



Avaya Solution & Interoperability Test Lab

Application Notes for configuring novaalert V10 from novalink with Avaya IP Office R11.1 - Issue 1.0

Abstract

These Application Notes describe the configuration steps for novaalert from novalink with Avaya IP Office R11.1. novaalert integrates with Avaya IP Office using SIP trunks.

Readers should pay attention to **Section 2**, in particular the scope of testing as outlined in **Section 2.1** as well as the observations noted in **Section 2.2**, to ensure that their own use cases are adequately covered by this scope and results.

Information in these Application Notes has been obtained through DevConnect compliance testing and additional technical discussions. Testing was conducted via the DevConnect Program at the Avaya Solution and Interoperability Test Lab.

1. Introduction

These Application Notes describe the configuration steps for novaalert from novalink to interoperate with Avaya IP Office R11.1. novaalert integrates with Avaya IP Office using SIP trunks connecting to the primary server.

The Avaya IP Office consists of an IP Office Server Edition running on a virtual platform as the primary server with an IP Office IP500 V2 running as an expansion cabinet. Both systems are linked by IP Office Line IP trunks that can enable voice networking across these trunks to form a multi-site network. Each system in the solution automatically learns each other's extension numbers and usernames. This allows calls between systems and support for a range of internal call features.

novaalert is an application which is used in a health care, hotel or industrial environment for alerting, messaging or information services. novaalert can react to external alarm stimuli which indicate the existence of an emergency situation by informing affected persons of the situation. Alarms can be triggered from various possible input sources including manual input via IoT Devices, Web browser, Smartphone Apps, Databases, E-Mails, serial interfaces, potential free contacts, http(s) GET&POST, XML, SNMP, OPC, SMS, IP, etc. "Direct" alarms can also be defined which allow alarms to be input and triggered via telephone calls. The alarm triggering described is restricted to those methods which involve interaction with Avaya IP Office.

Once an alarm has been triggered, the medium selected when the alarm was configured is used to deliver the alarm. Possible delivery interfaces include phone calls (including conferences), IoT Devices, XML, http(s) GET & POST, Smartphone App's, Desktop-Clients, E-Mail, Pager, SMS, Fax, Printers, etc. Multiple recipients can be configured for an alarm, thus possibly creating multiple simultaneous telephone calls. If an alarm needs to be positively acknowledged, and it is not, novaalert can escalate that situation to other recipients, groups and devices. These Application Notes focus on those delivery methods which involve interaction with Avaya IP Office. The triggering of alarms was restricted to those methods which involve interaction with IP Office, that being alarms in the form of announcements being sent from novaalert to endpoints on IP Office, also using these endpoints to call into novaalert and record announcement messages to be sent out to other IP Office endpoints.

Alarms which are triggered via Avaya IP Office can include pre-recorded or ad hoc voice messages or can generate voice messages via a text-to-speech mechanism. The calling party name can also be configured to contain a brief alarm message, so that this alarm message will appear in the caller list of intended recipients who are unable to answer an alarm call. Alarms can be sent to busy stations that are already on a call by using the Service Observe feature. If novaalert detects a busy signal, it then uses the Coaching Intrusion or Call Intrude feature on IP Office to break into that call and play the alarm message.

2. General Test Approach and Test Results

This section describes the compliance testing used to verify interoperability of novaalert with IP Office and covers the general test approach and the test results. Alarms were initiated from novaalert and sent to IP Office phone sets and hunt groups over SIP trunks. IP Office Server Edition Primary Server with an IP500 V2 Expansion was used for compliance testing. Various Avaya endpoints were registered to the Server Edition and the IP500V2, see **Section 4**, using all endpoints during compliance testing. The SIP trunk was connected between the Primary Server and novaalert with a dial-plan setup accordingly.

novaalert was manually configured using the web interface to send alert messages to endpoints on IP Office.

DevConnect Compliance Testing is conducted jointly by Avaya and DevConnect members. The jointly defined test plan focuses on exercising APIs and/or standards-based interfaces pertinent to the interoperability of the tested products and their functionalities. DevConnect Compliance Testing is not intended to substitute full product performance or feature testing performed by DevConnect members, nor is it to be construed as an endorsement by Avaya of the suitability or completeness of a DevConnect member's solution.

Avaya recommends our customers implement Avaya solutions using appropriate security and encryption capabilities enabled by our products. The testing referenced in these DevConnect Application Notes included the enablement of supported encryption capabilities in the Avaya products. Readers should consult the appropriate Avaya product documentation for further information regarding security and encryption capabilities supported by those Avaya products.

Support for these security and encryption capabilities in any non-Avaya solution component is the responsibility of each individual vendor. Readers should consult the appropriate vendor-supplied product documentation for more information regarding those products.

For the testing associated with these Application Notes, the interface between Avaya IP Office and novaalert did not include use of any specific encryption features as requested by novalink.

2.1. Interoperability Compliance Testing

The interoperability compliance testing evaluated the ability of novaalert to carry out a variety of alarming functions, in various conditions, to multiple types of endpoints according to the configuration made via the web interface. These included recording of alarms from SIP/H.323/Digital endpoints.

- Triggering of Alarms from novaalert GUI.
- Triggering of Alarms from Avaya endpoints.
- Triggering of Alarms from the PSTN.
- Delivery of voice recorded and TTS alarm to groups of SIP/H.323/Digital endpoints.
- Conference, with "Conference" ticked in the Alarm, the endpoints will be held by novaalert after the alarm message and put into a voice conference with all other voice targets/endpoints.

- Delivery of voice recorded and TTS alarm to SIP/H.323/Digital endpoints.
- Delivery of voice recorded and TTS alarm to Hunt Groups.
- Delivery of voice recorded and TTS alarm to groups of SIP/H.323/Digital endpoints.
- Conference, with “Conference” ticked in the Alarm, the endpoints will be held by novaalert after the alarm message and put into a voice conference with all other voice targets/endpoints.
- Verification of Alarm Display messages on each handset.
- Delivery of Alarms to the phone set speaker directly using Dial Paging.
- Following Call Forwarding to deliver alarms.
- DTMF PIN entry to demonstrate permission verification to trigger alarms.
- Intrusion of Alarms to busy extensions using the Call Intrude Short Code.
- Escalation, delivery of an alarm to another user such as a manager or perhaps a secretary if the initial user fails to answer the alarm.
- Serviceability testing.

Serviceability testing consisted of verifying the ability of novaalert to recover from simulated network interruption to both IP Office and novaalert.

2.2. Test Results

All functionality and serviceability test cases were completed successfully. The following issues and observations were noted during the compliance testing.

1. A Short Code for FNE was added in order to initiate the Call Intrude Short Code; this was done because using the Call Intrude Short Code directly by novaalert results in a forbidden so it must use the FNE for Mobile Call Control followed by the Call Intrude Short Code.
2. DTMF will only work using SIP INFO. See **Section 6.1** to view this specific setup.

2.3. Support

Technical support can be obtained for novaalert from the website <http://www.novalink.ch/en/> or from the following.

novalink GmbH
 Business tower
 Zuercherstrasse 310
 8500 Frauenfeld
 Switzerland
helpdesk@novalink.ch
 Phone: +41 52 762 66 77
 Fax: +41 52 762 66 99

3. Reference Configuration

The configuration in **Figure 1** is used to compliance test novalink novaalert with Avaya IP Office Server Edition and Avaya IP Office IP500 V2 Expansion. The connection between the novaalert and the IP Office solution uses SIP trunks.

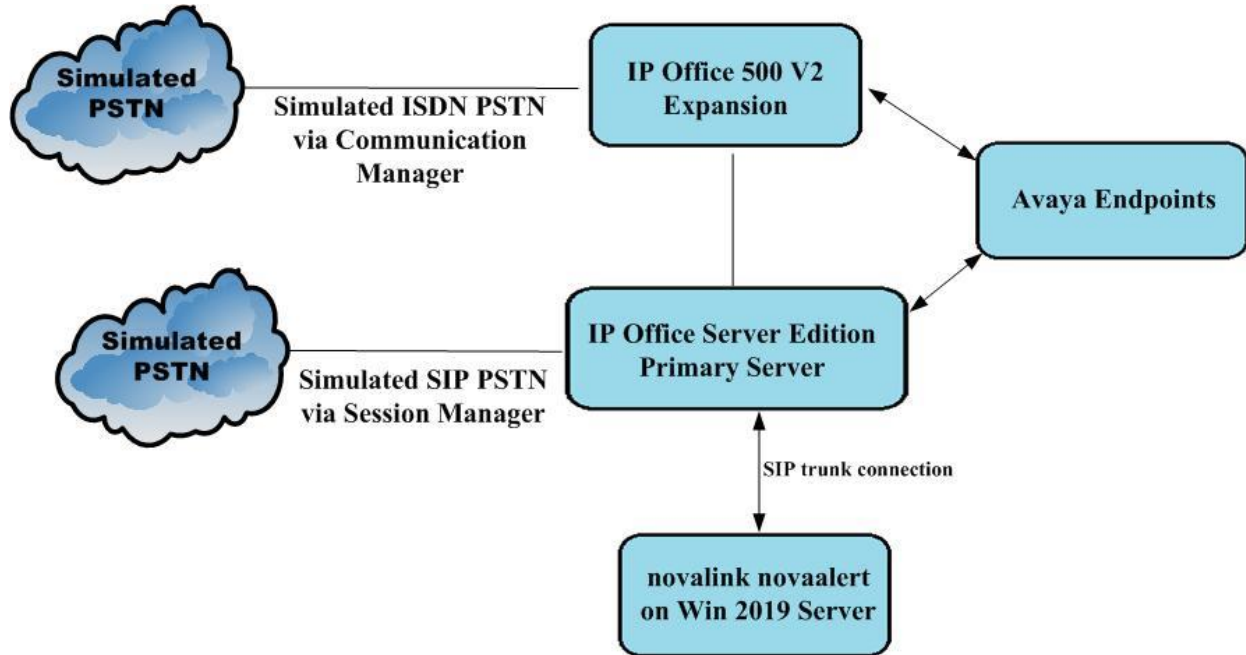


Figure 1: Connection of novaalert from novalink with Avaya IP Office Server Edition & Expansion

4. Equipment and Software Validated

The following equipment and software were used for the sample configuration provided:

Equipment/Software	Release/Version
Avaya IP Office Server Edition Primary Server running on a Virtual Platform	11.1.2.2.0 Build 20
Avaya IP Office 500 V2 Expansion	11.1.2.2.0 Build 20
Avaya J179 IP Phone (H.323)	6.8304
Avaya J159 IP Phone (SIP)	4.0.7.0.7
Avaya 9508 Digital Deskphone	R0.60
Avaya Workplace for Windows (SIP)	R3.22.0.64(SIP)
novalink novaalert running on a Windows 2019 virtual server	10.5.0.9

Compliance Testing is applicable when the tested solution is deployed with a standalone IP Office 500 V2 and also when deployed with IP Office Server Edition in all configurations.

Testing was performed with IP Office Server Edition R11.1. Note that IP Office Server Edition requires an Expansion IP500 V2 R11.1 to support analog or digital endpoints.

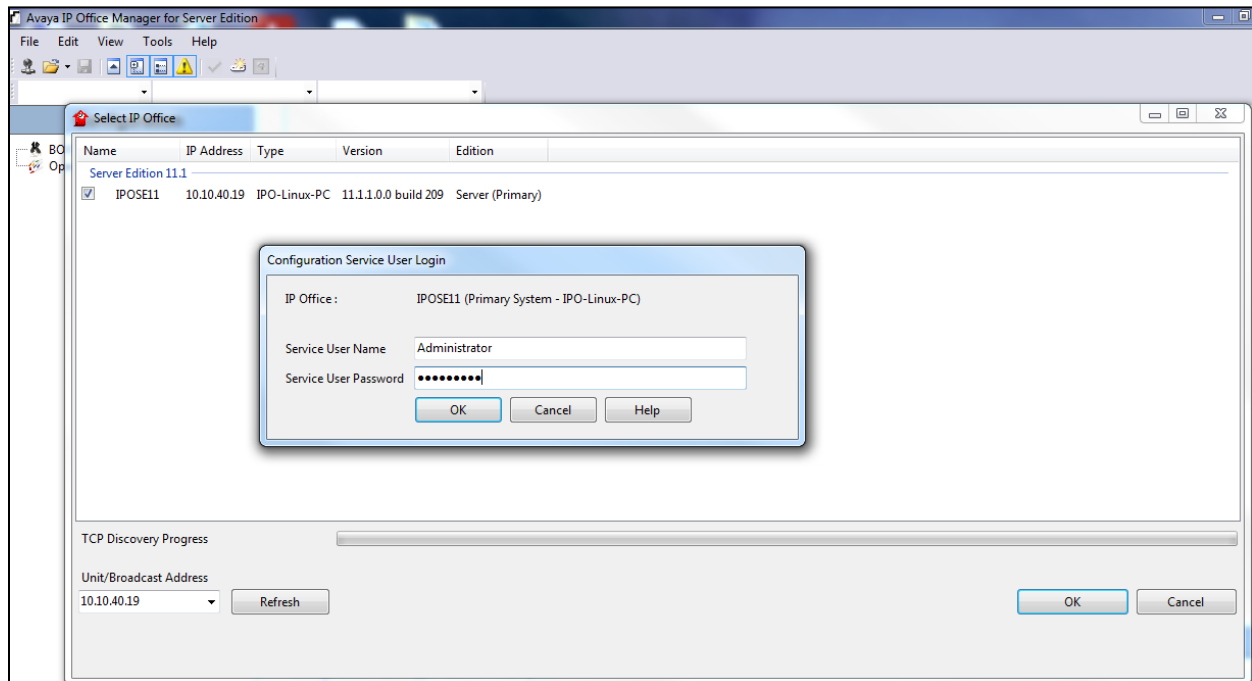
5. Configure Avaya IP Office

Configuration and verification operations on Avaya IP Office illustrated in this section were all performed using Avaya IP Office Manager. The information provided in this section describes the configuration of IP Office for this solution. It is implied a working system is already in place. For all other provisioning information such as initial installation and configuration, please refer to the product documentation in **Section 9**. The configuration operations described in this section can be summarized as follows:

- Launch IP Office Manager.
- Display LAN Configuration.
- Configure Incoming Route for SIP Trunk.
- Configure SIP Trunk.
- Configure User for Mobile Call Control.
- Configure Short Codes.
- Save Configuration.

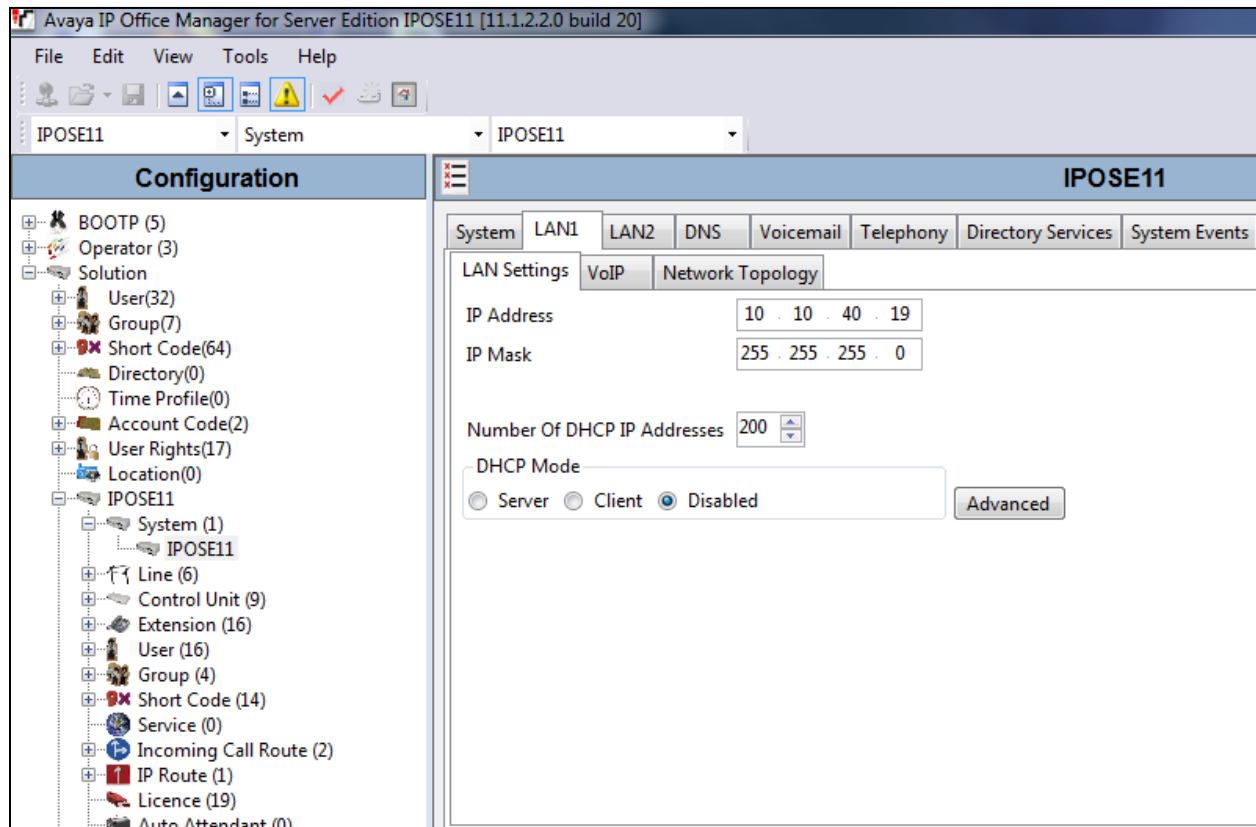
5.1. Launch IP Office Manager

From the IP Office Manager PC, go to **Start → Programs → IP Office → Manager** to launch the Manager application (not shown). Tick the required server to log in to, this should be the **Primary Server (Server Edition)** and log in to IP Office using the appropriate credentials to receive its configuration.



5.2. Display LAN Configuration

In the IP Office window expand the configuration tree in the left pane and double-click **System** (this may have a different name depending on the site). Select the **LAN Settings** tab within the **LAN1** tab and note the **IP Address** of the IP Office that will be required in **Section 6.1** for the configuration of the SIP Trunk on novaalert.



Click on the **VoIP** tab and ensure that the following are set correctly.

1. **SIP Trunks Enable.**
2. **SIP Registrar Enable.**
3. **SIP Domain Name**, set this to the telephony domain name.
4. **UDP** set the UDP Port to **5060**.
5. **TCP** set the TCP Port to **5060**.

Note: novaalert uses UDP to connect to IP Office.

The screenshot shows the 'IPOSEPG' configuration window with the 'VoIP' tab selected. The 'LAN Settings' sub-tab is active. The 'H323 Gatekeeper Enable' checkbox is checked, and 'H323 Remote Extn Enable' is unchecked. 'H.323 Signalling over TLS' is set to 'Disabled', and 'Remote Call Signalling Port' is 1720. Under 'SIP Trunks Enable', 'SIP Registrar Enable' is checked, and 'SIP Remote Extn Enable' is unchecked. The 'SIP Domain Name' is 'devconnect.local'. The 'SIP Registrar FQDN' is empty. For 'Layer 4 Protocol', 'UDP' is checked with port 5060, 'TCP' is checked with port 5060, and 'TLS' is checked with port 5061. Remote ports are also set to 5060 for UDP and TCP, and 5061 for TLS. 'Challenge Expiry Time (secs)' is 10. The 'RTP' section shows 'Port Number Range' from 40750 to 50750, and 'Port Number Range (NAT)' from 40750 to 50750. 'Enable RTCP Monitoring on Port 5005' is checked, and the 'RTCP collector IP address for phones' is 0.0.0.0. 'Keepalives' are enabled with a scope of 'RTP-RTCP' and a periodic timeout of 30.

IPOSEPG

System LAN1 LAN2 DNS Voicemail Telephony Directory Services System Events SMTP SMDR **VoIP** VoIP Security Contact Center

LAN Settings **VoIP** Network Topology

☒ H323 Gatekeeper Enable
☐ Auto-create Extn ☐ Auto-create User ☐ H323 Remote Extn Enable
H.323 Signalling over TLS Disabled Remote Call Signalling Port 1720

☒ SIP Trunks Enable
☒ SIP Registrar Enable
☐ Auto-create Extn/User ☐ SIP Remote Extn Enable
SIP Domain Name devconnect.local
SIP Registrar FQDN
☒ UDP UDP Port 5060 Remote UDP Port 5060
☒ TCP TCP Port 5060 Remote TCP Port 5060
☒ TLS TLS Port 5061 Remote TLS Port 5061
Challenge Expiry Time (secs) 10

RTP
Port Number Range
Minimum 40750 Maximum 50750
Port Number Range (NAT)
Minimum 40750 Maximum 50750
☒ Enable RTCP Monitoring on Port 5005
RTCP collector IP address for phones 0 . 0 . 0 . 0
Keepalives
Scope RTP-RTCP Periodic timeout 30
Initial keepalives Enabled

Click on the **Telephony** tab. Ensure that **Telephony** settings are correct for that particular setup. Below is just an example of what was used during compliance testing.

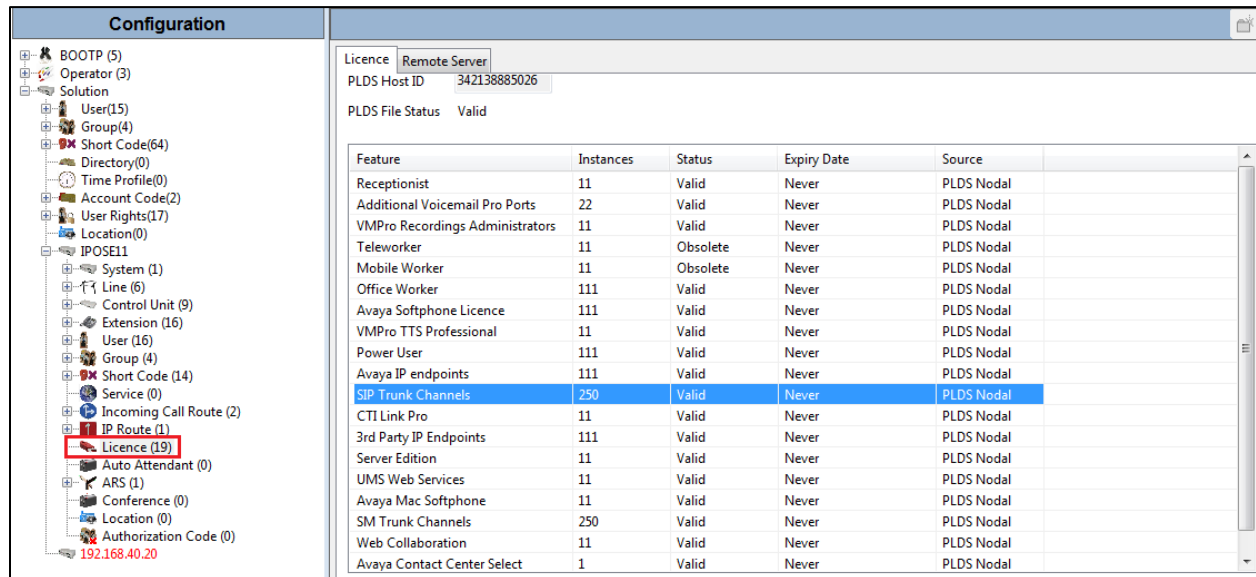
The screenshot shows the IPOSEPG web interface with the 'Telephony' tab selected. The interface includes a top navigation bar with tabs for System, LAN1, LAN2, DNS, Voicemail, Telephony, Directory Services, System Events, SMTP, SMDR, VoIP, VoIP Security, and Contact Center. The 'Telephony' tab is highlighted. Below the navigation bar, there are sub-tabs for Telephony, Park & Page, Tones & Music, Ring Tones, SM, Call Log, and TUI. The main content area is divided into two columns. The left column contains settings for Dial Delay Time (secs) set to 8, Dial Delay Count set to 2, Default No Answer Time (secs) set to 30, Hold Timeout (secs) set to 120, Park Timeout (secs) set to 300, Ring Delay (secs) set to 10, Call Priority Promotion Time (secs) set to Disabled, Default Currency set to EUR, Default Name Priority set to Favour Trunk, Media Connection Preservation set to Enabled, Phone Failback set to Automatic, Login Code Complexity with Enforcement checked and Minimum length set to 4, and RTP Collector Configuration with Send RTPC to an RTPC Collector checked, Server Address set to 0.0.0.0, UDP Port Number set to 5005, and RTPC reporting interval (secs) set to 5. The right column contains the Companding Law section with Switch set to A-Law and Line set to A-Law Line, DSS Status unchecked, Auto Hold checked, Dial By Name checked, Show Account Code checked, Inhibit Off-Switch Forward/Transfer unchecked, Restrict Network Interconnect unchecked, Include location specific information unchecked, Drop External Only Impromptu Conference checked, Visually Differentiate External Call unchecked, High Quality Conferencing checked, Directory Overrides Barring checked, Advertise Callee State To Internal Callers unchecked, and Internal Ring on Transfer unchecked.

Click on the **VoIP** tab. Ensure that the correct codecs are selected. Again, below servers to show what was used during compliance testing.

The screenshot shows the IPOSEPG web interface with the 'VoIP' tab selected. The interface includes a top navigation bar with tabs for System, LAN1, LAN2, DNS, Voicemail, Telephony, Directory Services, System Events, SMTP, SMDR, VoIP, VoIP Security, and Contact Center. The 'VoIP' tab is highlighted. Below the navigation bar, there are sub-tabs for System, LAN1, LAN2, DNS, Voicemail, Telephony, Directory Services, System Events, SMTP, SMDR, VoIP, VoIP Security, and Contact Center. The main content area contains settings for Ignore DTMF Mismatch For Phones checked, Allow Direct Media Within NAT Location unchecked, RFC2833 Default Payload set to 101, and a section for Available Codecs with G.711 ULAW 64K, G.711 ALAW 64K, G.722 64K, and G.729(a) 8K CS-ACELP all checked. The Default Codec Selection section shows an Unused list and a Selected list. The Selected list contains G.711 ALAW 64K, G.711 ULAW 64K, G.722 64K, and G.729(a) 8K CS-ACELP.

5.3. Display License Information

To ensure that there are enough licenses for all that is required, click on **License** in the left window and observe the licenses shown in the main window. **SIP Trunk Channels** is of significance here as the alarms are sent over a SIP trunk to IP Office.

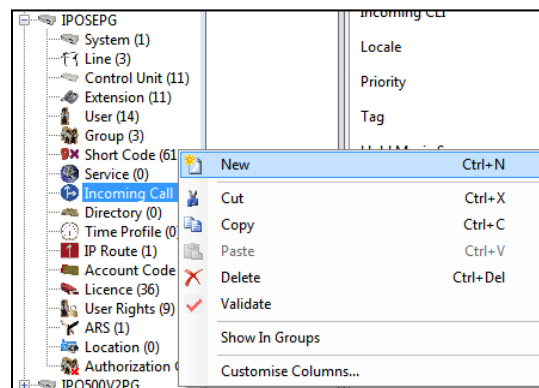


The screenshot shows the Avaya IP Office Configuration window. The left pane displays the Configuration tree with 'License (19)' selected. The right pane shows the License tab with a table of licenses. The table has columns: Feature, Instances, Status, Expiry Date, and Source. The 'SIP Trunk Channels' license is highlighted in blue.

Feature	Instances	Status	Expiry Date	Source
Receptionist	11	Valid	Never	PLDS Nodal
Additional Voicemail Pro Ports	22	Valid	Never	PLDS Nodal
VMPro Recordings Administrators	11	Valid	Never	PLDS Nodal
Teleworker	11	Obsolete	Never	PLDS Nodal
Mobile Worker	11	Obsolete	Never	PLDS Nodal
Office Worker	111	Valid	Never	PLDS Nodal
Avaya Softphone Licence	111	Valid	Never	PLDS Nodal
VMPro TTS Professional	11	Valid	Never	PLDS Nodal
Power User	111	Valid	Never	PLDS Nodal
Avaya IP endpoints	111	Valid	Never	PLDS Nodal
SIP Trunk Channels	250	Valid	Never	PLDS Nodal
CTI Link Pro	11	Valid	Never	PLDS Nodal
3rd Party IP Endpoints	111	Valid	Never	PLDS Nodal
Server Edition	11	Valid	Never	PLDS Nodal
UMS Web Services	11	Valid	Never	PLDS Nodal
Avaya Mac Softphone	11	Valid	Never	PLDS Nodal
SM Trunk Channels	250	Valid	Never	PLDS Nodal
Web Collaboration	11	Valid	Never	PLDS Nodal
Avaya Contact Center Select	1	Valid	Never	PLDS Nodal

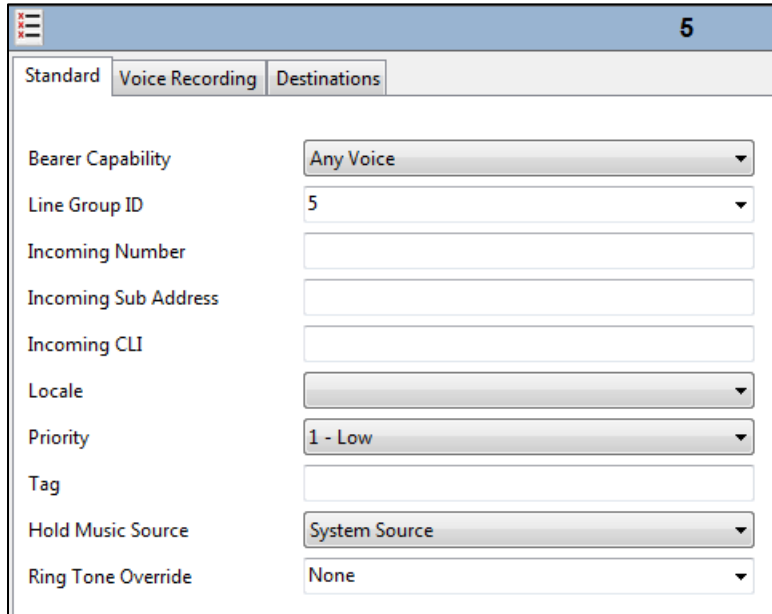
5.4. Configure Incoming Route for SIP Trunk

An incoming route must be added for the SIP trunk that will be setup in **Section 5.5**. Navigate to **Primary Server → Incoming Call Route**. Right click on **Incoming Call Route** select **New**.



The screenshot shows the Avaya IP Office Configuration window. The left pane displays the Configuration tree with 'Incoming Call Route' selected. A right-click context menu is open over 'Incoming Call Route' with 'New' selected.

From the **Standard** tab, enter an available **Line Group ID**; this can be kept the same as the SIP Line that is to be created for convenience. **Bearer Capability** can be set to **Any Voice**.



Standard Voice Recording Destinations

Bearer Capability: Any Voice

Line Group ID: 5

Incoming Number:

Incoming Sub Address:

Incoming CLI:

Locale:

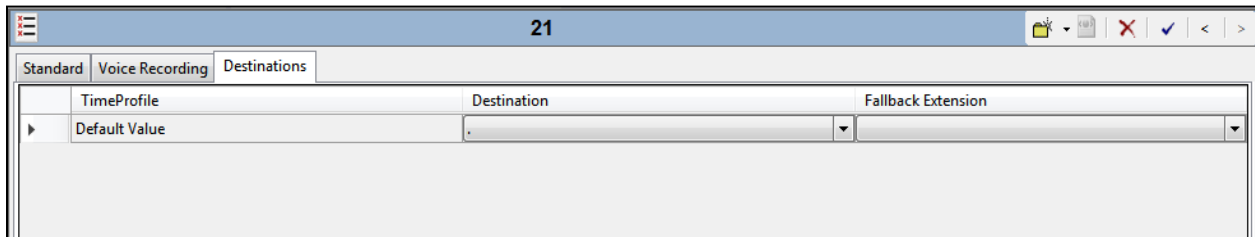
Priority: 1 - Low

Tag:

Hold Music Source: System Source

Ring Tone Override: None

From the **Destinations** tab, select . for the **Destination**. Click on **OK** at the bottom of the screen (not shown).



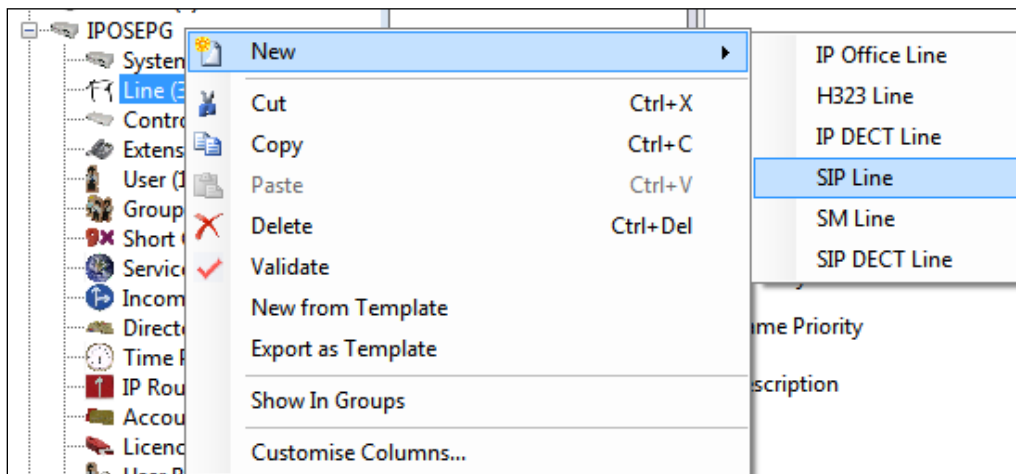
Standard Voice Recording Destinations

TimeProfile	Destination	Fallback Extension
Default Value	.	

5.5. Configure SIP Trunk

This section shows how to add a new SIP Trunk in order to facilitate the connection to novaalert. Navigate to the Server Edition or the IP Office module that novaalert is connecting to. During compliance testing novaalert connected to the IP Office Server Edition using SIP trunks, the SIP Line was therefore created on the Server Edition.

Navigate to **Primary Server** → **Line**, then right click on **Line** and select **New** → **SIP Line**.



Click the **SIP Line** tab and select the new **Line Number** and insert the IP Address of the novaalert server for the **ITSP Domain Name**.

A screenshot of the 'SIP Line - Line 5*' configuration window. The window has a title bar and a tabbed interface with the following tabs: SIP Line, Transport, Call Details, VoIP, SIP Credentials, SIP Advanced, and Engineering. The 'SIP Line' tab is active. The configuration fields are as follows:
Line Number: 5
ITSP Domain Name: 10.10.40.120
Local Domain Name: devconnect.local
URI Type: SIP URI (dropdown)
Location: Cloud (dropdown)
Prefix: (empty field)
National Prefix: 0
International Prefix: 00
Country Code: (empty field)
Name Priority: System Default (dropdown)
Description: (empty field)
In Service: ☒
Check OOS: ☒
Session Timers:
Refresh Method: Auto (dropdown)
Timer (seconds): On Demand (dropdown)
Redirect and Transfer:
Incoming Supervised REFER: Auto (dropdown)
Outgoing Supervised REFER: Auto (dropdown)
Send 302 Moved Temporarily: ☐
Outgoing Blind REFER: ☐

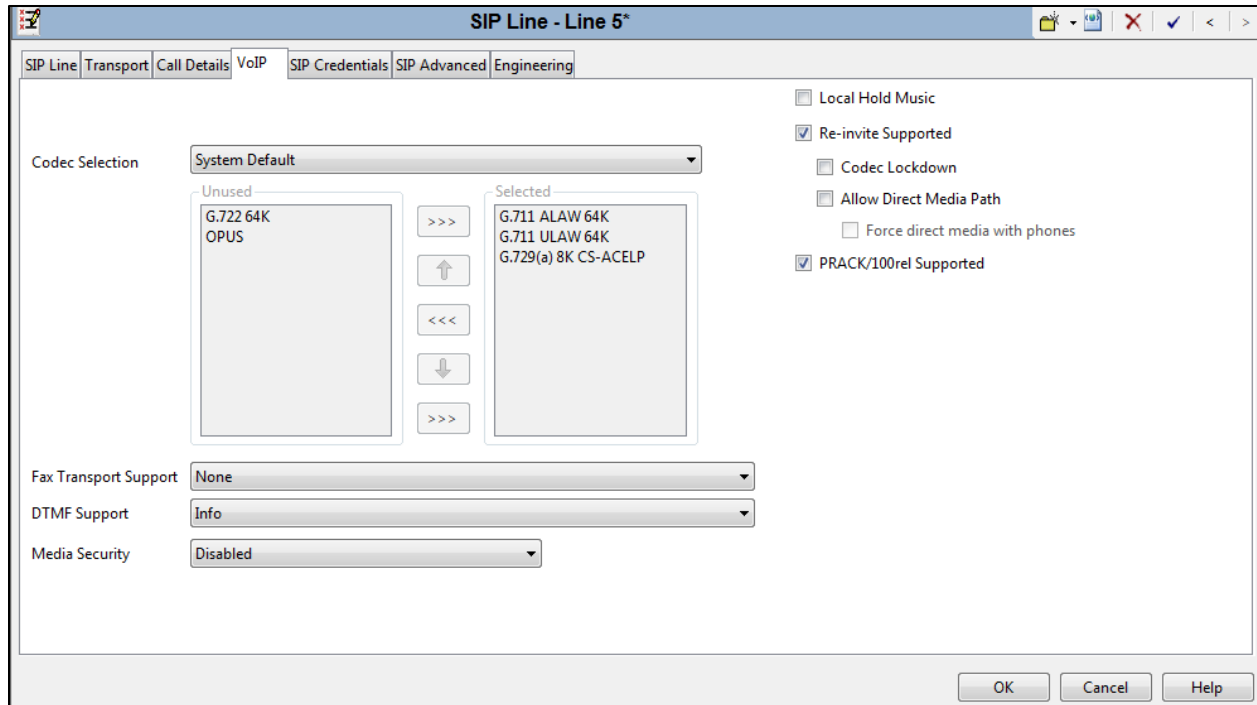
Click on the **Transport** tab and enter the IP Address of the novaalert server for **ITPS Proxy Address**. Ensure that the **Layer 4 Protocol** is set to **UDP** and that the **Send Port** and **Listen Port** are both set to **5060**.

The screenshot shows the 'SIP Line - Line 5*' configuration window with the 'Transport' tab selected. The 'ITSP Proxy Address' is set to '10.10.40.120'. Under 'Network Configuration', 'Layer 4 Protocol' is set to 'UDP', 'Send Port' is '5060', 'Use Network Topology Info' is 'None', and 'Listen Port' is '5060'. 'Explicit DNS Server(s)' are set to '0 . 0 . 0 . 0 . 0'. 'Calls Route via Registrar' is unchecked. 'Separate Registrar' is empty.

Click on the **Call Details** tab and click on **Add**. The **Incoming Group** and **Outgoing Group** are added here. **Max Sessions** was set to **10** for compliance testing, this number will depend on the number of SIP Licenses on IP Office and novaalert. Other settings were left as default, as shown below.

The screenshot shows the 'SIP Line - Line 5*' configuration window with the 'Call Details' tab selected. A table of SIP URIs is visible with one entry: URI 1, Groups 5 98890, Credential 0: <None>, Local URI Auto, Contact Auto. Below this, a 'New URI' dialog is open. In the dialog, 'Incoming Group' is set to '5', 'Outgoing Group' is '98890', and 'Max Sessions' is '10'. The 'Field meaning' section shows settings for Outgoing Calls, Forwarding/Twinning, and Incoming Calls, all set to 'None' or 'Original Caller'.

Select the **VoIP** tab and ensure that the correct **Codecs** are **Selected**. The **Re-invite Supported** and **Prack/100rel Supported** boxes are also ticked. **DTMF Support** must be set to **Info** in order to support the DTMF on novaalert which will be setup to use SIP INFO. Everything else can be left as default or as is shown below.



Under the **SIP Advanced** Tab, ensure that **Caller ID from From header** and **Send From In Clear** are both ticked. Click on **OK** at bottom of screen and that will complete the **SIP Line** setup.

The screenshot shows the 'SIP Line - Line 5*' configuration window with the 'SIP Advanced' tab selected. The window is divided into several sections:

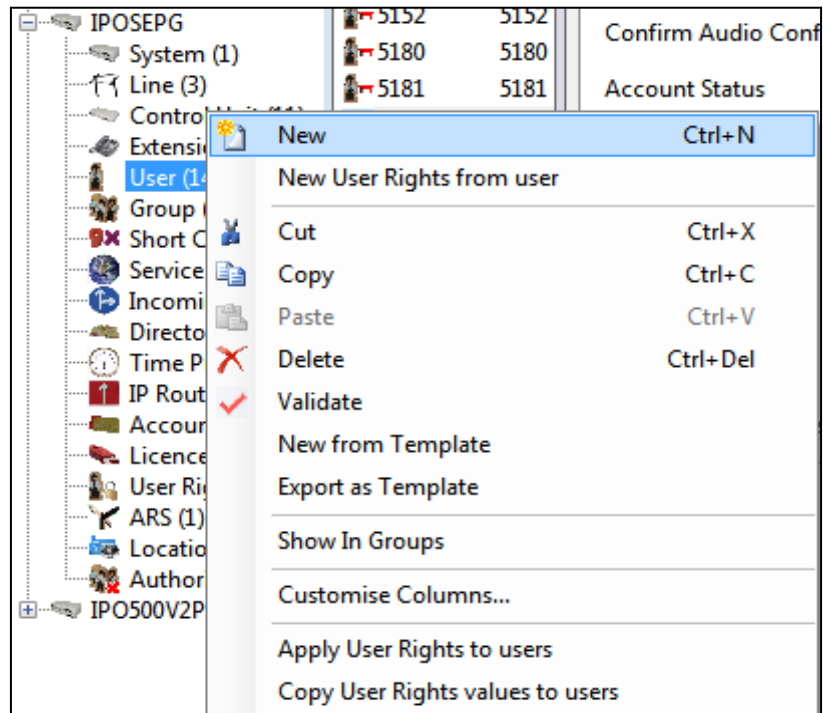
- Addressing:**
 - Association Method: By Source IP address
 - Call Routing Method: Request URI
 - Use P-Called-Party: ☐
 - Suppress DNS SRV Lookups: ☐
- Identity:**
 - Use "phone-context": ☐
 - Add user=phone: ☐
 - Use + for International: ☐
 - Use PAI for Privacy: ☐
 - Use Domain for PAI: ☐
 - Caller ID from From header: ☒
 - Send From In Clear: ☒
 - Cache Auth Credentials: ☒
 - User-Agent and Server Headers:
 - Send Location Info: Never
 - Add UUI header: ☐
 - Add UUI header to: ☐
- Media:**
 - Allow Empty INVITE: ☐
 - Send Empty re-INVITE: ☐
 - Allow To Tag Change: ☐
 - P-Early-Media Support: None
 - Send SilenceSupp=Off: ☐
 - Force Early Direct Media: ☐
 - Media Connection Preservation: Disabled
 - Indicate HOLD: ☐
 - Media Security: ☐
- Call Control:**
 - Call Initiation Timeout (s): 4
 - Call Queuing Timeout (m): 5
 - Service Busy Response: 486 - Busy Here
 - on No User Responding Send: 408-Request Timeout
 - Action on CAC Location Limit: Allow Voicemail
 - Suppress Q.850 Reason Header: ☐
 - Emulate NOTIFY for REFER: ☐
 - No REFER if using Diversion: ☐

At the bottom right, there are 'OK' and 'Cancel' buttons.

5.6. Configure User for Mobile Call Control

A new user needs to be created on IP Office in order to use FNE - Mobile Call Control. The FNE Short Code is used by novaalert in order to initiate the Call Intrude and Coaching Intrusion Short Codes.

Navigate to **Primary Server** → **Users** and right-click and select **New** as shown below.



Under the **User** tab, enter a suitable **Name**, **Password**, **Confirm Password** and **Extension** and ensure that **Power User** is selected as the **Profile**.

NovaAlert: 5155

User | Voicemail | DND | ShortCodes | Source Numbers | Telephony | Forwarding | Dial In | Voice Recording | Button Programming | Menu Programming | Mobility

Name: NovaAlert

Password: ••••

Confirm Password: ••••

Unique Identity:

Audio Conference PIN:

Confirm Audio Conference PIN:

Account Status: Enabled

Full Name:

Extension: 5155

Email Address:

Locale:

Priority: 5

System Phone Rights: None

Profile: Power User

- ☐ Receptionist
- ☒ Enable Softphone
- ☒ Enable one-X Portal Services
- ☒ Enable one-X TeleCommuter
- ☒ Enable Remote Worker
- ☒ Enable Communicator
- ☒ Enable Mobile VoIP Client
- ☐ Send Mobility Email
- ☐ Web Collaboration

☐ Exclude From Directory

Device Type: All Other Phone Types

Under the **Telephony** tab and again under the **Supervisor Settings** tab ensure that **Can Intrude** is ticked as shown.

The screenshot shows the 'NovaAlert: 5155' configuration window. The 'Telephony' tab is selected, and within it, the 'Supervisor Settings' sub-tab is active. The 'Call Settings' section includes fields for Login Code, Confirm Login Code, Login Idle Period (secs), Monitor Group (set to '<None>'), Coverage Group (set to '<None>'), Status on No-Answer (set to 'Logged On (No change)'), and Privacy Override Group (set to '<None>'). There is also a 'Reset Longest Idle Time' section with radio buttons for 'All Calls' (selected) and 'External Incoming'. On the right side, a list of checkboxes includes 'Force Login', 'Force Account Code', 'Force Authorization Code', 'Incoming Call Bar', 'Outgoing Call Bar', 'Inhibit Off-Switch Forward/Transfer', 'Can Intrude' (checked), 'Cannot be Intruded', 'Can Trace Calls', and 'Deny Auto Intercom Calls'.

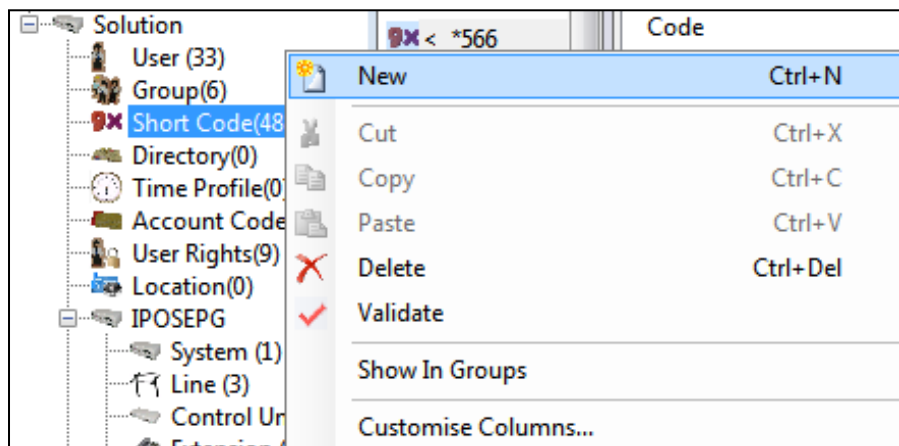
Under the **Mobility** tab, tick the **Mobility Features** box and enter the number associated with novaalert, this is the number configured in **Section 6.1**. Ensure that all the tick boxes shown below are selected. Click on **OK** at the bottom of the screen to complete the setup (not shown).

The screenshot shows the 'NovaAlert: 5155' configuration window with the 'Mobility' tab selected. The 'Internal Twinning' section has 'Twinned Handset' set to '<None>' and 'Maximum Number of Calls' set to '1'. Below this, 'Twin Bridge Appearances', 'Twin Coverage Appearances', and 'Twin Line Appearances' are unchecked. The 'Mobility Features' checkbox is checked. Under 'Mobile Twinning', 'Twinned Mobile Number (including dial access code)' is '0049123456789', 'Twinning Time Profile' is '<None>', 'Mobile Dial Delay (secs)' is '0', and 'Mobile Answer Guard (secs)' is '0'. Several checkboxes are checked: 'Hunt group calls eligible for mobile twinning', 'Forwarded calls eligible for mobile twinning', 'Twin When Logged Out', 'one-X Mobile Client', 'Mobile Call Control', and 'Mobile Callback'.

5.7. Configure Short Codes

Short Codes can be created for both systems, i.e., both the Primary Server and the Expansion Server. A short code such as Call Intrude or Coaching Intrusion would need to be created across all systems so navigate to **Solution → Short Code**, right-click on **Short Code** and select **New** as shown.

Note: A short code may already be in place to dial out to the PSTN, however a new short code will need to be added to dial out to novaalert to create an alarm from IP Office phones.



5.7.1. Short Code for FNE Service

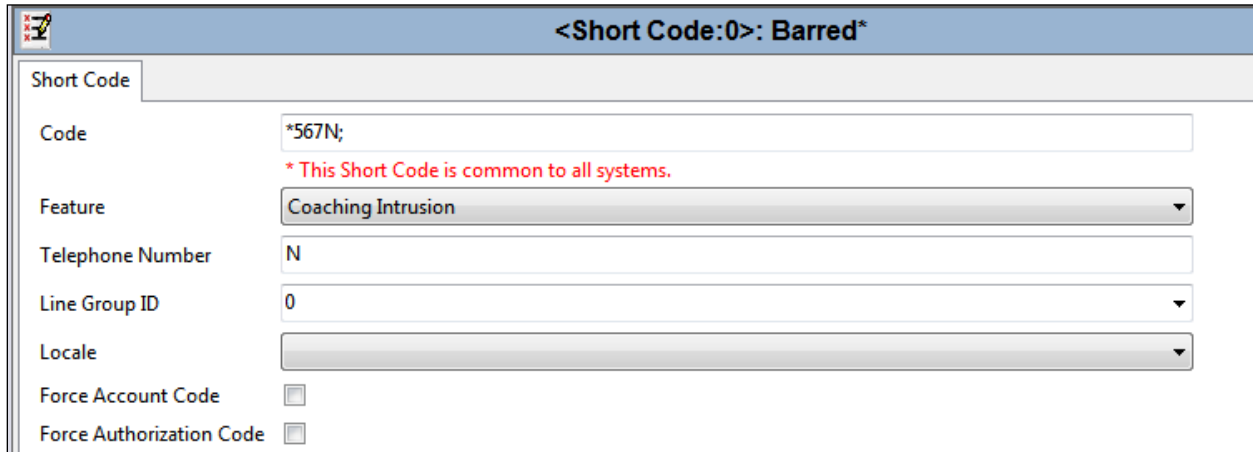
FNE – Mobile Call Control is used to allow a user called or calling the system to invoke mobile call control and to then handle and make calls as if they were at their system extension. FNE **31** is setup as a short code, and this is done as shown below. ***566** is used to initiate the **FNE Service** and this will be configured on the novaalert system in **Section 6.1**.

*566: FNE Service	
Short Code	
Code	<input type="text" value="*566"/>
* This Short Code is common to all systems.	
Feature	<input type="text" value="FNE Service"/>
Telephone Number	<input type="text" value="31"/>
Line Group ID	<input type="text" value="0"/>
Locale	<input type="text"/>
Force Account Code	<input type="checkbox"/>
Force Authorization Code	<input type="checkbox"/>

5.7.2. Short Code for Coaching Intrusion

Coaching Intrusion is used in order to break in on an existing call when the phone set is busy. *567N; was used for this Short Code where N is the number that was dialled. This same Short Code will be configured in **Section 6.1**.

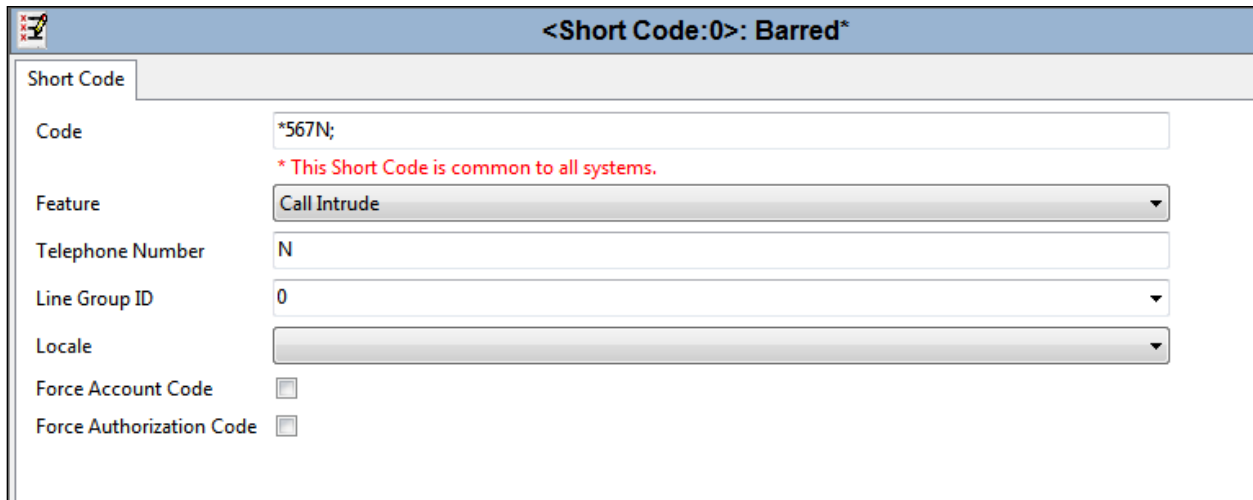
Note: Each user must have "Cannot be intruded" unchecked under the telephony tab.



The screenshot shows a configuration window titled "<Short Code:0>: Barred*". The "Short Code" tab is selected. The "Code" field contains "*567N;". Below it, a red message states: "* This Short Code is common to all systems." The "Feature" dropdown is set to "Coaching Intrusion". The "Telephone Number" field contains "N". The "Line Group ID" dropdown is set to "0". The "Locale" dropdown is empty. The "Force Account Code" and "Force Authorization Code" checkboxes are both unchecked.

5.7.3. Short Code for Call Intrude

The same Short Code is illustrated here for Call Intrude. Note that the difference between Call Intrude and Coaching Intrusion is that Coaching Intrusion allows the Alarm to intrude on another user's call and play without being heard by the other call parties to which they can still talk. Call Intrude will play the Alarm to all users on the call.

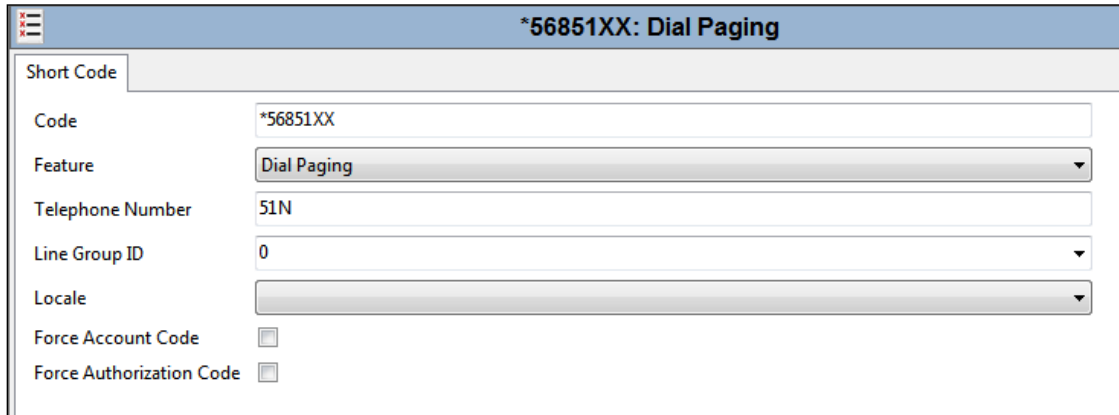


The screenshot shows a configuration window titled "<Short Code:0>: Barred*". The "Short Code" tab is selected. The "Code" field contains "*567N;". Below it, a red message states: "* This Short Code is common to all systems." The "Feature" dropdown is set to "Call Intrude". The "Telephone Number" field contains "N". The "Line Group ID" dropdown is set to "0". The "Locale" dropdown is empty. The "Force Account Code" and "Force Authorization Code" checkboxes are both unchecked.

5.7.4. Short Code for Dial Paging

Dial paging is used to play an alarm directly to the phoneset speaker. When novaalert uses this short code with the extension number, that alarm gets played out on the extension's speaker.

***568** was used as the Short Code for **Dial Paging**, seeing as 51xx is the extension range for the Primary Server the full Short Code is ***56851XX** and this was used to initiate the alarm to extensions 51xx.

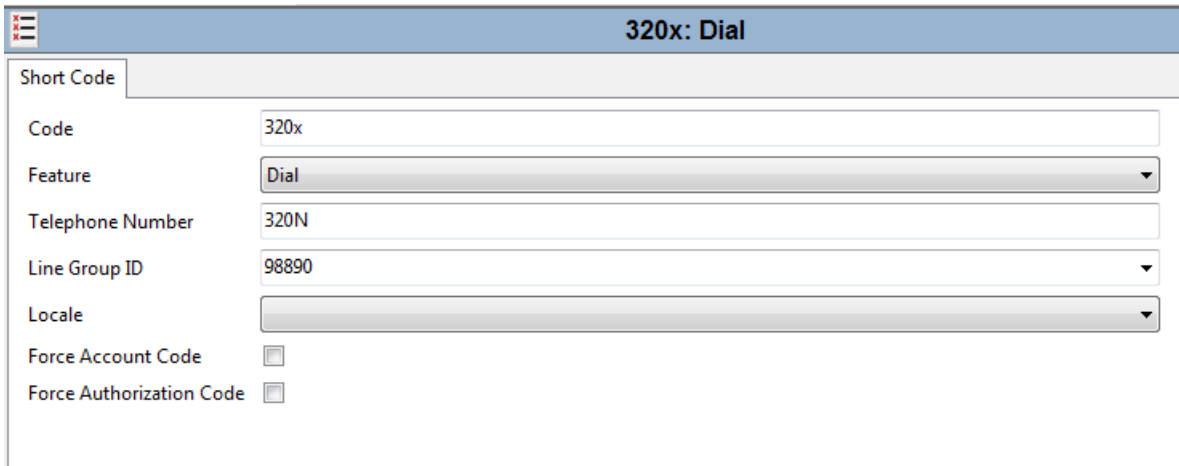


The screenshot shows a configuration window titled '*56851XX: Dial Paging'. It contains the following fields and options:

- Short Code**: *56851XX
- Code**: *56851XX
- Feature**: Dial Paging (selected from a dropdown)
- Telephone Number**: 51N
- Line Group ID**: 0 (selected from a dropdown)
- Locale**: (empty dropdown)
- Force Account Code**: ☐
- Force Authorization Code**: ☐

5.7.5. Short Code to dial into novaalert

The following short code was added to dial out from IP Office over the SIP Trunk created in **Section 5.5**. It was decided that 3200 – 3209 would be assigned to novaalert to dial into various services, therefore **320x** was added as a short code to dial out of the same **Line Group ID** that was created in **Section 5.5**.

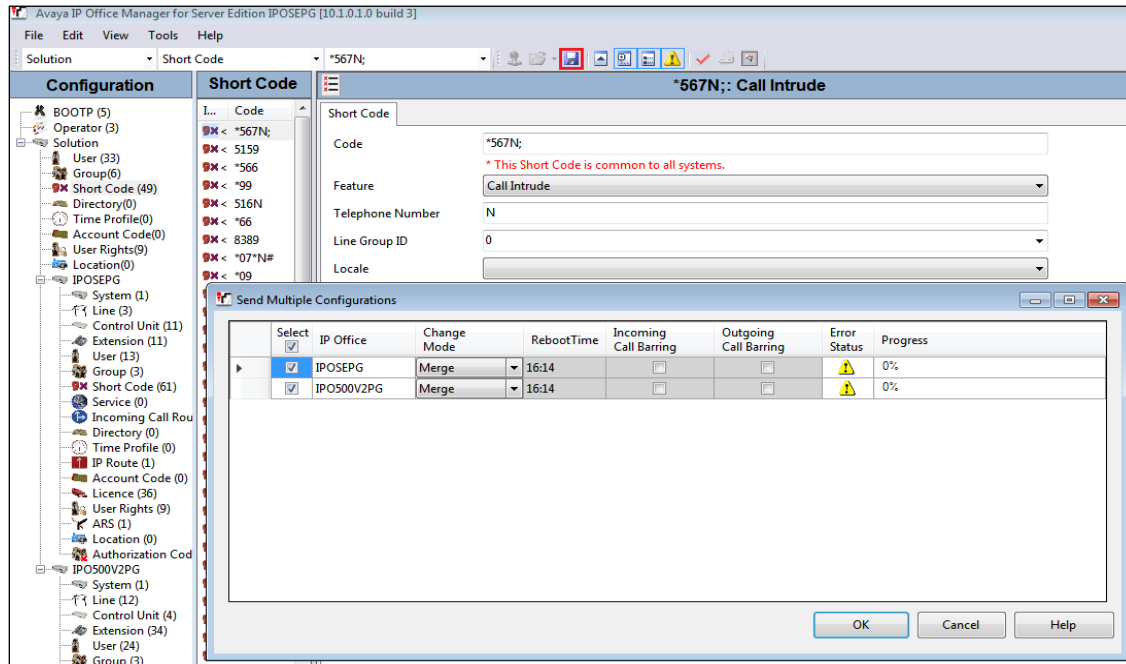


The screenshot shows a configuration window titled '320x: Dial'. It contains the following fields and options:

- Short Code**: 320x
- Code**: 320x
- Feature**: Dial (selected from a dropdown)
- Telephone Number**: 320N
- Line Group ID**: 98890 (selected from a dropdown)
- Locale**: (empty dropdown)
- Force Account Code**: ☐
- Force Authorization Code**: ☐

A similar Short Code was added on the Expansion Cabinet to allow calls to come across to the Primary Server, the Line Group ID will be that of the SCN line. Calls to 320x are made out from the Primary Server so these calls must come to the Primary Server from any and all expansion cabinets first.

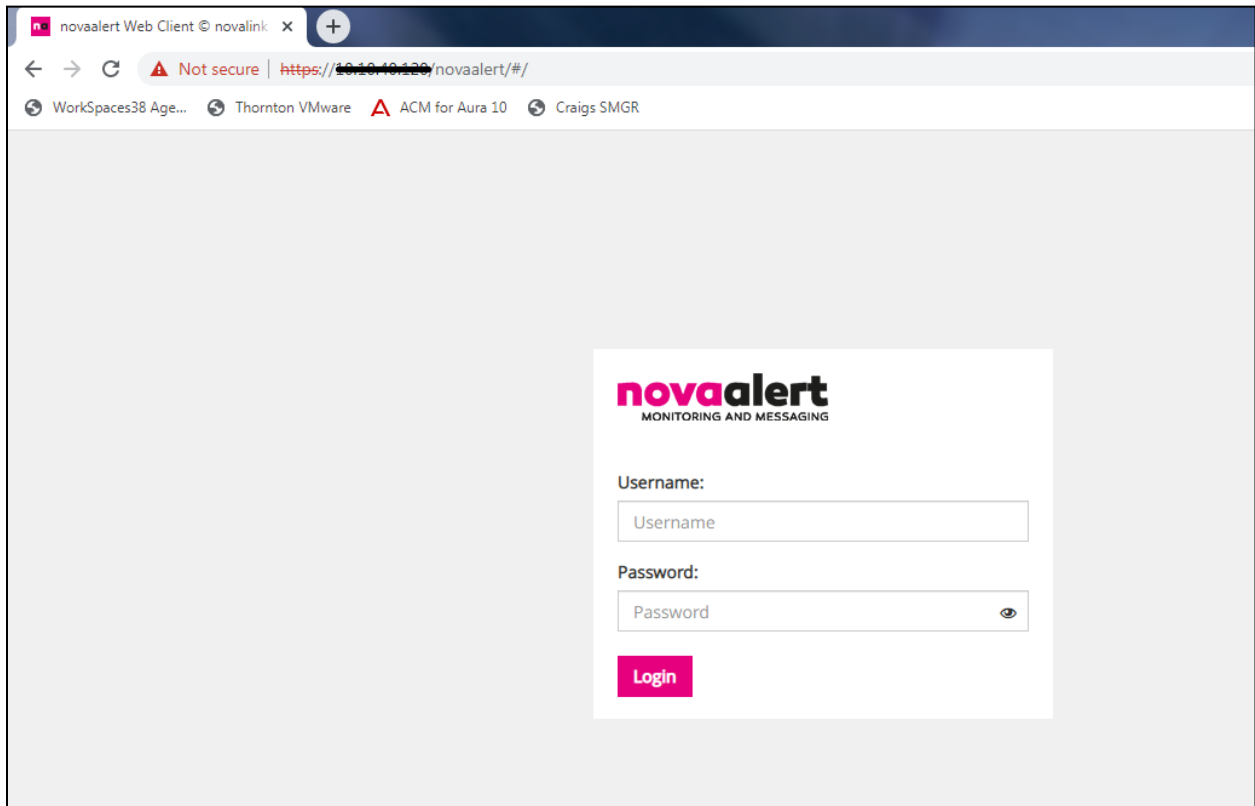
Once the configuration has been made it must be sent to the IP Office. Click on the **Save** Icon at the top left of the screen as shown below. Once the **Save Configuration** window opens, either the **Merge** or **Immediate** button will be filled in depending on the changes that are made. Click on the **OK** button.



6. Configuration of novalink novaalert

It is assumed that novaalert is already installed and configured by a novalink-certified engineer. The following shows the steps that can be carried out in order to make changes or to examine a working system. The screen shots were taken after compliance testing was completed successfully and will show the configuration that was used for a successful integration to IP Office. This can be used as an example of a fully working system.

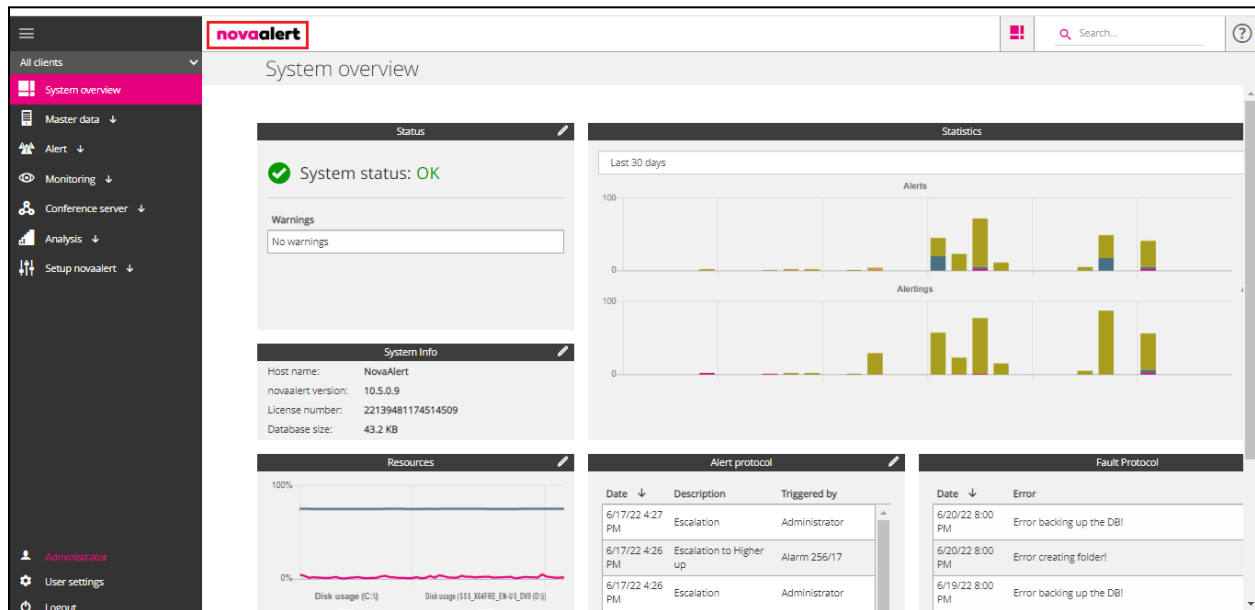
All configuration changes are made to novaalert using a web browser session to the novaalert server. Open a web browser session to the IP Address of the novaalert server followed by /novaalert, for example, for compliance testing **https://<novaalertIP>/novaalert** was used. The following screen shown is asking for the **Username** and **Password**, enter these and click on the **Login** button.



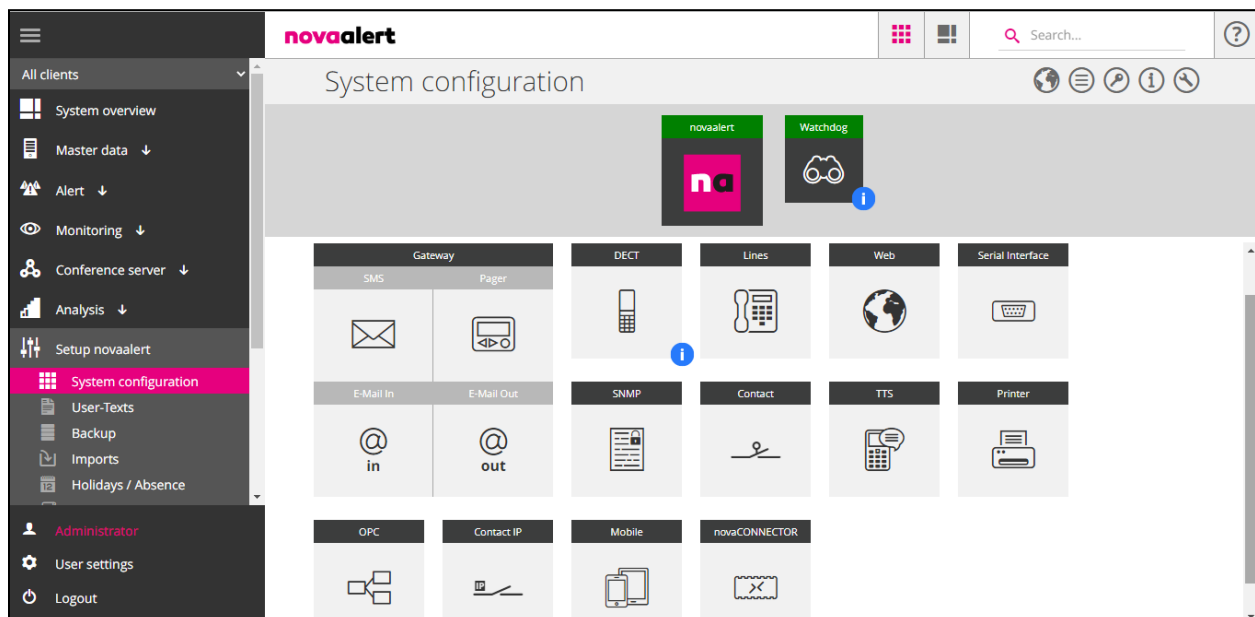
The screenshot displays a web browser window with the title 'novaalert Web Client © novalink'. The address bar shows a 'Not secure' warning and the URL 'https://10.10.10.120/novaalert/#/'. The browser's tab bar includes 'WorkSpaces38 Age...', 'Thornton VMware', 'ACM for Aura 10', and 'Craigs SMGR'. The main content area is a light gray background with a white login box in the center. The login box contains the 'novaalert MONITORING AND MESSAGING' logo, followed by 'Username:' and a text input field, 'Password:' and a password input field with an eye icon, and a pink 'Login' button.

6.1. Connection setup to IP Office (SIP Trunk)

Once logged in, click on the **novaalert** icon at the top of the page, this will get to the System Configuration area.




Click on the **Lines** icon, in the main window. All configuration with regards to the SIP connection to IP Office is set in this area.



The first section shows the **Line Configuration**. Displayed below was the setup used for compliance testing; the most notable field is the **Intrusion code** which is referenced in **Section 5.7**. This allows an alarm to get to a telephone, even if it is busy. The **Intrusion code** is entered using the FNE short code first followed by the Call Intrude/Coaching Intrusion short code, and this looks like ***566;*567<Nr>#**. This will call *566 first then using the FNE Mobile Call Control Service *567xxxx# is entered using DTMF.

Note: It is important to copy and paste the following ***566;*567<Nr>#** directly from here into the **Intrusion code** field on the PC, as the “pipe” icon may not work correctly if typed from the local keyboard.

System configuration > **Lines**

 **Lines**

Line Configuration (Lines)

Intrusion code	<input type="text" value="*566;*567<Nr>#"/>
Line allocation 1	<input type="text" value="1"/>
Line allocation 2	<input type="text" value="2"/>
Line allocation 3	<input type="text" value="3"/>
Line allocation 4	<input type="text" value="4"/>
Line5	<input type="text" value="5"/>
Line6	<input type="text" value="6"/>
Min Connection Time	<input type="text" value="5"/>
Reserved Lines for Alarm Triggering	<input type="text" value="0"/>
Static Direct Alarm	<input type="text"/>
Timeout external calls	<input type="text" value="30"/>
Timeout internal calls	<input type="text" value="30"/>

Select **Voice over IP Configuration** which is the next section. The settings shown below are what were used during compliance testing. Most notable that being **Driver Preferences**, which should be set to **SIP** and the **SIP Gateway** which has the IP Address of the IP Office Primary Server as per **Section 5.1**. If DNS is not being used, please enter the IP Address in both fields, **Realm** and **IP-Address**.

System configuration > Lines

Lines

Search...

Voice over IP Configuration (VoIP)

Driver Preferences

SIP

(DriverPref)

+

✖

H323 GateKeeper Address

(H323_GateKeeperAddr...

+

✖

H323 GateKeeper Password

(H323_GateKeeperPwd)

+

✖

H323 GateKeeper Zone

(H323_GateKeeperZone)

+

✖

H323 Gateway

IP-Address

Prefix

IP-Address

Prefix

(H323_Gateway)

+

✖

Local User Name

DevConnect

(LocalUserName)

+

✖

SIP Alias

Host

Alias

Username

Password

Realm

Host

Alias

Username

Password

Realm

(SIP_Alias)

+

✖

SIP Gateway

Realm

IP-Address

Prefix

Local Interface

devconnect.l...

10.10.40.19

Prefix

Local Interface

(SIP_Gateway)

+

✖

SIP Listener Config

*:5060

(SIP_ListenerConfig)

+

✖

Add entry

Close

Save

Click on **Call Control**, which is the next section down. The following shows the configuration used for compliance testing. The **PBX Type** is set to **Avaya IPO** and the **Card Driver** set to **VoIP (H.323/SIP)**. The **Default Calling Party** is entered and this much match exactly the Twinned Mobile Number configured for the FNE User in **Section 5.6. Signaling outgoing DTMF** is chosen as shown on the next page.

System configuration > Lines

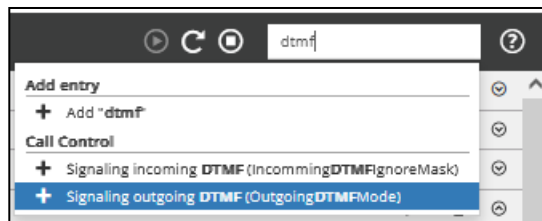
Setting	Value	Configuration Icon
Call Retries	2	(CallVersuche)
Calling Name Identification	Yes	(CNIPAktiv)
Calling Party Configuration	Yes	(CallingPartyAktiv)
Card Driver	VoIP (H.323/SIP)	(CardDriver)
Default Calling Party	0049123456789	(DefaultCallingParty)
Dialed Number Identification	Use called party information	(GewählteNummer)
Interface	VoIP	(Interface)
Intrusion Configuration	Recall with add. intrusion digits prior call no.	(AufschaltenAktiv)
Minimum Digits	0	(MinDigits)
PBX Type	Avaya IPO	(PBXType)
QSIG Standard	Disabled	(QSIGStandard)
Signaling outgoing DTMF	As sound formatted information message (H.245 sign)	(OutgoingDTMFMode)
Timeout Call List	8	(RufZeitAnrufliste)

Close Save

Intrusion Configuration is set as follows, **Recall with add. intrusion digits prior call no.**

Setting	Value
Call Retries	2
Calling Name Identification	Yes
Calling Party Configuration	Yes
Card Driver	<No selection>
Default Calling Party	At 1st call with intrusion digits prior call no.
Dialed Number Identification	Native Intrusion per QSIG
Interface	Native Intrusion per QSIG at 1st call
Intrusion Configuration	Recall with add. intrusion digits prior call no.
Minimum Digits	0

For compliance testing, the **Signaling outgoing DTMF** field was not present, and this field needed to be added manually. From the top right corner there is a search field where dtmf can be entered as shown below, this will bring up the various fields that can be added for DTMF. The **Signaling outgoing DTMF (OutgoingDTMFMode)** is added.



This entry is added to the **Call Control** Section and the **Key** is **OutgoingDTMFMode** as shown.

 A screenshot of the 'Add entry' form. The form has a title 'Add entry' at the top. Below the title, there is a 'Section:' label followed by a dropdown menu showing 'Call Control'. Below that, there is a 'Key:' label followed by a text input field containing 'OutgoingDTMFMode'. At the bottom right of the form, there are two buttons: 'Close' and 'Add'.

Signaling outgoing DTMF will determine what DTMF is used by novaalert when sending digits to IP Office. For compliance testing SIP Info was used and this must be set up on the SIP Line as shown in **Section 5.5**. The corresponding setting here is **As sound formatted information message (H.245 signal or SIP INFO)**.

 A screenshot of a configuration table. The table has four rows: 'Signaling outgoing DTMF', 'TLS mode', 'TLS Secure RTP', and 'TLS local certificate'. The 'Signaling outgoing DTMF' row is selected, and its dropdown menu is open. The dropdown menu shows several options: '<No selection>', 'Default setting for the chosen protocol', 'Q.931 Information Elements (H.323 only)', 'Simple string as information message (H.245 string or SIP INFO)', 'As sound formatted information message (H.245 signal or SIP INFO)', 'According to RFC 2833 as RTP package', and 'In-Band DTMF tones'. The option 'As sound formatted information message (H.245 signal or SIP INFO)' is highlighted in blue.

With everything entered correctly, click **Save** at the bottom right of the screen.

Calling Party Configuration	Yes	(CallingPartyAktiv)	+-	✖
Card Driver	VoIP (H.323/SIP)	(CardDriver)	+-	✖
Default Calling Party	0049123456789	(DefaultCallingParty)	+-	✖
Dialed Number Identification	Use called party information	(GewählteNummer)	+-	✖
Interface	VoIP	(Interface)	+-	✖
Intrusion Configuration	Recall with add. intrusion digits prior call no.	(AufschaltenAktiv)	+-	✖
Minimum Digits	0	(MinDigits)	+-	✖
PBX Type	Avaya IPO	(PBXType)	+-	✖
QSIG Standard	Disabled	(QSIGStandard)	+-	✖
Signaling outgoing DTMF	As sound formatted information message (H.245 signal)	(OutgoingDTMFMode)	+-	✖
Timeout Call List	<No selection> According to RFC 2833 as RTP package As sound formatted information message (H.245 signal or SIP INFO) Default setting for the chosen protocol In-Band DTMF tones Q.931 Information Elements (H.323 only)	(RufzeitAnrufliste)	+-	✖
		Add entry		
			Close	Save

Changes saved successfully should be displayed at the top right of the screen and **Close** can then be clicked at the bottom right.

System overview > Lines

Lines

Line Configuration (Lines)

Fax Configuration (Fax)

Radio Configuration (Radio)

Voice over IP Configuration (VoIP)

Call Control (CallInfo)

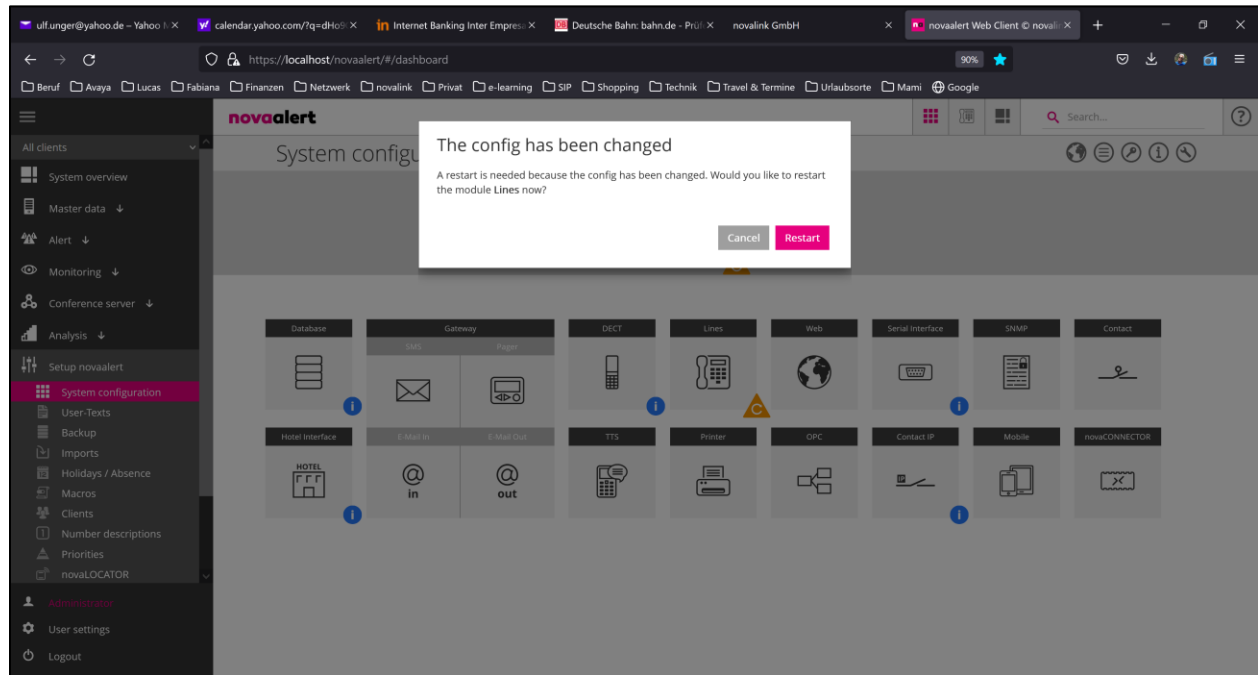
Search...

Changes saved successfully!

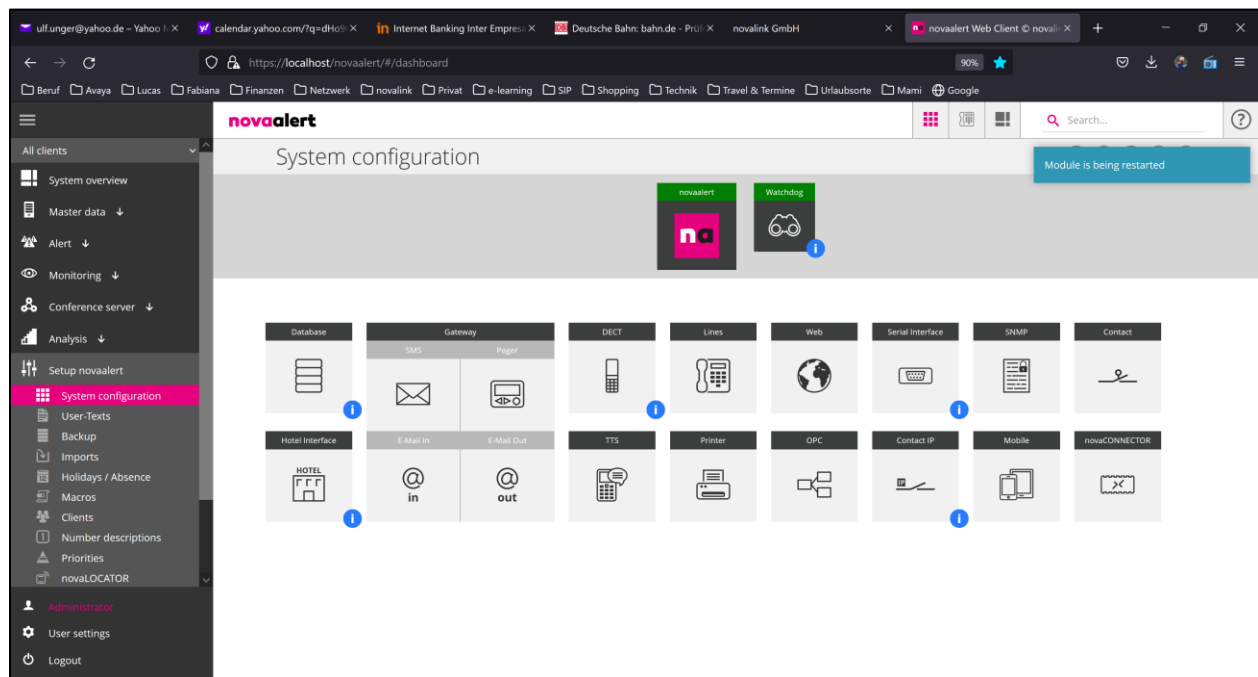
Close

Save

Once the setup is saved the following screen is popped asking to restart the module, click on **Restart**.



A message is displayed in the top right corner saying **Restarted module successfully** (not shown).



6.2. Create an Alarm to send to IP Office

An alarm can be created and sent to a single IP Office user or a group of IP Office users. This section outlines the steps required to create an alarm that is ready to be sent.

In order to send an alarm to IP Office, a user/extension will need to be added. This extension is then called by novaalert when the alarm is activated. From the main menu, navigate to **Master data** → **User master data**. In the main window select **Add new** as shown below.

No.	Name	Personal no.	Client	Username
17	1140 5221		All	
12	1603SW-I 5152		All	
11	1616I 5151		All	
15	9508 5201		All	
16	9641 5250		All	
13	Communicator WIN 5122		All	
21	Hunt Group IP500 V2		All	
22	Hunt Group IPO ALL		All	
20	Hunt Group SE		All	
23	Huntgroup Paging		All	
19	ISDN PSTN		All	
14	J129 5123		All	
1	NovaLink SYSTEM		All	Administrator
18	SIP PSTN		All	

Note: The following screens show the data for an existing user, these are used to demonstrate what is required when adding a new user. Click on the **Common** tab and enter a suitable **Name** and **PIN code**.

Edit person 9641 5250 (16) Add new

Common Numbers Authorization Mobile/Desktop/Touch Allocation

Name: 9641 5250

PIN code: 1234

Additional information:

Street:

Zip / City:

Language: English

Client: All

Personal number:

Inactive: ☐

Logged in: ●

No parallel alerts: ☐

Notes:

Click on the **Numbers** tab and enter the IP Office telephone number for this user and click on **Save Changes** at the bottom of the screen (not shown).

Common	Numbers	Authorization	Mobile/Desktop/Touch	Allocation
<div> <div>Office 1:</div> <div>5250</div> <div>✓</div> </div> <div>Office 2:</div> <div></div> <div>✓</div>				
<div> <div>Home 1:</div> <div>5250</div> <div>✓</div> </div> <div>Home 2:</div> <div></div> <div>✓</div>				
<div> <div>Mobile 1:</div> <div></div> <div>✓</div> </div> <div>Mobile 2:</div> <div></div> <div>✓</div>				
<div> <div>Pager 1:</div> <div></div> <div>Tone call</div> <div>✓</div> </div> <div>Pager 2:</div> <div></div> <div>Tone call</div> <div>✓</div>				
<div> <div>SMS GSM 1:</div> <div></div> <div>✓</div> </div> <div>SMS GSM 2:</div> <div></div> <div>✓</div>				

The next step is to create the Alert Definition, navigate to **Alert definition** in the left window and click on **Add new** in the main window.

Again, this example shows an existing Alert and is used to demonstrate what needs to be configured for any Alert definition. Click on the **Common** tab and enter a suitable **Description**. The **Alert type** can be set depending on the type of Alert; this was set to **Group Call** for the example below. A **PIN code for trigger** also needs to be added.

Edit alert Alarm to H323 5250 (13) Add new

Common Messages Alert-list Alert interfaces Escalation Mobile/Desktop/Touch Various

Description:
Alarm to H323 5250

PIN code for trigger:
1234

Priority:
Highest Priority

Alert type:
Group Call

Voice-No.:
68

Number of attempts:
1

Client:
All

Number of person to be contacted:
All

Notes:

Click on the **Messages** tab, a message can be delivered to the phone set display by opening the **Phone display** section and entering a suitable **Message** as shown below.

Common **Messages** Alert-list Alert interfaces Escalation Mobile/Desktop/Touch Various

Fill messages with alert description

Phone display

Message:
This is an Alarm Message

Event text:
No

Call type:
Duration

Phone TTS

Numeric pager

Alphanumeric pager

SMS GSM

WLAN/DECT paging

The list of users to be alerted by this alarm is entered under the **Alert-list** tab. In the example below one user **5250** (that created previously in this section) was added. However, a number of users can be added here depending on who should receive the alarm. The **Intr.** tick box was checked which would allow call intrusion for this user. If the user is busy, then the alarm can intrude on the call and get played.

Name	Medium / State	Conf.	Aknw.	Intr.	Logg.	Delay
↑↓ 9641 5250 (16)	Office 1	<input type="checkbox"/>	<input type="checkbox"/>	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>	0

Under the **Escalation** tab an Escalation can be added to send the alarm to another user such as a manager or perhaps a secretary if the initial user fails to answer the alarm. This escalation must be configured first (not shown here) but can then be referenced under this Escalation tab.

Click on **Save** at the bottom right of the screen (not shown) and this will save the Alert Definition. This concludes the setup of an alarm that will be sent to this IP Office user 5250.

Condition	Pers. count	Add	Alert	Ori. text
↑↓ Num. of persons <	1	<input type="checkbox"/>	Escalation (33)	<input type="checkbox"/>

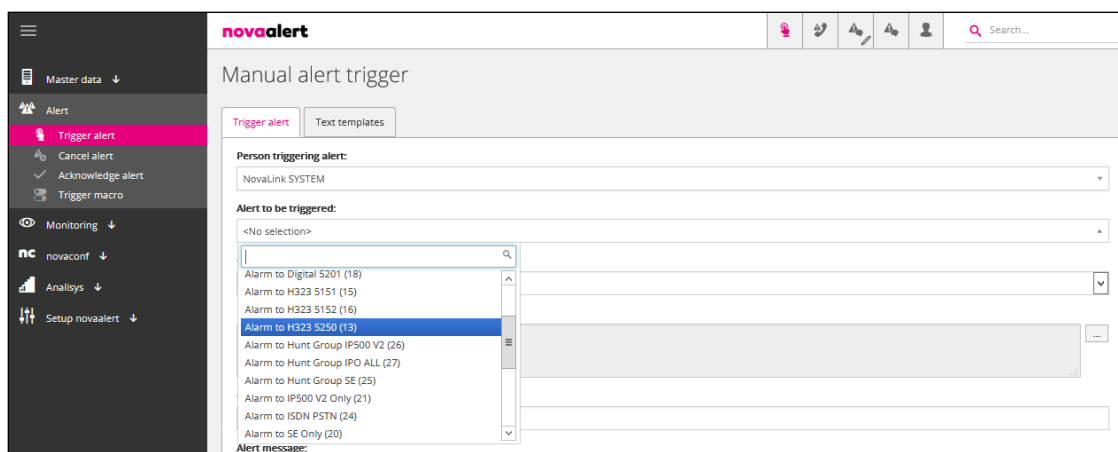
7. Verification Steps

This section illustrates the steps necessary to verify that the novaalert is configured correctly to send an alarm to extensions on IP Office using SIP trunks.

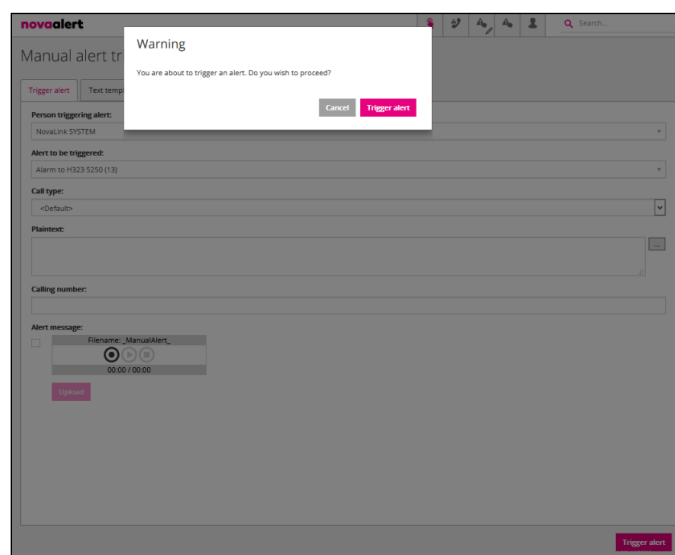
7.1. Trigger an Alarm on novaalert

Log into novaalert as per **Section 6**. From the left menu navigate to **Alert → Trigger alert**. From the main window click on the **Alert to be triggered** drop down box and select the Alert to be triggered. In the example below the alert was **Alarm to H323 5250** which was created in **Section 6.2**.

Note: Typically, alarms are sent to multiple endpoints, but for simplicity of demonstrating this one endpoint was chosen.

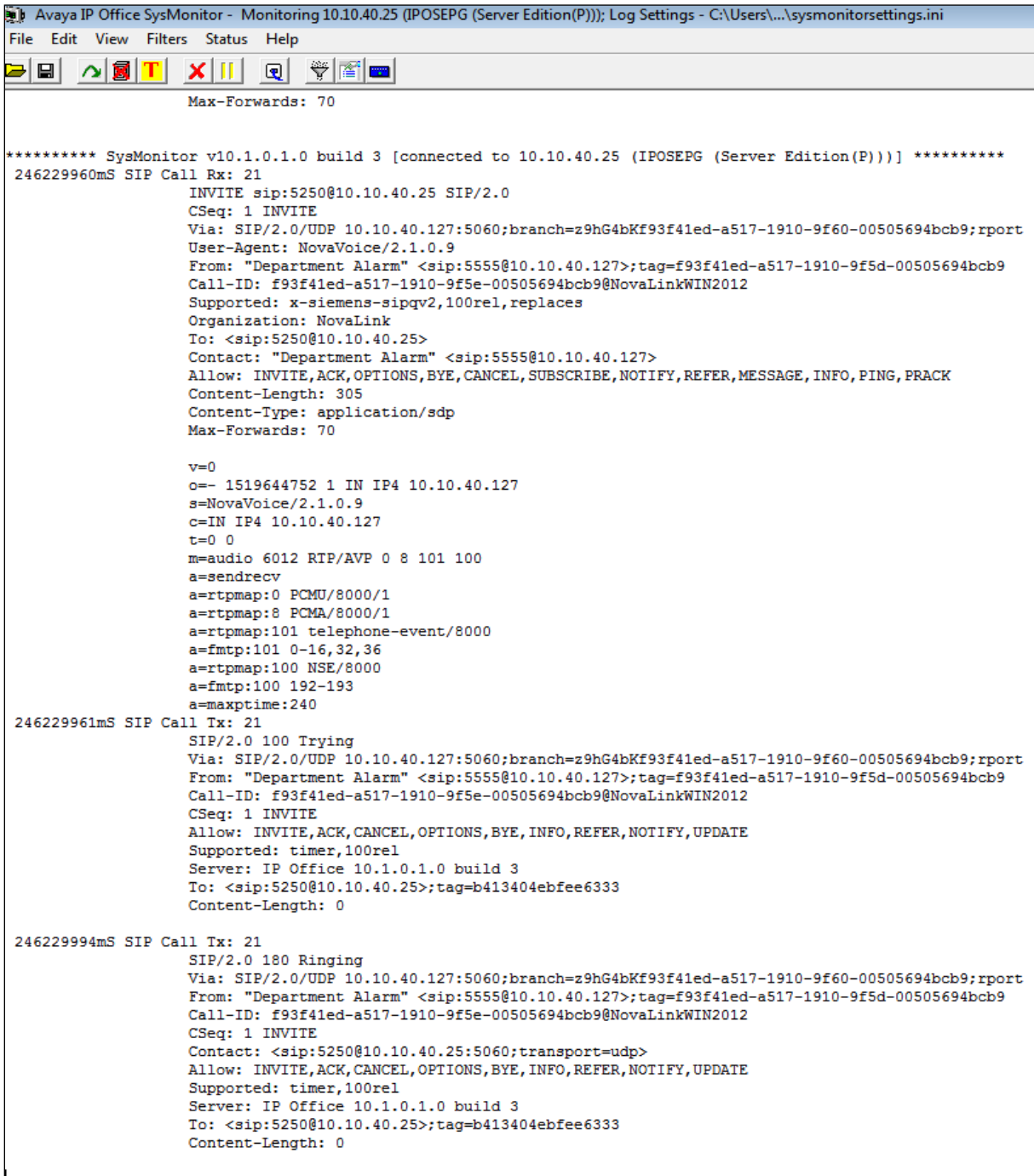


Click on **Trigger alert** at the bottom right of the screen and a window opens asking to confirm the alarm trigger. Click on **Trigger alert** in that window.



7.2. Verify SIP Trunk Messages

SIP messages can be viewed by opening the IP Office **SysMonitor** as shown below. Click on **Filters** at the top of the screen and select the appropriate SIP messages that are to be viewed. This will then display all these filtered SIP messages coming to and going from the IP Office. If there is an issue with the alarms not being sent, then this is a way to troubleshoot what is happening.



The screenshot shows the Avaya IP Office SysMonitor application window. The title bar reads "Avaya IP Office SysMonitor - Monitoring 10.10.40.25 (IPOSEPG (Server Edition(P))); Log Settings - C:\Users\...\sysmonitorsettings.ini". The menu bar includes File, Edit, View, Filters, Status, and Help. Below the menu bar is a toolbar with various icons. The main display area shows a log of SIP messages. The first message is an INVITE from "Department Alarm" to sip:5250@10.10.40.25. The second message is a SIP/2.0 100 Trying response. The third message is a SIP/2.0 180 Ringing response. The log text is as follows:

```
***** SysMonitor v10.1.0.1.0 build 3 [connected to 10.10.40.25 (IPOSEPG (Server Edition(P)))] *****
246229960mS SIP Call Rx: 21
    INVITE sip:5250@10.10.40.25 SIP/2.0
    CSeq: 1 INVITE
    Via: SIP/2.0/UDP 10.10.40.127:5060;branch=z9hG4bKf93f41ed-a517-1910-9f60-00505694bcb9;rport
    User-Agent: NovaVoice/2.1.0.9
    From: "Department Alarm" <sip:5555@10.10.40.127>;tag=f93f41ed-a517-1910-9f5d-00505694bcb9
    Call-ID: f93f41ed-a517-1910-9f5e-00505694bcb9@NovaLinkWIN2012
    Supported: x-siemens-sipqv2,100rel,replaces
    Organization: NovaLink
    To: <sip:5250@10.10.40.25>
    Contact: "Department Alarm" <sip:5555@10.10.40.127>
    Allow: INVITE,ACK,OPTIONS,BYE,CANCEL,SUBSCRIBE,NOTIFY,REFER,MESSAGE,INFO,PING,PRACK
    Content-Length: 305
    Content-Type: application/sdp
    Max-Forwards: 70

    v=0
    o=- 1519644752 1 IN IP4 10.10.40.127
    s=NovaVoice/2.1.0.9
    c=IN IP4 10.10.40.127
    t=0 0
    m=audio 6012 RTP/AVP 0 8 101 100
    a=sendrecv
    a=rtpmap:0 PCMU/8000/1
    a=rtpmap:8 PCMA/8000/1
    a=rtpmap:101 telephone-event/8000
    a=fmtp:101 0-16,32,36
    a=rtpmap:100 NSE/8000
    a=fmtp:100 192-193
    a=maxptime:240
246229961mS SIP Call Tx: 21
    SIP/2.0 100 Trying
    Via: SIP/2.0/UDP 10.10.40.127:5060;branch=z9hG4bKf93f41ed-a517-1910-9f60-00505694bcb9;rport
    From: "Department Alarm" <sip:5555@10.10.40.127>;tag=f93f41ed-a517-1910-9f5d-00505694bcb9
    Call-ID: f93f41ed-a517-1910-9f5e-00505694bcb9@NovaLinkWIN2012
    CSeq: 1 INVITE
    Allow: INVITE,ACK,CANCEL,OPTIONS,BYE,INFO,REFER,NOTIFY,UPDATE
    Supported: timer,100rel
    Server: IP Office 10.1.0.1.0 build 3
    To: <sip:5250@10.10.40.25>;tag=b413404ebfee6333
    Content-Length: 0
246229994mS SIP Call Tx: 21
    SIP/2.0 180 Ringing
    Via: SIP/2.0/UDP 10.10.40.127:5060;branch=z9hG4bKf93f41ed-a517-1910-9f60-00505694bcb9;rport
    From: "Department Alarm" <sip:5555@10.10.40.127>;tag=f93f41ed-a517-1910-9f5d-00505694bcb9
    Call-ID: f93f41ed-a517-1910-9f5e-00505694bcb9@NovaLinkWIN2012
    CSeq: 1 INVITE
    Contact: <sip:5250@10.10.40.25:5060;transport=udp>
    Allow: INVITE,ACK,CANCEL,OPTIONS,BYE,INFO,REFER,NOTIFY,UPDATE
    Supported: timer,100rel
    Server: IP Office 10.1.0.1.0 build 3
    To: <sip:5250@10.10.40.25>;tag=b413404ebfee6333
    Content-Length: 0
```

7.3. novaalert on different media

Below are screen shots which show novaalert in various other environments, for example a mobile phone with novaalert **mobileAPP**, showing an alert below.

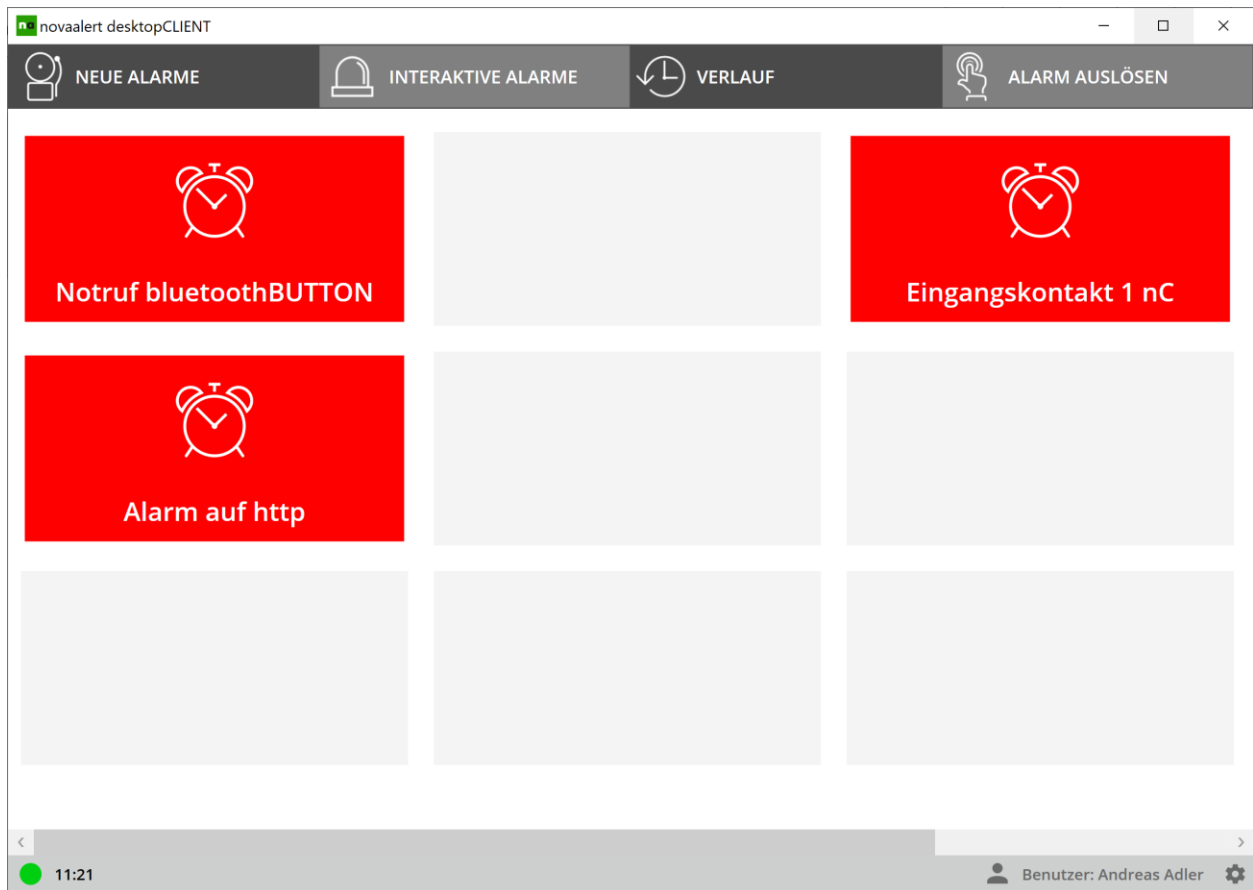
Note: These were not taken as part of compliance testing but are here to demonstrate the use of novaalert on different media.



This example shows a Wallboard and **touchCLIENT**, to receive and trigger alerts.



The example below shows the **desktopCLIENT**, to be used on Windows PCs., to receive and trigger alerts.



8. Conclusion

These Application Notes describe the configuration steps required for novaalert v10 from novalink to interoperate with Avaya IP Office R11.1. All feature functionality and serviceability test cases were completed successfully with any issues and observations noted in **Section 2.2**.

9. Additional References

This section references the Avaya and novalink product documentation that are relevant to these Application Notes.

Product documentation for Avaya products may be found at <http://support.avaya.com>.

[1] *Avaya IP Office R11.1 Manager 11.1*

[2] *Avaya IP Office R11.1 Doc library*

Technical support can be obtained for novaalert from the website <http://www.novalink.ch/en/> or from the following.

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Businessstower
Zuercherstrasse 310
8500 Frauenfeld
Switzerland
helpdesk@novalink.ch
Phone: +41 52 762 66 77
Fax: +41 52 762 66 99

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