> Send fax to email of extension	<input type="text" value="email of extension 88"/>	

Same as Out of Office hours

Kies routing.

Daarna wordt er een outbound rule gemaakt.

↑ Outbound Rule Settings ? Help

General

Rule Name ?

Apply this rule to these calls

Define to which outbound calls the rule must apply

Calls to numbers starting with prefix ?

Calls from extension(s) ?

Calls to Numbers with a length of ?

Select Extension Group ?

Calls from extension group ...

Make outbound calls on

Configure up to 3 routes for calls. The second and third route will be used as backup. For each route, digits can be stripped or added.

Route		Strip Digits	Prepend
Route 1	<input type="text" value="SigNet"/>	<input type="text" value="0"/>	<input type="text"/>
Route 2	<input type="text"/>	<input type="text" value="1"/>	<input type="text"/>
Route 3	<input type="text"/>	<input type="text" value="1"/>	<input type="text"/>

Kies Finish.

Klik dan op de aangemaakte SIP Trunk links in de navigatiebalk, zodat je de accountgegevens die je eerder niet kon invullen kan instellen.

🌐 SIP Trunk Settings ? Help

General | Advanced | Outbound Parameters | Inbound Parameters | Source ID | DID

Provider Details

Enter the hostname and port of your provider's SIP Server.

SIP server hostname or IP ?

SIP server port ?

Outbound proxy hostname or IP ?

Outbound proxy port (default is 5060) ?

Account Details

Enter the Authentication ID or SIP User, Password and number of your account

External Number ?

Authentication ID ?

Authentication Password ? ***

3 Way Authentication ID ?

Simultaneous Calls

Maximum Simultaneous Calls ?

En bij advanced nog "require registration" instellen






SIP Trunk Capabilities

Configure Advanced options

- Supports Re-Invite 
- Supports 'Replace' 
- PBX Delivers Audio 
- Switch on Secure RTP (SRTP) 
- Disable Video 
- Put Public IP in SIP VIA Header

Registration Settings

Configure Advanced options

- Time between registration attempts (in seconds) 
- Require registration for: 
- Which IP to use in 'Contact' and 'Connection' SIP fields
 - Use Default Settings 
 - Local IP Address (Windows Default Route) 
 - Use this IP Address 

Codec priorities

Specify which codecs to use and their priority

Available Codecs

- Speex
- iLBC
- G729
- G722

Assigned Codecs

- G.711 U-law
- G.711 A-law
- GSM-FR