



Avaya Solution & Interoperability Test Lab

Application Notes for CSS Mindshare 100500 MaxPlus Dispatch Console integration with Avaya IP Office 11.1 - Issue 1.0

Abstract

These Application Notes describe the configuration steps required to integrate CSS Mindshare 100500 MaxPlus Dispatch Console 3.27.2 with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500 V2 Expansion System 11.1. CSS Mindshare 100500 MaxPlus Dispatch Console incorporates telephony to integrate both radio and telephone functions. This solution also includes Console Builder for creating a user console.

Readers should pay attention to **Section 2**, in particular the scope of testing as outlined in **Section 2.1** as well as any 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 required to integrate CSS Mindshare 100500 MaxPlus Dispatch Console 3.27.2 with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500 V2 Expansion System 11.1. CSS Mindshare 100500 MaxPlus Dispatch Console is purposely built for radio dispatch applications required for 24/7 operation. By integrating PC and audio processor components into a single device, CSS Mindshare 100500 Max Plus Dispatch Console provides a complete dispatch console workstation.

2. General Test Approach and Test Results

The interoperability testing scope is limited to MaxPlus Dispatch Console telephony integration with IP Office. The interoperability compliance test included feature and serviceability testing. The feature testing focused on establishing calls between MaxPlus Dispatch Console, Avaya SIP / H.323 desk phones, and the PSTN, and exercising basic telephony features, such as hold/resume, mute, and transfer. MaxPlus Dispatch does not support conferencing. Additional telephony features, such as call forward, call coverage, call park/unpark, and call pickup were also verified using IP Office Short Codes. The serviceability testing focused on verifying that MaxPlus Dispatch Console comes back into service after IP network interruption.

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 systems and MaxPlus Dispatch Console did not include use of any specific encryption features as requested by CSS Mindshare.

2.1. Interoperability Compliance Testing

Interoperability compliance testing covered the following features and functionality:

- SIP user registration of MaxPlus Dispatch Console with IP Office.
- Calls between MaxPlus Dispatch Console and Avaya SIP / H.323 Deskphones with Direct IP Media (Shuffling) enabled and disabled.
- Calls between MaxPlus Dispatch Console and the PSTN.
- G.711MU and G.729 codec support.
- Proper DTMF tone generation.
- Basic telephony features including hold/resume, mute, redial, and blind and attended call transfer.
- Extended telephony features using IP Office short codes for Call Forwarding, Call Park/Unpark, and Call Pickup All.
- Use of programmable buttons (Console Builder button controls) for Call Pickup All on MaxPlus Dispatch Console.
- Proper system recovery after a loss of IP connectivity.

2.2. Test Results

All test cases passed with the following observations:

- Call Conferencing is not supported.
- Voicemail MWI is not supported.
- Call on Hold Reminder is not supported.
- Audio tones for invalid numbers or outbound call screening are not given but MaxPlus Dispatch Console line indicator display notifications are made.
- MaxPlus Dispatch Console line indicator display does not show called parties. In most cases, MaxPlus Dispatch Console line indicator displays the calling party only during ringing. Once that call is answered, the display clears. The display will continue to show the calling party if the call is not answered as per MaxPlus Dispatch Console design.
- If a call is answered, MaxPlus Dispatch Console line indicator display may show the calling party. Variations occur among the type of endpoints calling and whether they are registered to IP Office Server Edition or IP Office 500 V2 Expansion System.
- Calling the IP Office voicemail system via IP Office default short code *17 is not reliable. This is addressed in future MaxPlus Dispatch Console release 3.28.3.
- Calls cannot be forwarded on busy/ring no answer/forward unconditional to MaxPlus Dispatch Console. This is addressed in future MaxPlus Dispatch Console release 3.28.4.

2.3. Support

For technical support and information on MaxPlus Dispatch Console, contact CSS Mindshare Technical Support at:

- Phone: +1 402-261-8688 x2
- Email: techsupport@css-mindshare.com
- Website: <https://support.css-mindshare.com>

3. Reference Configuration

Figure 1 illustrates a sample configuration with an Avaya SIP-based network:

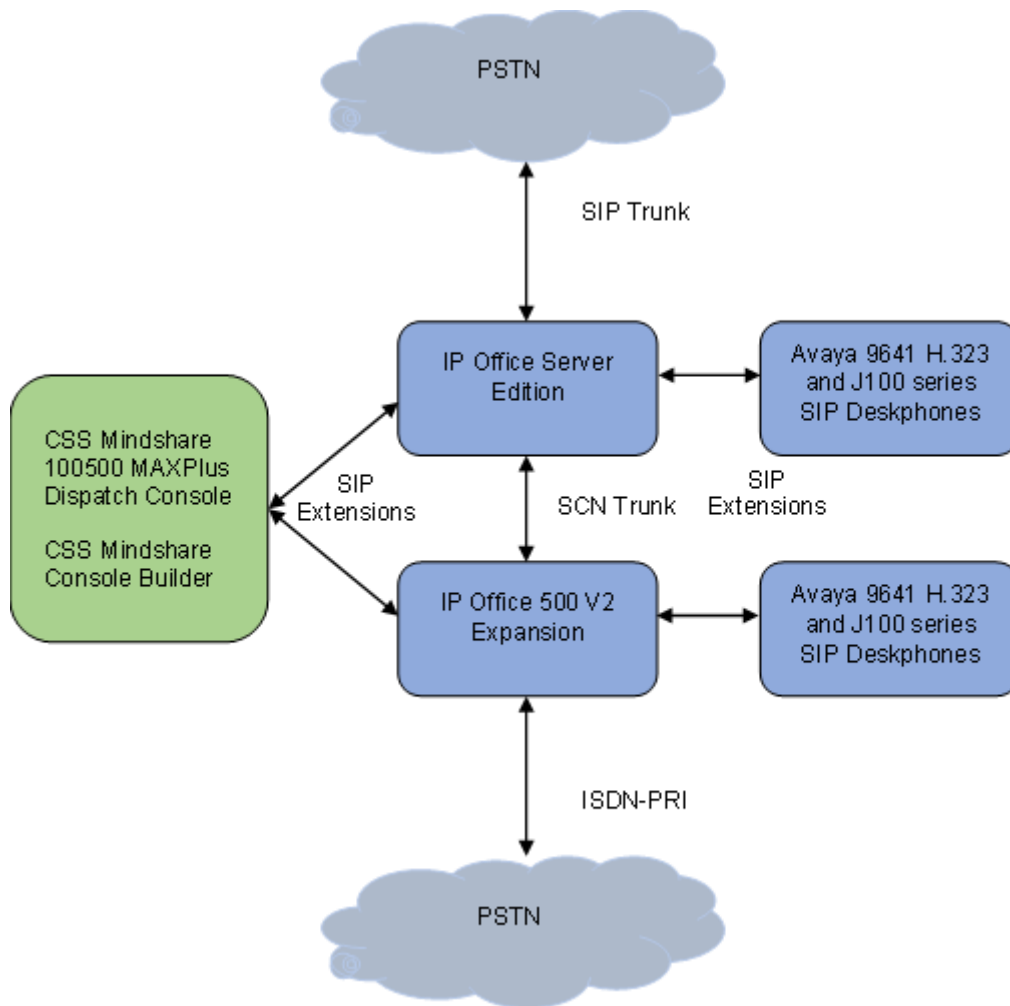


Figure 1: Avaya SIP Network with CSS Mindshare 100500 MaxPlus Dispatch Console

4. Equipment and Software Validated

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

Equipment/Software	Release/Version
Avaya IP Office 500 V2 Expansion	11.1.2.2.0
Avaya IP Office Server Edition	11.1.2.2.0
Avaya 9641G IP Deskphone	6.8.3.0.4 (H.323)
Avaya J179 IP Phone	4.0.7.0.7 (SIP)
CSS Mindshare 100500 MaxPlus Dispatch Console	3.27.2 Debian GNU/Linux 10 (buster) Gnome 3.30.2
CSS Mindshare Console Builder	3.27.2

***Note:** 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.*

5. Configure Avaya IP Office Server Edition

This section provides the procedures for configuring Avaya IP Office Server Edition. The procedures include the following areas:

- Verify License
- Obtain LAN IP Address
- Administer SIP Registrar
- Administer IP Codecs
- Administer SIP Extension for MaxPlus Dispatch Console
- Administer SIP User for MaxPlus Dispatch Console

Note: This section covers the configuration of Avaya IP Office Server Edition, but the configuration is the same for Avaya IP Office 500 V2 Expansion System.

5.1. Verify License

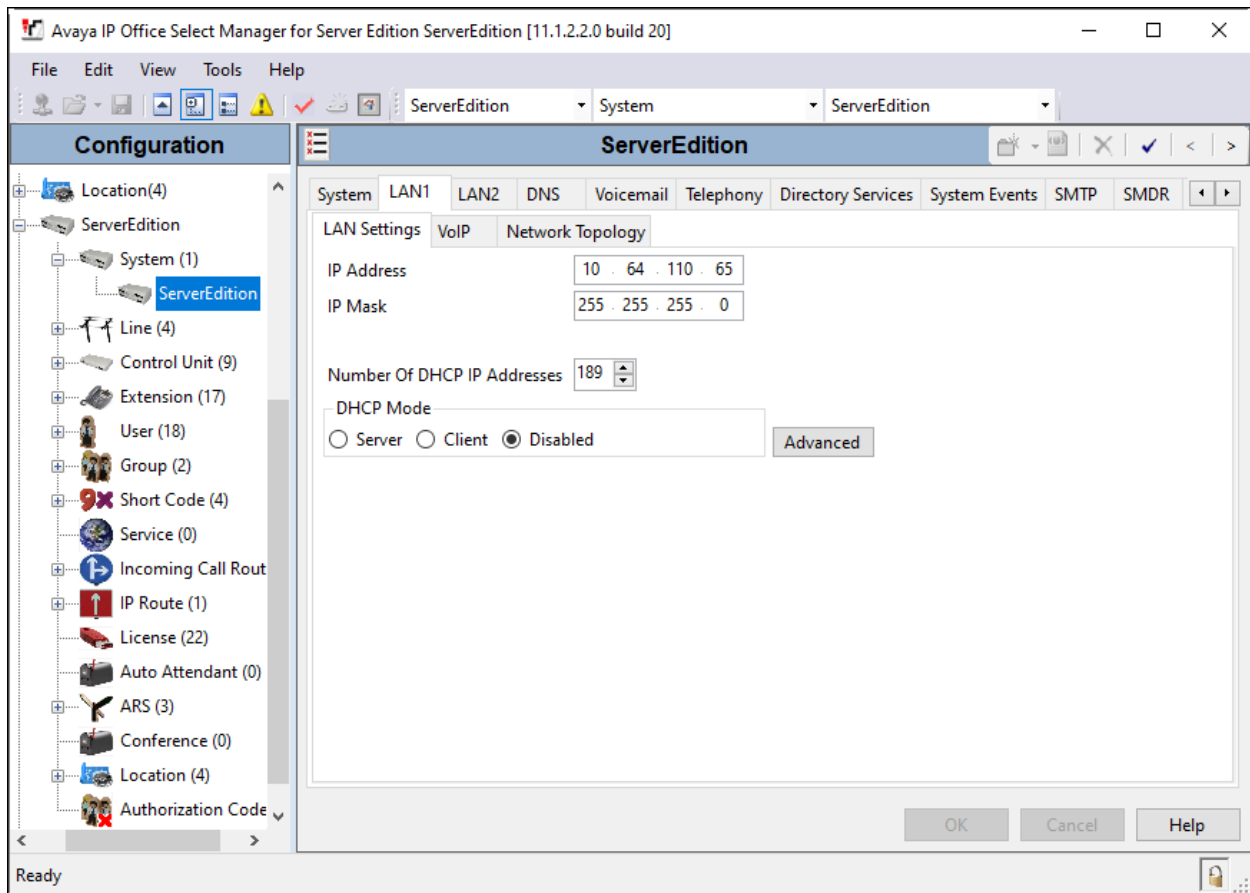
From a PC with **IP Office Admin Suite** installed, invoke **IP Office Manager**. Select the proper primary IP Office system (not shown), and log in using the appropriate credentials. Avaya IP Office Manager for Server Edition screen is displayed. From the configuration tree in the left pane, select **License** under the IP Office system that will be used to display a list of licenses in the right pane. Verify that there are sufficient licenses for **3rd Party IP Endpoints**.

The screenshot displays the 'Avaya IP Office Select Manager for Server Edition ServerEdition [11.1.2.2.0 build 20]' application window. The left pane shows a configuration tree with 'License (22)' selected. The right pane shows a table of licenses under the 'Remote Server' tab.

Feature	Instances	Status	Expiration Date	Source
Receptionist	10	Valid	Never	PLDS Nodal
Additional Voicemail Pro Ports	252	Valid	Never	PLDS Nodal
VMPro Recordings Administrators	1	Valid	Never	PLDS Nodal
Office Worker	1000	Valid	Never	PLDS Nodal
VMPro TTS Professional	40	Valid	Never	PLDS Nodal
IPSec Tunnelling	1	Obsolete	Never	PLDS Nodal
Power User	1000	Valid	Never	PLDS Nodal
Avaya IP endpoints	1000	Valid	Never	PLDS Nodal
SIP Trunk Channels	256	Valid	Never	PLDS Nodal
IP500 Universal PRI (Additional cha...	100	Obsolete	Never	PLDS Nodal
CTI Link Pro	1	Valid	Never	PLDS Nodal
Wave User	16	Obsolete	Never	PLDS Nodal
3rd Party IP Endpoints	1000	Valid	Never	PLDS Nodal
Server Edition	150	Valid	Never	PLDS Nodal
UMS Web Services	1000	Valid	Never	PLDS Nodal
Avaya Mac Softphone	1000	Valid	Never	PLDS Nodal
Avaya Softphone Licence	1000	Valid	Never	PLDS Nodal
SM Trunk Channels	128	Valid	Never	PLDS Nodal

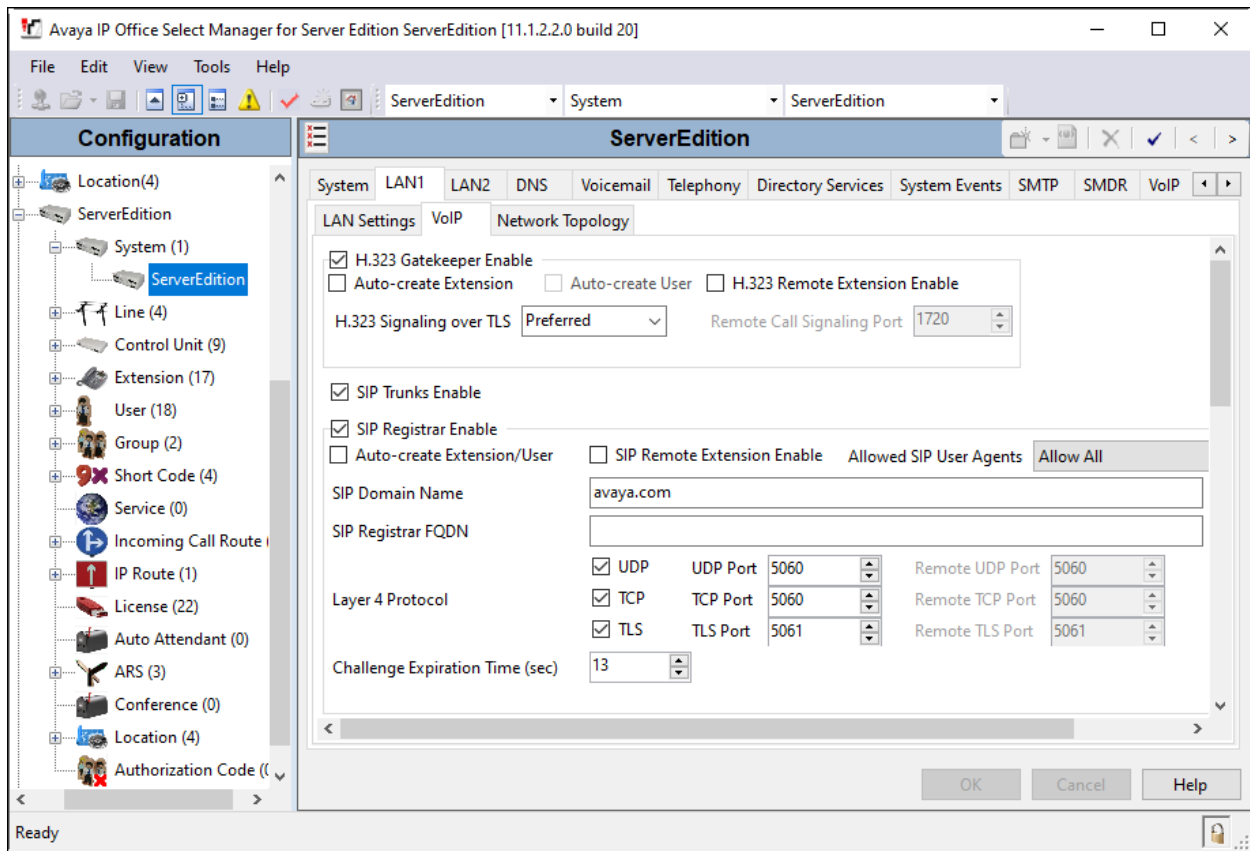
5.2. Obtain LAN IP Address

From the configuration tree in the left pane, select **System** to display the System screen for the IP Office Server Edition in the right pane. Select the **LAN1** tab, followed by the **LAN Settings** sub-tab in the right pane. Make a note of the IP Address (*e.g., 10.64.110.65*), which will be used in **Section 6.4** to configure MaxPlus Dispatch Console.



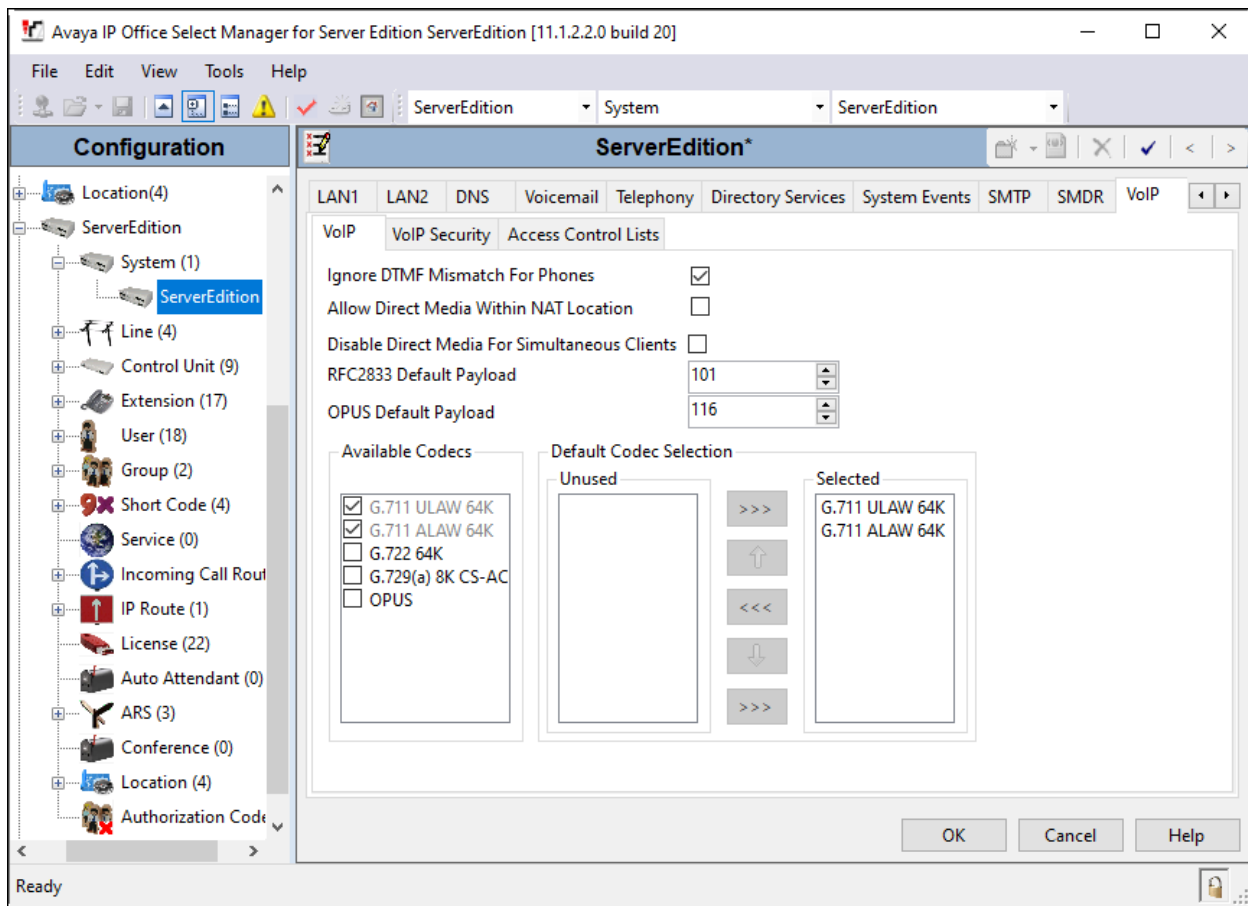
5.3. Administer SIP Registrar

Select the **VoIP** sub-tab in the **LAN1** tab. Ensure that **SIP Registrar Enable** is checked and enter a valid **Domain Name**. In the compliance testing, the **SIP Domain Name** field was set to *avaya.com*. TCP transport protocol was enabled for the **Layer 4 Protocol**, which was also used by MaxPlus Dispatch Console.



5.4. Administer IP Codecs

Select the **VoIP** tab in System. The **VoIP** sub-tab displays **Selected** codecs at the system level. *G.711 ULAW* and *G.711 ALAW* are selected. *G.729* can also be configured at the system level.



5.5. Administer SIP Extension for MaxPlus Dispatch Console

From the configuration tree in the left pane, right-click on **Extension** and select **New → SIP** from the pop-up list (not shown) to add a new SIP extension. Enter the desired extension for the **Base Extension** field as shown below. In this example, MaxPlus Dispatch Console was assigned extension **70010**. This is the extension that MaxPlus Dispatch Console will use to register with IP Office Server Edition. Enter an appropriate **Phone Password**. This will be used by MaxPlus Dispatch Console to register to IP Office Server.

The screenshot shows the Avaya IP Office Select Manager for Server Edition [11.1.2.2.0 build 20] window. The left pane displays the Configuration tree with the following structure:

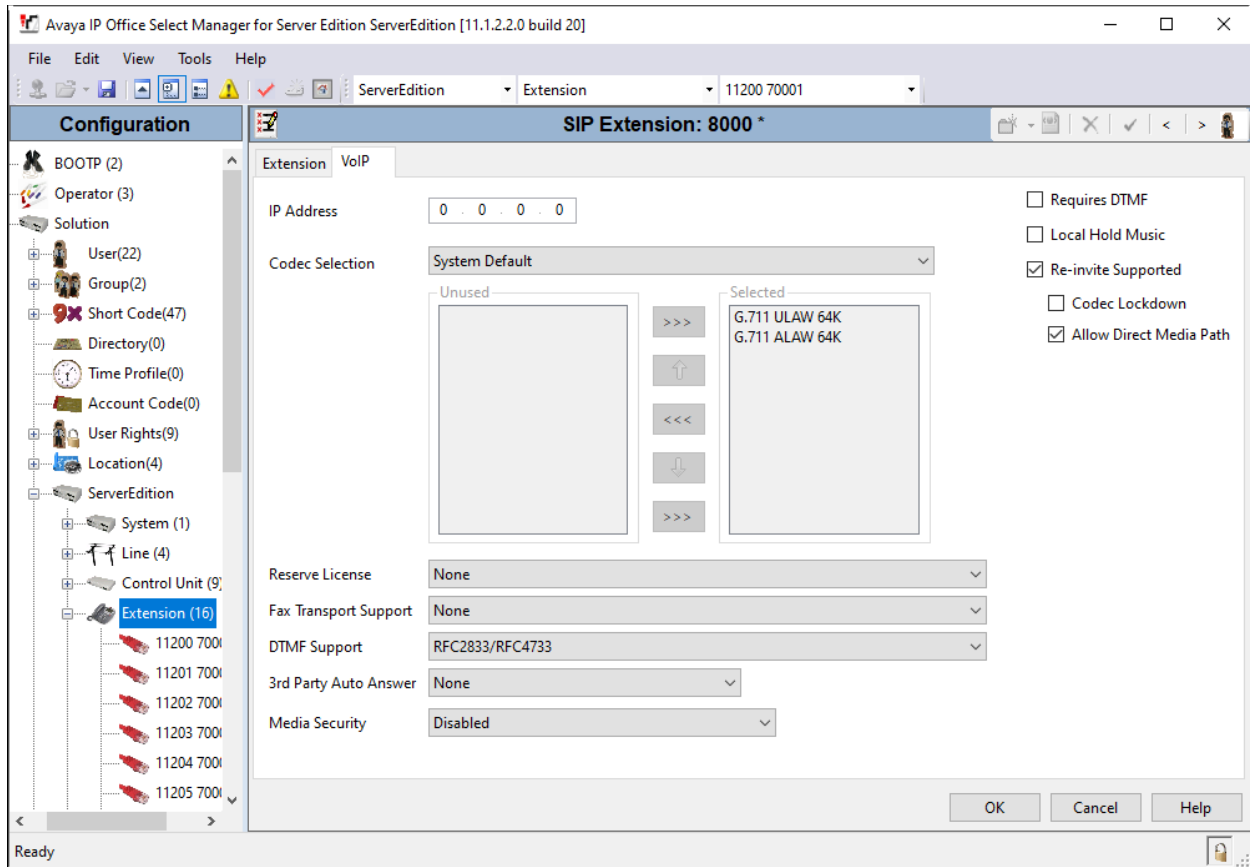
- BOOTP (2)
- Operator (3)
- Solution
 - User(22)
 - Group(2)
 - Short Code(47)
 - Directory(0)
 - Time Profile(0)
 - Account Code(0)
 - User Rights(9)
 - Location(4)
 - ServerEdition
 - System (1)
 - Line (4)
 - Control Unit (9)
 - Extension (16)**
 - 11200 70001
 - 11201 70002
 - 11202 70003

The right pane shows the configuration for the selected SIP Extension: 8000. The fields are as follows:

Field	Value
Extension ID	8000
Base Extension	70010
Phone Password	••••••
Confirm Phone Password	••••••
Caller Display Type	On
Reset Volume After Calls	<input type="checkbox"/>
Device Type	Unknown SIP device
Location	Automatic
Fallback As Remote Worker	Auto
Module	0
Port	0
Disable Speakerphone	<input type="checkbox"/>

At the bottom of the right pane are the buttons: OK, Cancel, and Help.

Select the **VoIP** tab. The **Codec Selection** is configured using the system level defaults from **Section 5.4** of *G.711 ULAW* and *G.711 ALAW*. Enable **Allow Direct Media Path** so that audio/RTP may flow directly between two SIP endpoints without using media resources in Avaya IP Office Server Edition. Select *disabled* for **Media Security**.



5.6. Administer SIP User for MaxPlus Dispatch Console

From the configuration tree in the left pane, right-click on **User** and select **New** from the pop-up list (not shown). Enter a value for the **Name** field (e.g., *Mindshare Cons2*). For the **Extension** field, enter the SIP extension administered in **Section 5.5** (e.g., *70010*).

Avaya IP Office Select Manager for Server Edition ServerEdition [11.1.2.2.0 build 20]

File Edit View Tools Help

ServerEdition User NoUser

Configuration

BOOTP (2)
Operator (3)
Solution
User (21)
Group(2)
Short Code(47)
Directory(0)
Time Profile(0)
Account Code(0)
User Rights(9)
Location(4)
ServerEdition
System (1)
Line (4)
Control Unit (9)
Extension (17)
User (17)
NoUser
72012 1100 L
72011 1608 L
72019 9641 L
71001 H323L

<User:0>: *

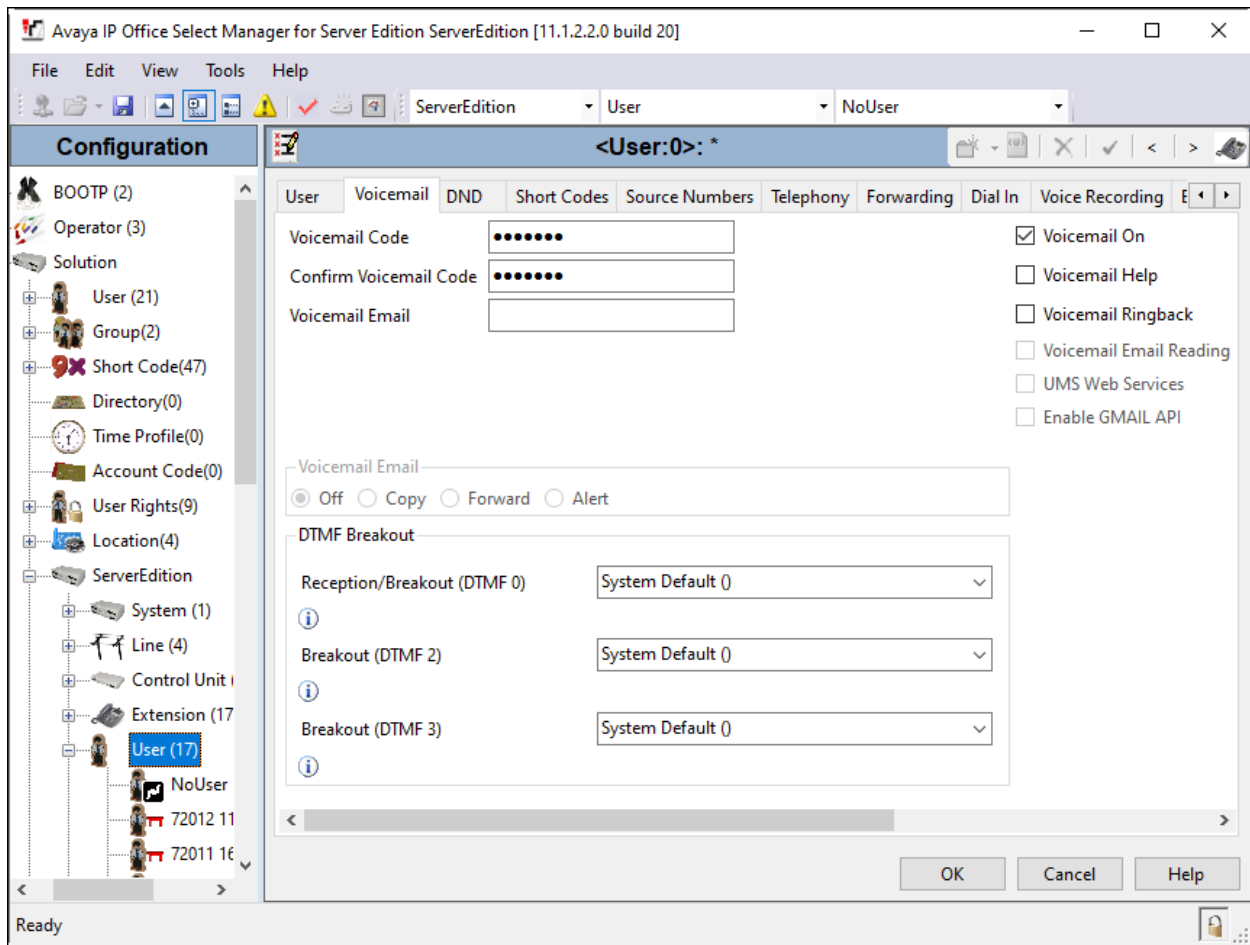
User Voicemail DND Short Codes Source Numbers Telephony Forwarding Dial In Voice Recording

Name Mindshare Cons2
Password
Confirm Password
Unique Identity
Conference PIN
Confirm Audio Conference PIN
Account Status Enabled
Full Name
Extension 70010
Email Address
Locale
Priority 5
System Phone Rights None
Profile Basic User
☐ Receptionist
☐ Enable Softphone
☐ Enable one-X Portal Services

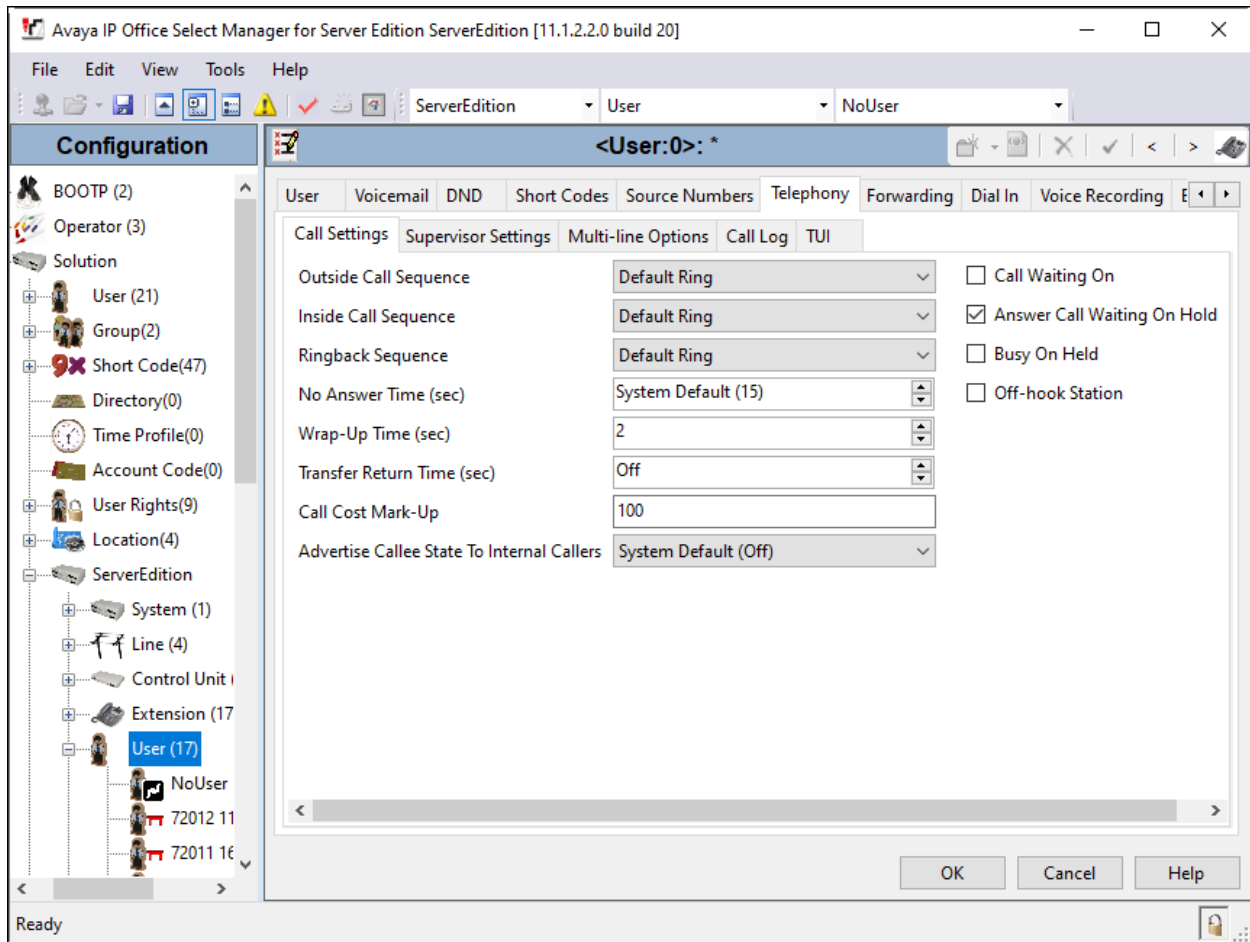
OK Cancel Help

Ready

Select the **Voicemail** tab and select **Voicemail On** to enable voicemail for MaxPlus Dispatch Console. Specify a **Voicemail Code** to be used when logging into voicemail.



Select the **Telephony** tab followed by the **Call Settings** sub-tab. Note the settings for the user.



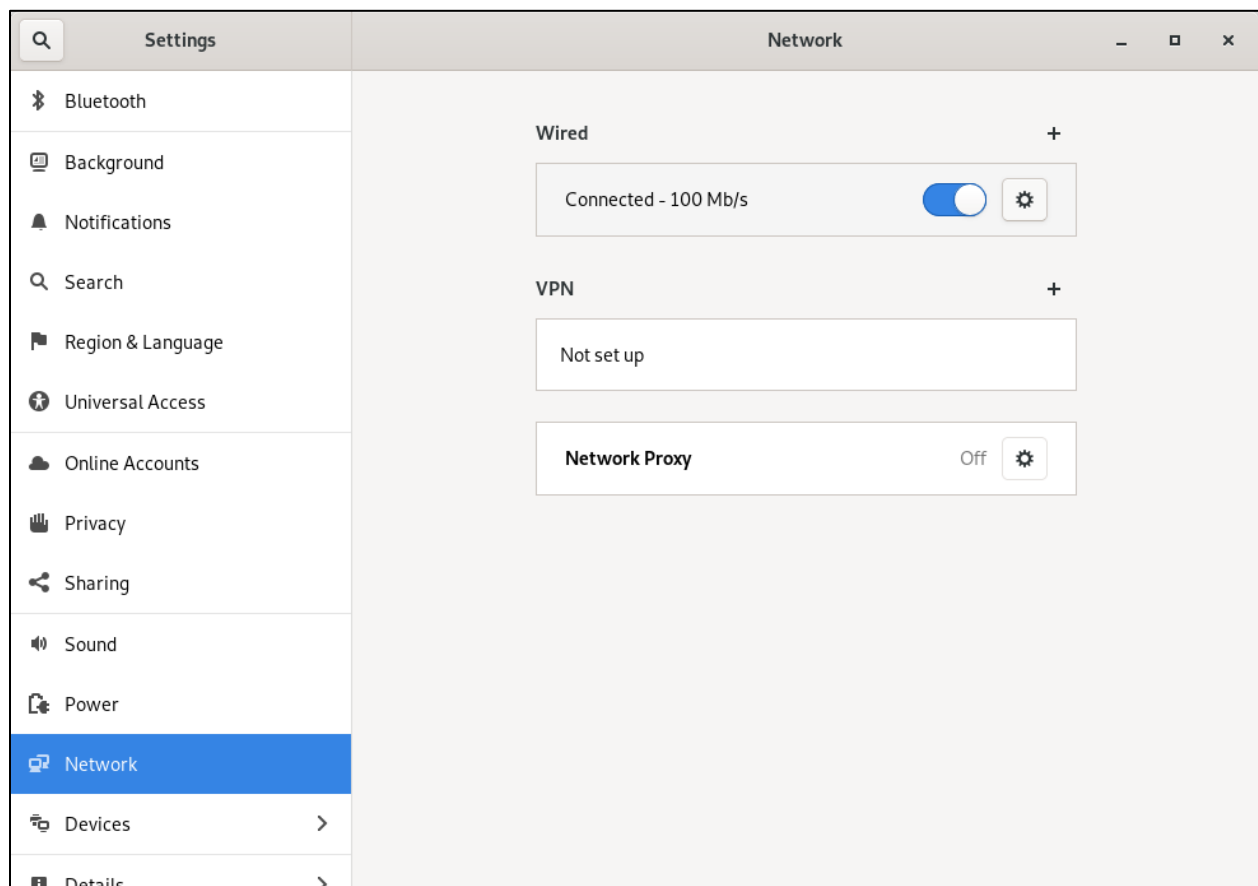
6. Configure CSS Mindshare 100500 MaxPlus Dispatch Console

This section covers MaxPlus Dispatch Console configuration using Console Builder . The procedure covers the following areas:

- Configure IP Address
- Launch Console Builder
- Configure Phone System Parameters
- Administer IP Comms
- Create Console Layout
- Save Layout to Configuration File

6.1. Configure IP Address

Note: MaxPlus Dispatch Console requires two IP addresses. The second IP address is internally assigned as the next numerical assignment, e.g., if a static address of *10.64.10.51* is assigned to the console, *10.64.10.52* is internally assigned. Static IP addresses can avoid address conflicts. MaxPlus Dispatch Console is configured for DHCP on power up. A static IP address can be assigned via the operating system desktop. Select the **Settings** button from the **System Menu** to open the Settings dialog. Click **Network** on the left side to access the **Wired** properties.



Click the **Wired** Settings button to the right of the **Connected** slider. After the **Wired** settings appear, click on **IPv4**. Assign an address by clicking the **Manual** radio button and input the appropriate network information. In this case, an internal IP address of *10.64.10.51* is assigned (and implicitly *10.64.10.52*).

Cancel

Wired

Apply

Details

Identity

IPv4

IPv6

Security

IPv4 Method

☐ Automatic (DHCP)

☐ Link-Local Only

☒ Manual

☐ Disable

Addresses

Address	Netmask	Gateway	
10.64.10.51	255.255.255.0	10.64.10.1	×
			×

DNS

Automatic ☒

10.64.110.100, 75.75.75.75

Separate IP addresses with commas

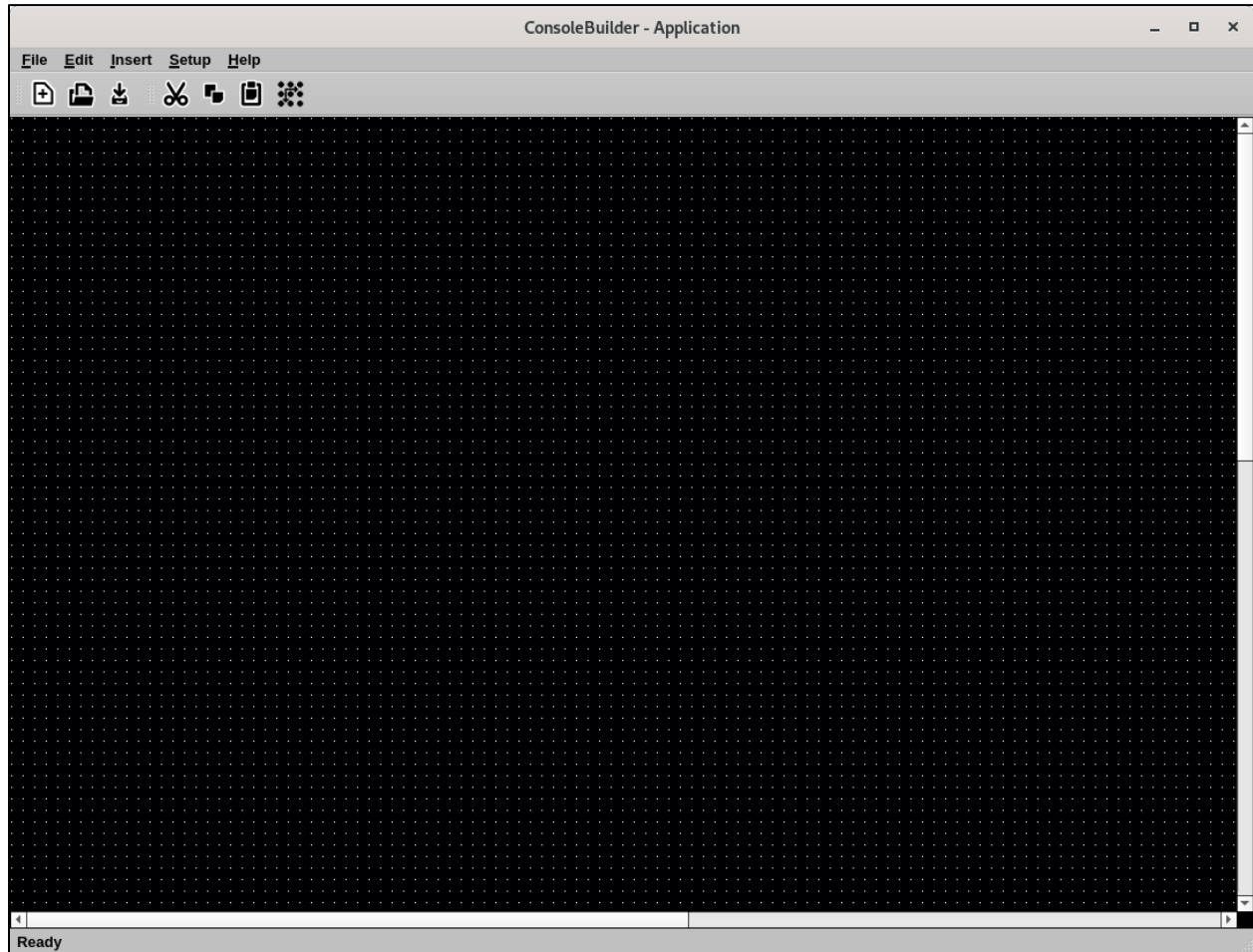
Routes

Automatic ☒

Address	Netmask	Gateway	Metric	
				×

6.2. Launch Console Builder

The Console user interface is configured using Console Builder, an application resident on the MaxPlus Dispatch Console environment. Launch the application from the Administrator account on the system through the **Applications → Mindshare → ConsoleBuilder** selection.



6.3. Configure Phone System Parameters

Select **Setup** → **Setup Phone System** from the menu. Input 20 hops for **Set SIP Packet Time to live**. Select the **TCP** checkbox. Set **SIP Time Before Retry** to 2000 ms. Clear any default digit mappings. Retain the default values in the remaining fields.

Setup Phone System Parameters

SIP Global Setup

SIP Packet Time to live:20hops

SIP Max Retry Count:3

SIP Time Before Retry:2000ms

SIP Registration Time:1800sec

SIP Local Port Number:5060

☒ Auto Hold

☒ TCP

Phone Line Tone Control Parameters

Guard Tone Frequency:2175Hz

Function F1 Frequency:1950Hz

Guard Tone Level:0dB

Function F2 Frequency:1850Hz

Guard Tone Duration:130ms

Function Tone Level:-10dB

Hold Tone Frequency:2175Hz

Function Tone Duration:40ms

Hold Tone Level:-20dB

Radio Tone Burst Interval:7sec

Phone Line Crosspatch VOX Parameters

VOX Trigger Level:-20dB

VOX Hangtime:3000ms

DTMF Signaling Parameters

DTMF Digit On Time:100ms

DTMF Flywheel:2000ms

DTMF Digit Off Time:100ms

DTMF Level:-10dB

DTMF Wait/Pause Time:500ms

RFC 2833 Flash Duration:1250ms

Phone Line Ringer Levels

All Lines OnHook:

One or more lines Offhook:

Ring Level:-8dB

Ring Level:-14dB

Speaker (1-8):1

Speaker (1-8):2

Speaker 1=Select, 2=Unselect1, 3=Unselect2, etc.

Digit Map:

Cancel

OK

6.4. Administer IP Comms

Select **Setup** → **Setup IP Comms** from the menu to administer lines on the console. Select *Phone* in the **Type** column. Input *70010* in the **Line Name** column. Select *uLaw* or *G.729* for the **Codec** column.

IP Address Setup Dialog
✕


	Type	Line Name	RX IP Address	RX Port	TX IP Address	TX Port	Delay	Max Buffer Size	TTL	TxMon	Codec	Channel Items	RX Block	Other Setup
1	Phone	70010	235.98.99.101	10001	235.98.99.101	12001	5	40	2	On	uLaw	Setup SIP	RX Block	Other
2	Off	70112	235.98.99.102	10002	235.98.99.102	12002	5	40	2	On	uLaw	None	RX Block	Other
3	Off	Line 3	235.98.99.103	10003	235.98.99.103	12003	5	40	2	On	uLaw	None	RX Block	Other
4	Off	Line 4	235.98.99.104	10004	235.98.99.104	12004	5	40	2	On	uLaw	None	RX Block	Other
5	Off	Line 5	235.98.99.105	10005	235.98.99.105	12005	5	40	2	On	uLaw	None	RX Block	Other
6	Off	Line 6	235.98.99.106	10006	235.98.99.106	12006	5	40	2	On	uLaw	None	RX Block	Other
7	Off	Line 7	235.98.99.107	10007	235.98.99.107	12007	5	40	2	On	uLaw	None	RX Block	Other
8	Off	Line 8	235.98.99.108	10008	235.98.99.108	12008	5	40	2	On	uLaw	None	RX Block	Other
9	Off	Line 9	235.98.99.109	10009	235.98.99.109	12009	5	40	2	On	uLaw	None	RX Block	Other
10	Off	Line 10	235.98.99.110	10010	235.98.99.110	12010	5	40	2	On	uLaw	None	RX Block	Other
11	Off	Line 11	235.98.99.111	10011	235.98.99.111	12011	5	40	2	On	uLaw	None	RX Block	Other
12	Off	Line 12	235.98.99.112	10012	235.98.99.112	12012	5	40	2	On	uLaw	None	RX Block	Other
13	Off	Line 13	235.98.99.113	10013	235.98.99.113	12013	5	40	2	On	uLaw	None	RX Block	Other
14	Off	Line 14	235.98.99.114	10014	235.98.99.114	12014	5	40	2	On	uLaw	None	RX Block	Other
15	Off	Line 15	235.98.99.115	10015	235.98.99.115	12015	5	40	2	On	uLaw	None	RX Block	Other
16	Off	Line 16	235.98.99.116	10016	235.98.99.116	12016	5	40	2	On	uLaw	None	RX Block	Other
17	Off	Line 17	235.98.99.117	10017	235.98.99.117	12017	5	40	2	On	uLaw	None	RX Block	Other
18	Off	Line 18	235.98.99.118	10018	235.98.99.118	12018	5	40	2	On	uLaw	None	RX Block	Other
19	Off	Line 19	235.98.99.119	10019	235.98.99.119	12019	5	40	2	On	uLaw	None	RX Block	Other
20	Off	Line 20	235.98.99.120	10020	235.98.99.120	12020	5	40	2	On	uLaw	None	RX Block	Other

Quality of Service Setting for all Lines:

Click **Setup SIP** in the **Channel Items** column to open the **Line SIP Setup** dialog. Enter the following:

- **SIP Display Name:** Enter a name, e.g., *Console2*.
- **SIP System User Name:** Enter the extension from **Section 5.5** e.g., *70010*.
- **SIP System Password:** Enter the password for the extension administered in **Section 5.5**.
- **SIP Server Address:** Enter IP Office Server Edition IP address from **Section 5.2**, e.g., *10.64.110.65*.
- **SIP Server Port Number:** Enter *5060*.
- **Backup SIP Server Port Number:** Enter *5060*.

Retain the default values in the remaining fields.



The image shows a 'Line SIP Setup' dialog box with a title bar containing a close button (X). The dialog contains several input fields and checkboxes. The 'SIP Display Name' field is filled with 'Console2'. The 'SIP System User Name' field is filled with '70010'. The 'SIP System Password' field is empty. Below these fields is a note: 'Leave the above three entries blank to have ConsoleExec prompt for them at startup.' The 'SIP Authorization Username' field is empty. The 'SIP Server Address' field is filled with '10.64.110.65'. The 'SIP Server Port Number' field is filled with '5060'. The 'Backup Server Address' field is empty. The 'Backup SIP Server Port Number' field is filled with '5060'. The 'SIP DTMF Mode' dropdown is set to 'RFC2833'. The 'Ringer Sound' dropdown is set to 'Ringer 1'. There are two checkboxes: 'Disable Ringer' and 'SLA Assist', both of which are unchecked. Below these checkboxes is a note: 'Check SLA Assist to populate SIP login and auto-dial strings based on SLA Console Number and SLA Line Number.' The 'SLA Console Number' field is filled with '1'. The 'SLA Line Number' field is filled with '1'. A 'Close' button is located at the bottom right of the dialog.

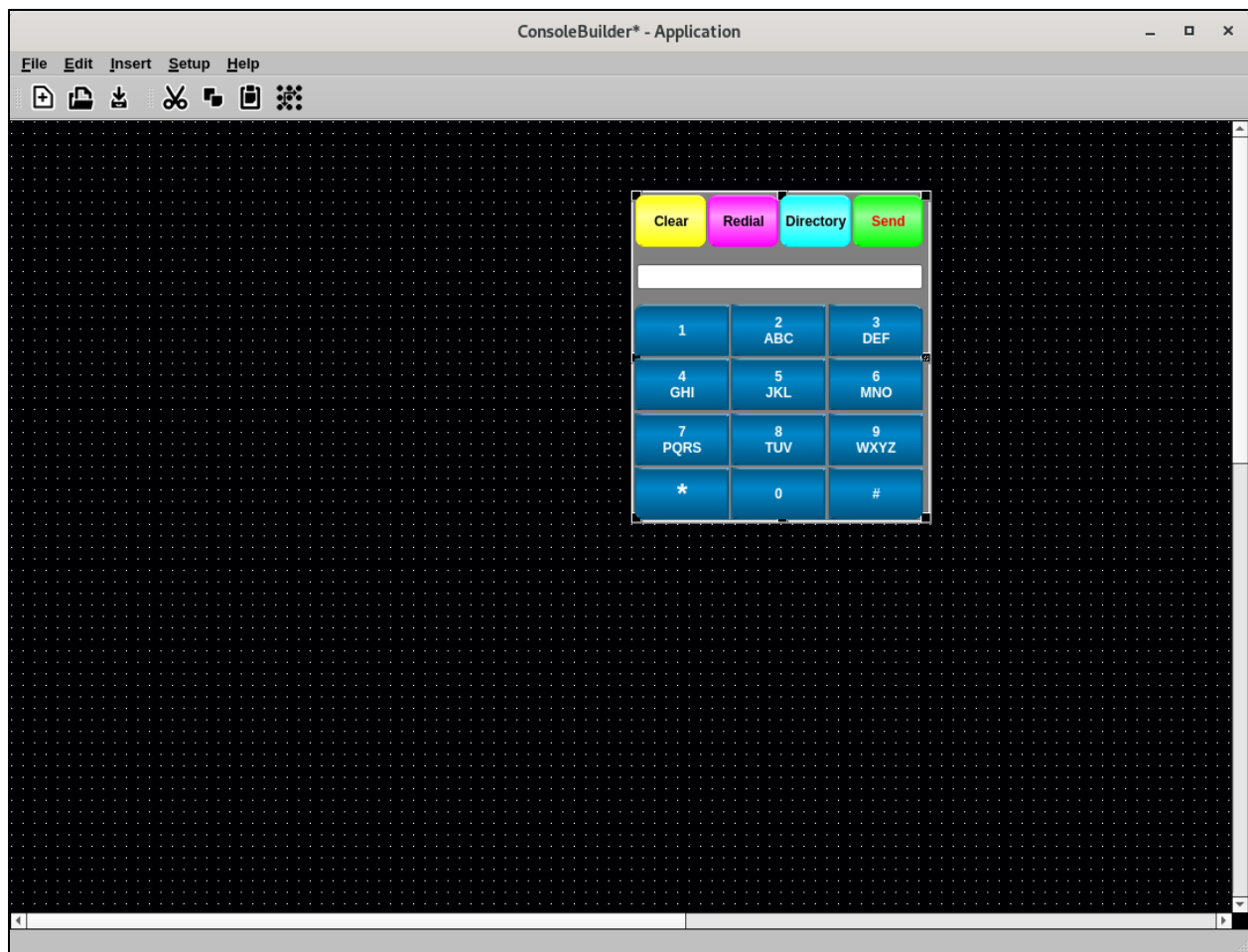
Line SIP Setup	
SIP Display Name:	Console2
SIP System User Name:	70010
SIP System Password:	
Leave the above three entries blank to have ConsoleExec prompt for them at startup.	
SIP Authorization Username:	
SIP Server Address:	10.64.110.65
SIP Server Port Number:	5060
Backup Server Address:	
Backup SIP Server Port Number:	5060
SIP DTMF Mode:	RFC2833
Ringer Sound:	Ringer 1
<input type="checkbox"/> Disable Ringer	
<input type="checkbox"/> SLA Assist	
Check SLA Assist to populate SIP login and auto-dial strings based on SLA Console Number and SLA Line Number.	
SLA Console Number:	1
SLA Line Number:	1
Close	

6.5. Create Console Layout

Insert and configure console elements used for the VoIP user. Elements include the Dial Keypad, Line Indicator, Phone Line, Close Console, and Feature buttons. Feature buttons employed for interoperability testing included Mute, Hold, Transfer, Redial, Call Pickup Any, Call Park, and Call Unpark buttons. Console builder button controls provide a set of User Interface Functions that are assigned to buttons. These tools can be employed to provide additional functionality. Basic controls and a possible Call Park and Call Pickup button configuration are illustrated. For other button implementation configurations consult refer to [2].

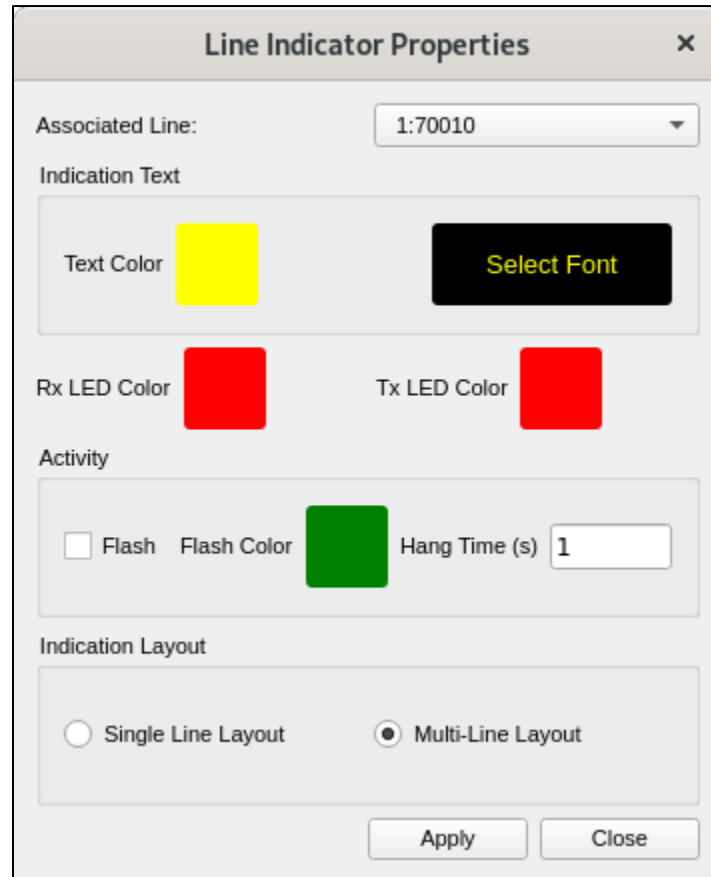
6.5.1. Dial Keypad

Select **Insert** → **Insert DTMF Keypad**. Adjust the size and position of the keypad on the grid.



6.5.2. Line Indicator

Select **Insert** → **Insert Line Indicator**. Adjust the size and position of the Line Indicator on the grid. Mouse over the Line Indicator and right click on **Properties**. Select the Line defined in **Section 6.4** for **Associated Line**, e.g., *70010*.



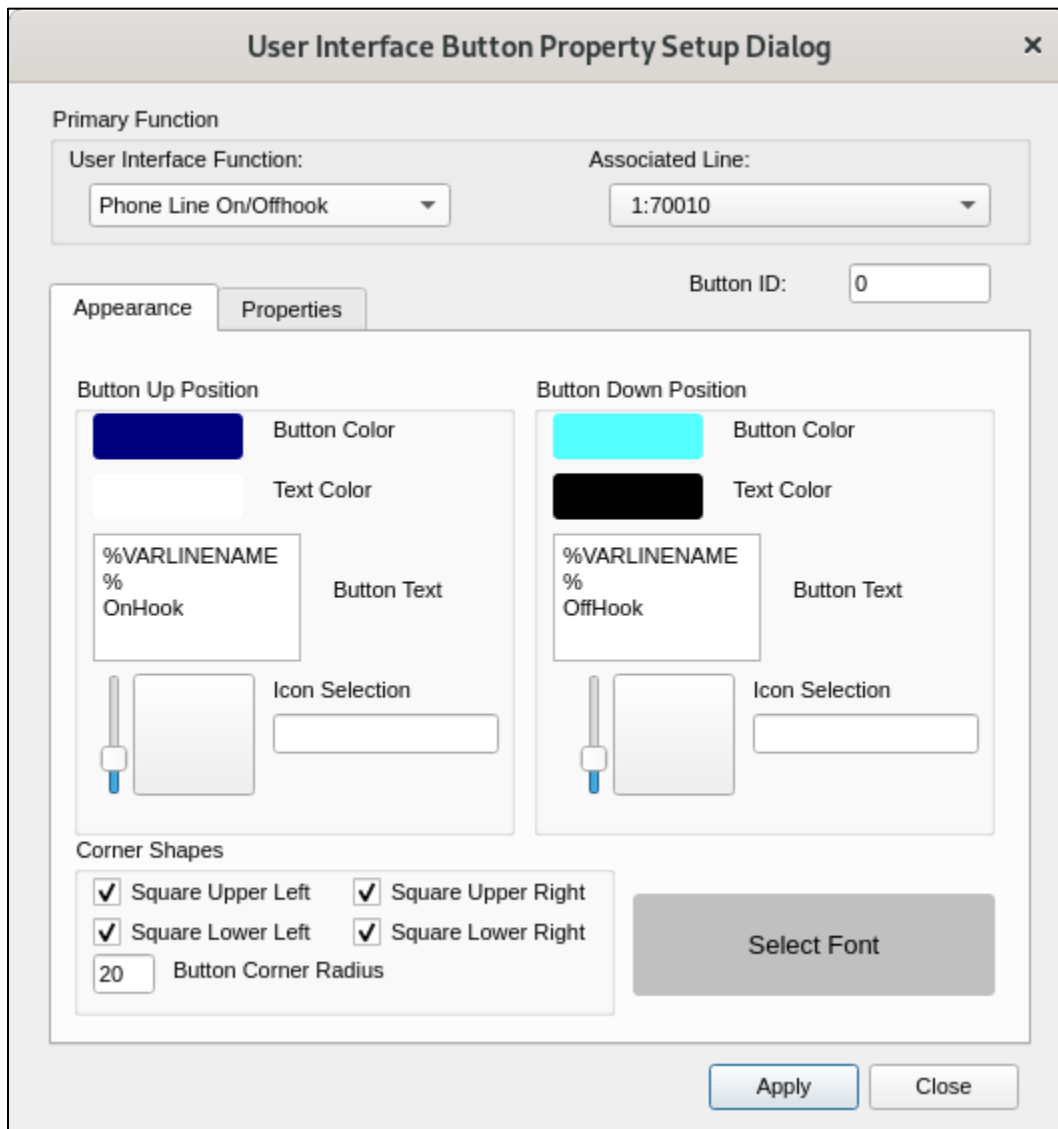
The image shows a 'Line Indicator Properties' dialog box with a close button (X) in the top right corner. The dialog is organized into several sections:

- Associated Line:** A dropdown menu showing '1:70010'.
- Indication Text:** A section containing a 'Text Color' color picker (yellow), a 'Select Font' button, and two red color swatches labeled 'Rx LED Color' and 'Tx LED Color'.
- Activity:** A section with a 'Flash' checkbox, a 'Flash Color' color picker (green), and a 'Hang Time (s)' input field set to '1'.
- Indication Layout:** A section with two radio buttons: 'Single Line Layout' and 'Multi-Line Layout' (which is selected).

At the bottom of the dialog are 'Apply' and 'Close' buttons.

6.5.3. Phone Line On/Offhook Button

Select **Insert → Insert Button Control**. Adjust the size and position of the On/Offhook button on the grid. Mouse over the new button and right click on **Properties**. Select the Line defined in **Section 6.4** for **Associated Line**, e.g., 70010. Select *Phone Line On/Offhook* for **User Interface Function**. Input `%VARLINENAME%[On/Off]Hook` for **Button Text** to display the line name of the **Associated Line** selected.

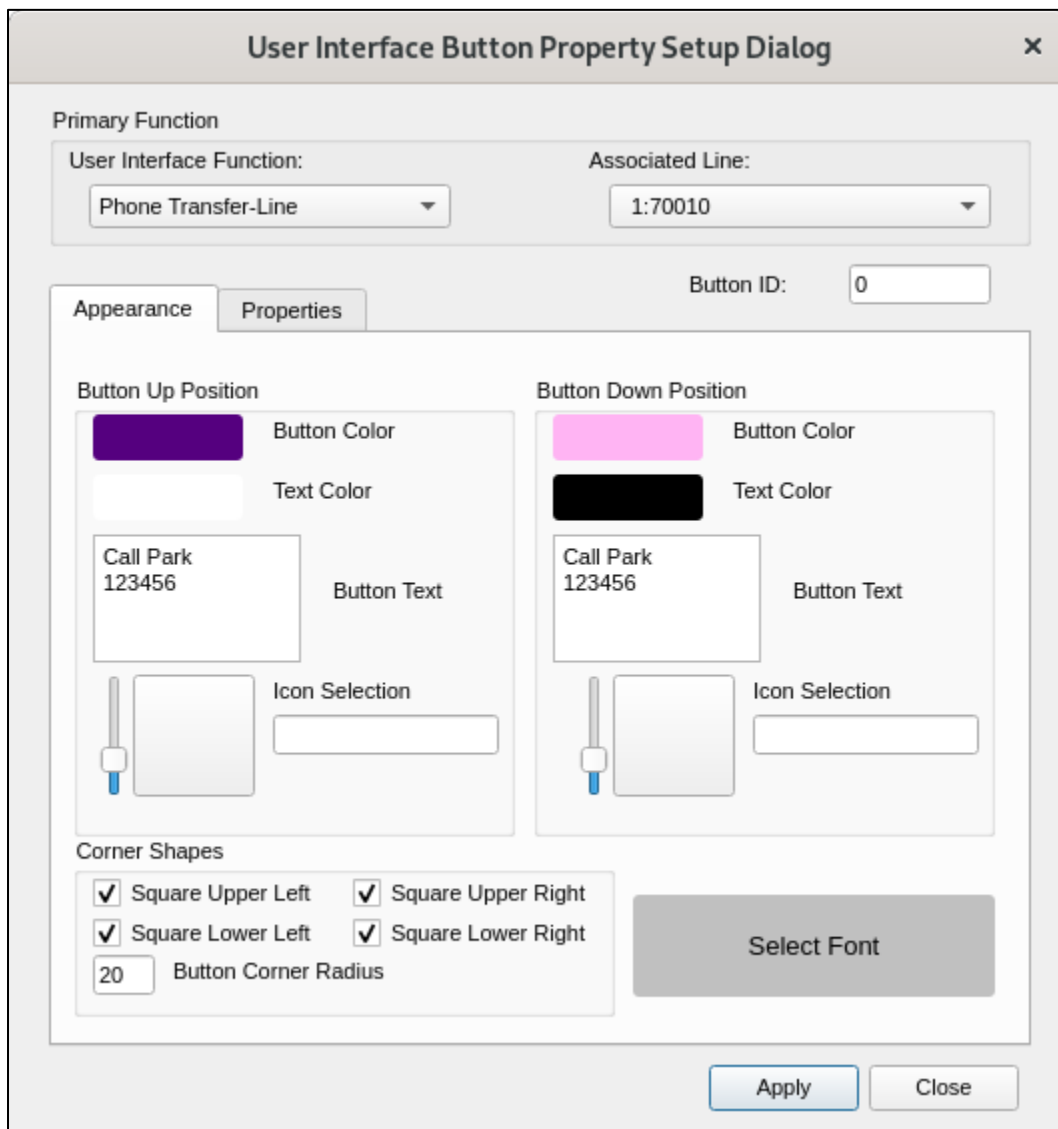


The dialog box is titled "User Interface Button Property Setup Dialog" and features a close button (X) in the top right corner. It is divided into several sections:

- Primary Function:** Contains two dropdown menus. "User Interface Function:" is set to "Phone Line On/Offhook". "Associated Line:" is set to "1:70010".
- Button ID:** A text field containing the value "0".
- Appearance and Properties tabs:** The "Appearance" tab is selected. It contains two main panels for "Button Up Position" and "Button Down Position".
 - Button Up Position:** Includes a "Button Color" selector (dark blue), a "Text Color" selector (white), a "Button Text" field containing "%VARLINENAME% OnHook", and an "Icon Selection" area with a vertical slider and a square icon.
 - Button Down Position:** Includes a "Button Color" selector (cyan), a "Text Color" selector (black), a "Button Text" field containing "%VARLINENAME% OffHook", and an "Icon Selection" area with a vertical slider and a square icon.
- Corner Shapes:** A section with four checkboxes: "Square Upper Left", "Square Upper Right", "Square Lower Left", and "Square Lower Right", all of which are checked. Below these is a "Button Corner Radius" field set to "20".
- Select Font:** A large grey button.
- Buttons:** "Apply" and "Close" buttons are located at the bottom right.

6.5.4. Call Park Button

Select **Insert → Insert Button Control**. Adjust the size and position of the Call Park button on the grid. Mouse over the new button and right click on **Properties**. Select the Line defined in **Section 6.4** for **Associated Line**, e.g., *70010*. Select *Phone Transfer-Line* for **User Interface Function**. Input appropriate **Button Text**.



The dialog box is titled "User Interface Button Property Setup Dialog" and features a close button (X) in the top right corner. It is divided into two main sections: "Primary Function" and "Appearance".

Primary Function:

- User Interface Function:** A dropdown menu with "Phone Transfer-Line" selected.
- Associated Line:** A dropdown menu with "1:70010" selected.
- Button ID:** A text field containing "0".

Appearance:

The "Appearance" section has two tabs: "Appearance" (selected) and "Properties".

Button Up Position:

- Button Color:** A color swatch showing purple.
- Text Color:** A color swatch showing white.
- Button Text:** A text field containing "Call Park 123456".
- Icon Selection:** A vertical slider and a square icon placeholder.

Button Down Position:

- Button Color:** A color swatch showing pink.
- Text Color:** A color swatch showing black.
- Button Text:** A text field containing "Call Park 123456".
- Icon Selection:** A vertical slider and a square icon placeholder.

Corner Shapes:

- ☒ Square Upper Left
- ☒ Square Upper Right
- ☒ Square Lower Left
- ☒ Square Lower Right
- Button Corner Radius:** A text field containing "20".

Select Font: A button labeled "Select Font".

Buttons: "Apply" and "Close" buttons are located at the bottom right.

Select the **Properties** tab. Check **Enable Autodial**. Input the Short Code assigned to Call Park on IP Office, e.g., *37*123456#. In this case, the park slot number used is 123456.

User Interface Button Property Setup Dialog ✕

Primary Function

User Interface Function:

Phone Transfer-Line
▼

Associated Line:

1:70010
▼

Appearance | **Properties**

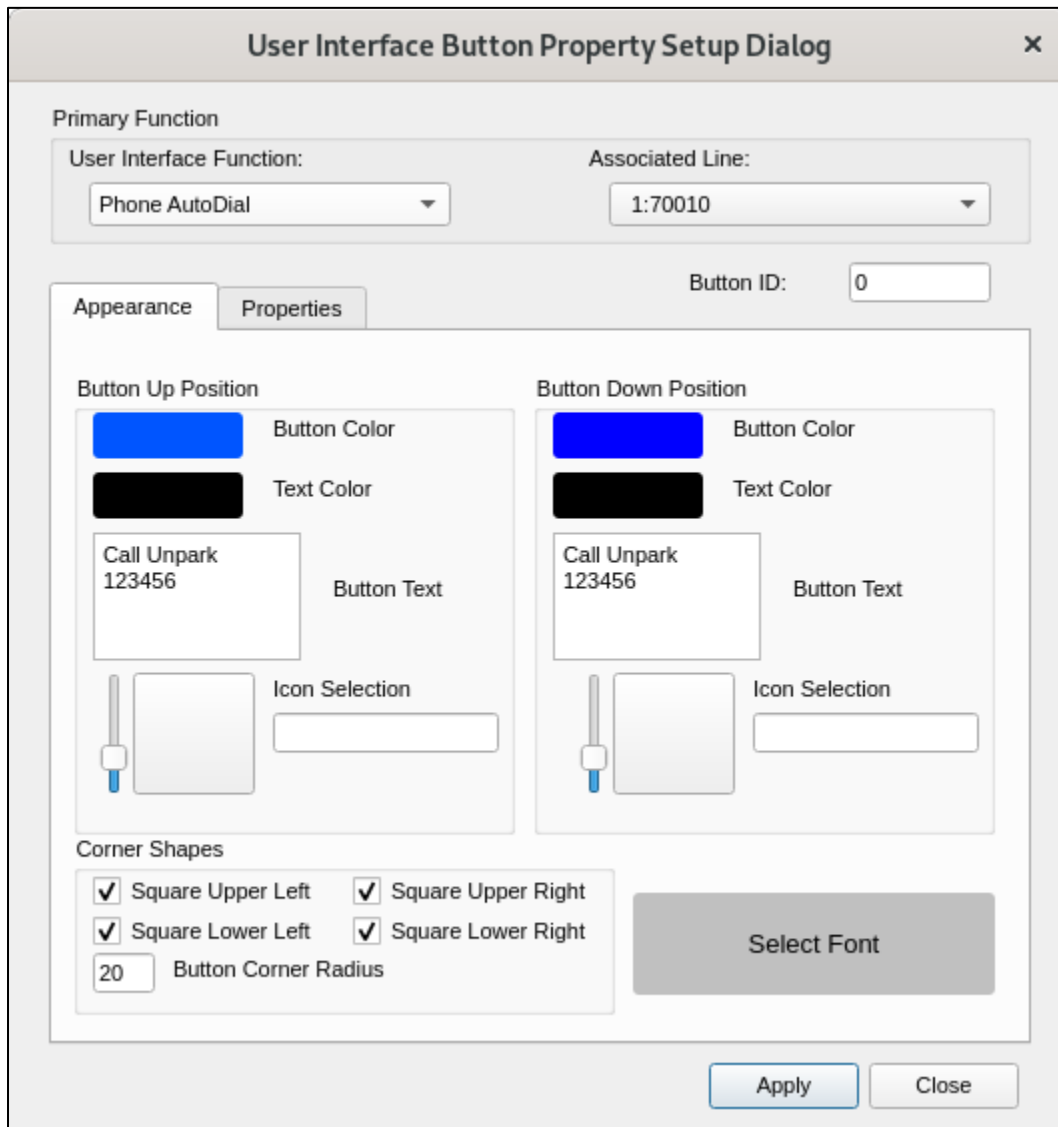
Button ID: 0

#	Property	Value(s)	Unit(s)
1	Enable Autodial:	<input checked="" type="checkbox"/>	
2	Number:	*37*123456#	
3	Transfer Blind:	<input type="checkbox"/>	
4	Preset Dial String:	<input type="checkbox"/>	
5	Popup Dialpad on Click:	<input type="checkbox"/>	
6			
7			
8			
9			
10			
11			
12			
13			
14			
15			
16			

Apply
Close

6.5.5. Call Unpark Button

Select **Insert → Insert Button Control**. Adjust the size and position of the Call Unpark button on the grid. Mouse over the new button and click properties. Select the Line defined in **Section 6.4** for **Associated Line**, e.g., *70010*. Select *Phone AutoDial* for **User Interface Function**. Input appropriate **Button Text**.



The dialog box is titled "User Interface Button Property Setup Dialog" and features a close button (X) in the top right corner. It is divided into two main sections: "Primary Function" and "Appearance".

Primary Function:

- User Interface Function:** A dropdown menu with "Phone AutoDial" selected.
- Associated Line:** A dropdown menu with "1:70010" selected.
- Button ID:** A text field containing "0".

Appearance:

The "Appearance" section has two tabs: "Appearance" (selected) and "Properties". It is further divided into "Button Up Position" and "Button Down Position" panels.

Button Up Position:

- Button Color:** A blue color swatch.
- Text Color:** A black color swatch.
- Button Text:** A text field containing "Call Unpark 123456".
- Icon Selection:** A vertical slider and a square icon placeholder.

Button Down Position:

- Button Color:** A blue color swatch.
- Text Color:** A black color swatch.
- Button Text:** A text field containing "Call Unpark 123456".
- Icon Selection:** A vertical slider and a square icon placeholder.

Corner Shapes:

- ☒ Square Upper Left
- ☒ Square Upper Right
- ☒ Square Lower Left
- ☒ Square Lower Right
- Button Corner Radius:** A text field containing "20".

Select Font: A button to open the font selection dialog.

Buttons: "Apply" and "Close" buttons are located at the bottom right of the dialog.

Select the **Properties** tab. Check **Dial on Associated Line**. Input the short Code to unpark the call on IP Office for **Dial String when Clicked**, e.g., *38*123456# that uses the analogous park slot number 123456 used in **Section 6.5.4**.

User Interface Button Property Setup Dialog ✕

Primary Function

User Interface Function: Phone AutoDial Associated Line: 1:70010

Button ID: 0

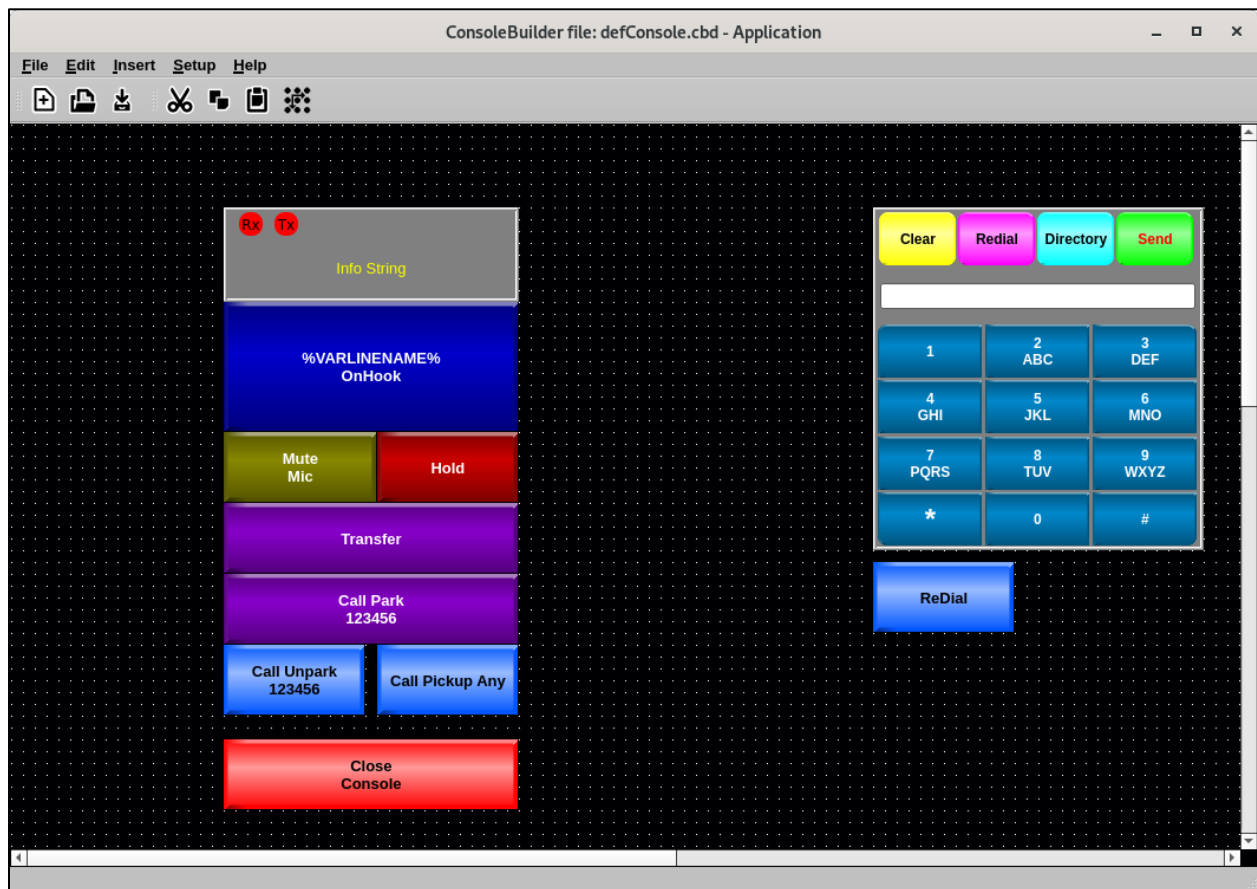
Appearance **Properties**

Property	Value(s)	Unit(s)
1	Dial String when Clicked: *38*123456#	
2	Preset Dial String: <input type="checkbox"/>	
3	Preset String: <input type="text"/>	
4	Popup Dialpad on Click: <input type="checkbox"/>	
5	Dial on Associated Line: <input checked="" type="checkbox"/>	
6	Enable Autodial #2: <input type="checkbox"/>	
7	Enable Preset Dial #2: <input type="checkbox"/>	
8	Number 2: <input type="text"/>	
9		
10		
11		
12		
13		
14		
15		
16		

Apply
Close

6.6. Save Layout to Configuration File

When the layout is complete, select **File → Save As** to save the layout configuration. The configuration file should be saved as **/opt/mindshare/console suite/defConsole.cbd**.

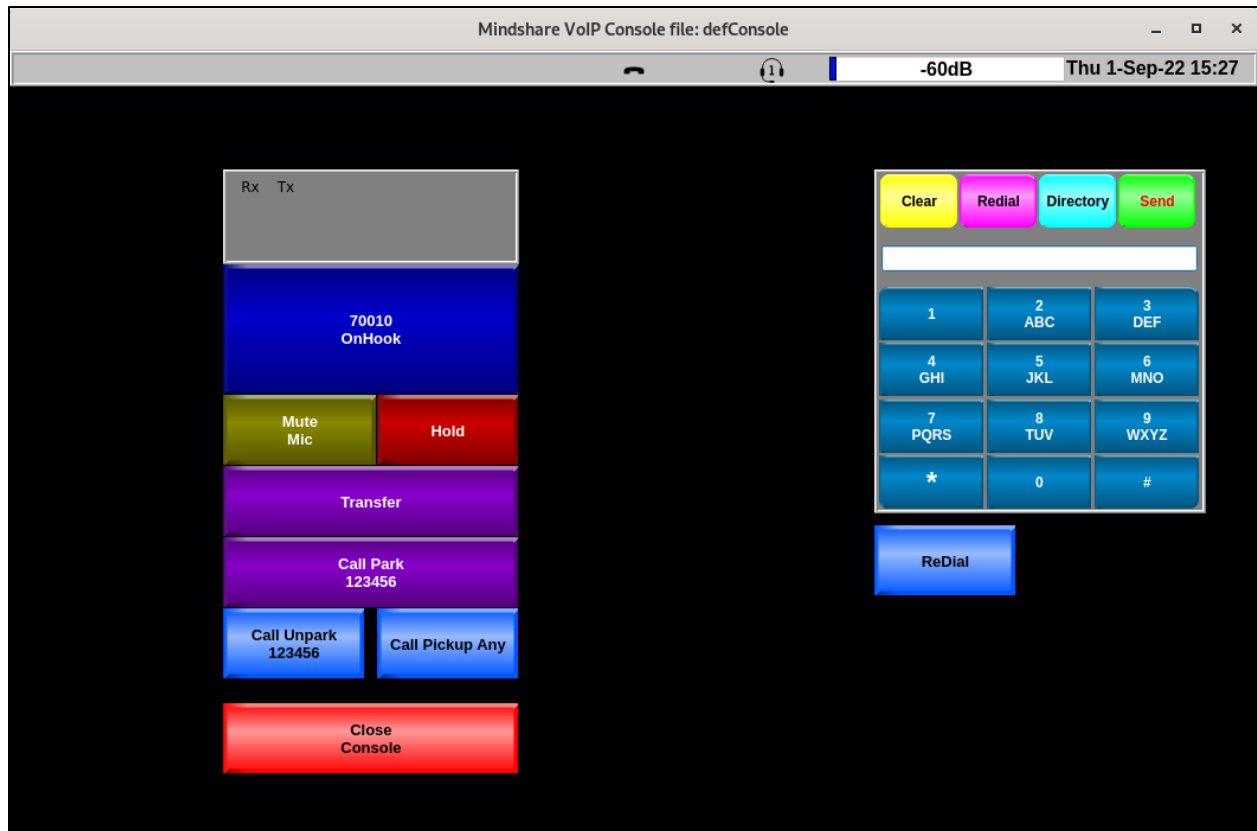


7. Verification Steps

This section provides the tests that can be performed to verify proper configuration of MaxPlus Dispatch Console with IP Office.

7.1. Launch Console

Launch the application from the Administrator account on the system through the **Applications** → **Mindshare** → **ConsoleBuilder** menu selection.



7.2. Registration Status

Verify that MaxPlus Dispatch Console has successfully registered with IP Office. From a PC with **IP Office Admin Suite** installed, invoke **IP Office System Status**, navigate to the MaxPlus Dispatch Console SIP extension and verify **Media Stream** is set to *RTP*, **Layer 4 Protocol** is set to *TCP*, and **Current State** is shown as *Idle*.

The screenshot shows the Avaya IP Office System Status application window. The title bar reads "Avaya IP Office System Status - ServerEdition (10.64.110.65) - IP Office Linux PC 11.1.2.2.0 build 20". The Avaya logo is in the top left, and the window title "IP Office System Status" is in the top center. A menu bar contains "Help", "Snapshot", "LogOff", "Exit", and "About".

On the left is a navigation tree with the following items: "System", "Alarms (3)", "Extensions (3)" (expanded), "70010" (selected), "72016", "72019", "Trunks (4)", "Active Calls", "Resources", "Voicemail", and "IP Networking Locations".

The main area displays the "Extension Status" for extension 70010. The status is as follows:

Extension Number:	70010
IP address:	10.64.10.51
Standard Location:	None
Registrar:	Primary
Telephone Type:	Unknown SIP Device
User-Agent SIP header:	ConsoleExec_v3.27.2-release_by_Mindshare
Media Stream:	RTP
Layer 4 Protocol:	TCP
Current User Extension Number:	70010
Current User Name:	Mindshare Cons2
Forwarding:	Off
Twinning:	Off
Do Not Disturb:	Off
Message Waiting:	Off
Number of New Messages:	0
Phone Manager Type:	None
SIP Device Features:	REFER
License Reserved:	No
Last Date and Time License Allocated:	9/6/2022 10:13:24 AM
DTMF Required:	No
Packet Loss Fraction:	
Jitter:	
Round Trip Delay:	
Connection Type:	
Codec:	
Remote Media Address:	

Below the status list is a table showing call details:

Call Ref	Current State	Time in State	Calling Number or Called Number	Direction	Other Party on Call
	Idle	00:00:32			

At the bottom of the main area are buttons: "Trace", "Trace All", "Pause", "Ping", "Call Details", "Print...", and "Save As...".

The bottom status bar shows the time "1:13:29 PM", the status "Online", and a lock icon.

7.3. Basic Calls

Establish a call between MaxPlus Dispatch Console and a local Avaya SIP desk phone. In **IP Office System Status**, navigate to the MaxPlus Dispatch Console SIP extension and verify that the **Current State** is *Connected* as shown. Verify two-way audio.

The screenshot shows the Avaya IP Office System Status application window. The title bar indicates the server edition (10.64.110.65) and the IP Office Linux PC version (11.1.2.2.0 build 20). The Avaya logo is prominently displayed. The main menu includes Help, Snapshot, LogOff, Exit, and About. The left sidebar contains a tree view with categories: System, Alarms (3), Extensions (3), Trunks (4), Active Calls, Resources, Voicemail, and IP Networking. The 'Extensions (3)' category is expanded, showing a list of extensions: 70010, 72016, and 72019. The '70010' extension is selected, and its status is displayed in the main pane. The status is 'Connected'. Below the status, a table shows call details for extension 610, which is currently 'Connected' and has been in that state for 00:00:25. The table also shows the calling number, direction, and other party on call. The bottom of the window features a status bar with the time (11:34:55 AM) and the system status (Online).

Avaya IP Office System Status - ServerEdition (10.64.110.65) - IP Office Linux PC 11.1.2.2.0 build 20

AVAYA **IP Office System Status**

Help Snapshot LogOff Exit About

System
Alarms (3)
Extensions (3)
 70010
 72016
 72019
Trunks (4)
Active Calls
Resources
Voicemail
IP Networking
 Locations

Extension Status

Extension Number: 70010
IP address: 10.64.10.51
Standard Location: None
Registrar: Primary
Telephone Type: Unknown SIP Device
User-Agent SIP header: ConsoleExec_v3.27.2-release_by_Mindshare
Media Stream: RTP
Layer 4 Protocol: TCP
Current User Extension Number: 70010
Current User Name: Mindshare Cons2
Forwarding: Off
Twinning: Off
Do Not Disturb: Off
Message Waiting: Off
Number of New Messages: 0
Phone Manager Type: None
SIP Device Features: REFER
License Reserved: No
Last Date and Time License Allocated: 9/6/2022 10:13:24 AM
DTMF Required: No
Packet Loss Fraction: 0%
Jitter: 0ms
Round Trip Delay: 0ms
Connection Type: VCM (SRTP)
Codec: G711 Mu
Remote Media Address: 10.64.10.225

Call Ref	Current State	Time in State	Calling Number or Called Number	Direction	Other Party on Call
610	Connected	00:00:25		Outgoing	Extn 72019, 9641 H323 User

Trace Trace All Pause Ping Call Details Print... Save As...

11:34:55 AM Online

8. Conclusion

These Application Notes have described the administration steps required to integrate CSS Mindshare MaxPlus 100500 Dispatch Console 3.27.2 with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500 V2 Expansion System 11.1. CSS Mindshare 100500 MaxPlus Dispatch Console successfully registered with IP Office as a SIP user, and basic and supplementary telephony features were verified. All test cases passed with observations noted in **Section 2.2**.

9. Additional References

This section references product documentation relevant to these Application Notes. The following Avaya product documentation is available online at support.avaya.com.

[1] *Administering Avaya IP Office Platform™ with Manager*

The following CSS Mindshare product documentation is accessible to registered users at customer.css-mindshare.com.

[2] *MS0101_UM_ConsoleApplicationManual*, Revision 1.15, June 23, 2022

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