

Innovaphone PBX (IP302) - How to interconnect with Innovaphone PBX (IP302)?

- 2N[®] VoiceBlue Next has these parameters:
 - IP address 192.168.22.42
 - Incoming port: 5060
 - Innovaphone PBX parameters: • IP address 192.168.22.227
 - Incoming port: 5060

2N[®] VoiceBlue Next settings

1. SIP trunk interconnection

For the settings of the trunk between the $2N^{\circledast}$ VoiceBlue Next and your PBX you need to configure SIP proxy (GSM \rightarrow IP) for GSM incoming calls. SIP proxy (IP \rightarrow GSM) is designed for secure communication just with traffic from your PBX. You can specify the IP address and from which SIP packets will be accepted. In case you leave there 0.0.0.0 it will be open for all traffic.



2. Configuration of the LCR (Least Cost Routing)

The GSM operator has e.g. in our country prefix 7 and 8 with a 9-digit number. The setting is below.





2N® TELEKOMUNIKACE a.s., http://wiki.2n.cz

Logout ()

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Gateway control

GSM groups assignment

Gateway configuration			
System parameters	Module:	Outgoing:	Incoming:
VolP parameters	0. module	1. Group 💌	1. Group 💌
GSM basic parameters	1. module	2. Group 💌	1. Group 💌

3. Configuration of GSM outgoing groups

You are able to set up different setting for each GSM group (CLIR, free minutes, virtual ring tone, roaming and others)

2N			Gateway
TELECOMMUNICATIONS		Gate	way Update Restan
Gateway control	GSM outgoing groups		
Gateway configuration	Service and a		
 System parameters 	1. GSM group 2. GSM group		
VolP parameters			
GSM basic parameters	General settings		
 GSM groups assignment 	Delay for CONNECT [s]:	Off 🗾	
 GSM outgoing groups 	Minimal ring duration to send "SMS at no answer" [s]:	Off 🗾	
GSM incoming groups	Delay for ALERTING [s]:	4 💌	
Prefixes	'Minute' parameter:	Count of minutes	*
LCR table	Day of deleting statistics in group (every month):	1	
CLIP Routing table	Generate virtual ring tone:	Z	
 Mobility Extension 	Call length counting:	Seconds	
Ethernet configuration			
Login configuration	Disconnect call		
Web configuration		_	
Report configuration	SIM limit exceeded:	-	
Configuration backup	No ALERTING before CONNECT:		
	Send CLIP from VoIP to GSM/UMTS		
Logout ()			

4. Incoming calls

For incoming calls you can define 2 groups with the different behavior and assign them to the GSM modules. The settings are similar to GSM groups assignment for outgoing calls.

GSM groups assignment				
Module:	Outgoing:		Incoming:	
0. module	1. Group	-	1. Group	
1. module	2. Group	-	1. Group	•

In GSM incoming groups you can define the behavior for each GSM incoming group. Choose the mode to Reject, Ignore, Accept incoming calls or Callback.





You can define the list of called numbers which will be automatically dialed after DTMF dialing timeout if the customer does not press any button within the specified time. From the configuration you can see 10 seconds for DTMF dialing and after that the call will be routed to the extension 100 to your PBX (if you set up SIP proxy (GSM->IP) in VoIP parameters).



Innovaphone PBX settings

1. You need to set up Domain and Address for the IP address of the 2N[®] VoiceBlue Next. Communication protocol is SIP with the Mode: "Gateway without registration).



http://192.168.22.227/RELAY0/mod_cmd.xml?cmd=xml4fs&id=GW1&xsl=relay_edt_voip.xsl Name 2N_VoiceBlue_Ne Disable Protocol SIP Mode Gateway without Registration Domain 192.168.22.42 Address(Proxy) 192.168.22.42 Address(Proxy) 192.168.22.42 Mask STUN Server Local Signaling Port 5060 Media Properties General Coder Preference G729A Framesize [ms] 30 Silence Compression Enable T.38 Enable SRTP No DTMF Detection Enable PCM Media-Relay Record to (URL) SIP Interop Tweaks Accept INVITE's from Anywhere (affects registered interfaces only) Enforce Sending Complete (affects incoming SIP calls only) No Inband Information on Error (affects incoming SIP calls only) From Header when Sending INVITE CQPN in user part of URL (affects registered interfaces only) Identity Header when Sending INVITE CQPN in user part of URL (affects registered interfaces only) Reliability of Provisional Responses Supported (affects outgoing SIP calls only) OK Cancel Apply Delete Help	🕙 GW1 2N_VoiceBlue_Ne - Mozilla Firefox	_ 🗆 ×
Name 2N_VoiceBlue_Ne Disable Image: Stress of the stress of t	http://192.168.22.227/RELAY0/mod_cmd.xml?cmd=xml-ifs&id=GW1&xsl=relay_edit_vojp.xsl	☆
Enforce Sending Complete (affects outgoing SIP calls only) No Inband Information on Error (affects incoming SIP calls only) From Header when Sending INVITE Fixed AOR	Interviting: Disable Protocol SIP Mode Gateway without Registration Domain 192.168.22.42 Address(Proxy) 192.168.22.42 How Image: Signaling Port Study Image: Signaling Port General Coder Preference G729A Framesize [ms] 30 Silence Compression Exclusive Local Network Coder G711A Framesize [ms] 30 Silence Compression Exclusive Enable T.38 Enable SRTP No DTMF Detection Enable PCM Media-Relay Record to (URL) SIP Interop Tweaks Accept INVITE's from Anywhere (affects registered interfaces only) Image: Site State Sta	
No inband information on Error (affects incoming SiP calls only) From Header when Sending INVITE Fixed AOR (affects registered interfaces only) Identity Header when Sending INVITE CGPN in user part of URI (affects registered interfaces only) Reliability of Provisional Responses Supported (affects outgoing SIP calls only) OK Cancel Apply Delete	Enforce Sending Complete (affects outgoing SIP calls only)	
Reliability of Provisional Responses Supported (affects outgoing SIP calls only) OK Cancel Apply Delete Help	From Header when Sending INVITE CGPN in user part of URI (affects registered interfaces only) Identity Header when Sending INVITE CGPN in user part of URI (affects registered interfaces only)	
OK Cancel Apply Delete Help	Reliability of Provisional Responses Supported 🗹 (affects outgoing SIP calls only)	
	OK Cancel Apply Delete Help	

2. Then you need to set up prefixes (routing table) which will be routed to the 2N® VoiceBlue Next. In the example there is defined prefix 7 only.



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http://192.168.22.227/RELAY0/mod_cm	d.xml?cmd=xml-map&rt=RT	1↦=0&xsl=relay_edit_r	oute.xsl	☆
Description		Disable 🗖		
GW1 2N_VoiceBlue_Ne TEL1 GW2 TEL2 GW2	7	→ 7	GW1_2N_VoiceBlue_Ne Cause(DISC)	
BRI1 GW3 TEST GW4 TONE GW5 TONE GW6 HTTP GW7 ECHO GW8 SIP1 GW9 SIP2 GW10 SIP3 GW11 SIP4 GW12	Add UUI Final Route Final Map No Reroute on wrong No Verify CGPN Interworking(QSIG,SIP) Rerouting as Deflection Routing on Diverting No Force enblock Add # Disable Echo Canceler Call Counter		max	
Hotovo				

3. In the picture below there is the setting for incoming calls to Innovaphone PBX. The call is directed to the phone 1.

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http://192.168.22.227/RELAY0/mod_cmc	l.xml?cmd=xml-map&rt=RT	2↦=0&xsl=relay_edit_ro	ute.xsl	☆
Image: Signal	fox I.xml?cmd=xml-map&rt=RT Add UUI Final Route Final Route Final Map No Reroute on wrong No Verify CGPN Interworking(QSIG,SIP) Rerouting as Deflection Routing on Diverting No Force enblock Add # Disable Echo Canceler Call Counter elete Help	2↦=0&xsl=relay_edit_rom	max	
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4. In the picture below there is the complete setting for the Innovaphone PBX routing.



onfiguration	General Interfac	es SIP GK <mark>Routes</mark> CDRO C	DR1 Calls	
Seneral				
Р	⊤→ From	То	Counter CGPN M	laps
THO	TEL1	$\Box \rightarrow \Box \to 7 \oplus 7$ GW1:2N_Voic	eBlue_Ne b →	
TH1	GW1:2N_Voice	Blue_Ne 🖆 🖶 → TEL1	\rightarrow	∨BN to InnovaPhone
DAP				
EL1				
EL2				
BRI1				
dministration				
BX				
ateway				
ownload				
Ipload				
iagnostics				
leset				

More product information:

2N[®] VoiceBlue Next (Official Website 2N)