1 Installing the Codec at the Studio

Studio IP Streaming Setup for Tieline Audio Codecs

The following instructions are intended to help you configure your internet connection and Tieline codecs at the studio to enable incoming calls over the internet from a remote Tieline codec. It is assumed that you have a basic understanding of your IP network and how to configure IP devices. If you have limited IT network knowledge, we recommend you engage the services of an IT professional to install the public IP address and perform the Network Address Translation (NAT) and port forwarding between the public internet and your private Local Area Network (LAN) at the studio.

Prerequisites

The following procedures are valid for:

- All firmware versions in Gateway and Gateway-4 codecs.
- All firmware versions in the Genie and Merlin codec families.
- All Bridge-IT and Bridge-IT XTRA codecs with firmware release v.2.x or higher.
- · All Commander G3 and i-Mix G3 codecs.

Getting Started at the Studio

To perform a typical codec installation at the studio you will need to:

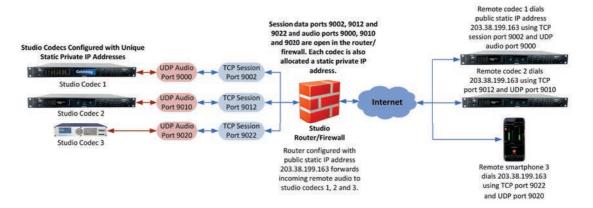
- Contact your Internet Service Provider and organize a dedicated high speed broadband connection at the studio for your codec with a public static IP address. Do not share this connection with other devices.
- 2. Install your codec at the studio and attach an active RJ-45 LAN cable to the "LAN" or "Ethernet" port on the rear of the codec. Please note:
 - The green LED underneath the "LAN" or "Ethernet" port will illuminate and the orange LED will flash steadily if you are connected to an active LAN connection.
 - Many Tieline IP codecs support two simultaneous Ethernet connections.
- 3. If you are connecting a single codec to a router without a firewall you can enter the public IP address, Subnet Mask and Gateway directly into the codec and your work is done. Note: your Telco should be able to provide this information.
- 4. Alternatively, if you are connected to a router with a firewall, configure Network Address Translation (NAT) in your router. NAT is performed between the public internet and your private Local Area Network (LAN) by your router. Your remote codec sends IP data packets to the studio router's public static IP address and the router performs NAT, which forwards these data packets to the private IP address allocated by the router to your codec. As part of this process we recommended you:
 - Connect to your router using a web-browser.
 - Configure it to allocate a static private IP address for each codec.



Important Note: The IP address may change if the codec is allocated a DHCP IP address by the router and it loses power or is temporarily disconnected from the LAN. This will cause problems for remote codecs attempting to dial and connect.

- 5. Ensure your router's firewall is configured with the relevant TCP and UDP IP ports open to allow data traffic between your codec and the remote codec. The process is fairly simple if you use the following procedure:
 - a. Connect to your router using a web-browser.
 - b. Navigate to https://portforward.com/router.htm
 - c. Select your router manufacturer from the list.

- d. Next, select your router model from the list.
- e. Follow the instructions to complete port forwarding.
- 6. Visit www.portforward.com and download the port checking application to verify your router's ports are open, or use a port checker like the one available at https://www.ipfingerprints.com/portscan.php.
- 7. Configure the static IP address in your codec using the instructions in the next section. To allow multiple codecs to share a single public static IP address behind a firewall and route the calls correctly, your codecs and the firewall need to be configured similarly to the example diagram which follows. Ensure the port, IP address, Subnet Mask and Gateway settings in your codecs match those configured in your router.



Port Forwarding to 3 Studio Codecs Sharing a Public Static IP Address



Important Note:

- The most common studio configuration issue is a firewall which blocks the incoming and/or outgoing TCP and UDP ports, or not configuring NAT and port forwarding correctly. The following table lists the firewall ports you need to open for each model of Tieline codec if they are dialing your router at the studio. If the remote codec is also connected to a LAN with a firewall you may also need to open the ports at the remote end of the link to connect successfully.
- Some firewalls require symmetric port configuration. The codec supports configuration of the "send" audio port (codec port at the remote end of the link to which you are sending audio) and "return" audio port (port used by the local codec to receive audio from the remote codec).

			Firew	all Ports	S					
Commander Bridge-IT /		Merlin and Genie		ViA Codec		Gateway /		Cloud		
		Bridge-I	Bridge-IT XTRA		Codec Families				vay-4	Codec
i-Mix G3								Codecs		Controller
TCP	UDP	TCP	UDP	TCP	UDP	TCP	UDP	TCP	UDP	TCP
IP1	IP1	Session	Audio	Session		I	Audio Port			HTTP
Session	Audio	Port	(Proto):		Port	Port:	Stream 1:	S	ee	80
Port:	Port:	(Sess):	9000	Port:	Stream 1:	9002	9000	Та	ble	
9002	9000	9002	O.D.	9002	9000	 	A 11 D 1	Be	low	LITTOO
IP2 Session	IP2 Audio	Web- GUI:	SIP	Alternati	Audio Port		Audio Port Stream 2:			HTTPS 443
Port:	Port:	80 80	Session:		Stream 2:	ve Session	9010			443
9012	9010	00	5060		9010	: 9012	3010			
0012	0010		0000	9012	0010	.0012				
Toolbox	Toolbox	Alternativ	SIP	Web-	Audio	Web-	SIP			
Software:		е	Audio:	GUI:	Port	GUI:	Session:			
5550	e: 5550	Session:	5004	80	Stream 3:	80	5060			
		9012			9020					
		Alternativ					SIP Audio:			
	Sessio		8999	ve Web-	1	ve Web-	5004-			
	n: 5060			I	Stream 4:	1	5054			
	SIP	8080 TLS/SSL		8080 TLS/SSL	9030	8080 TLS/SS	Fuse-IP			
	Audio:	443		11LS/SSL 443	Audio Port	115/55	8999			
	5004	443		443	Stream 5:	_	0999			
	0001				9040	''				
					Audio					
					Port					
					Stream 6:					
					9050					
					SIP				SIP	
					Session:				Session	
					5060				: 5060	
					SIP Audio:				SIP Audio:	
					5004-				5004-	
					5054				5054	
					Fuse-IP				Fuse-IP	
					8999				8999	

Gateway 16 Channel Default Ports

Audio Stream	TCP Session Port	UDP Audio Port	SmartStream + 1 (UDP)	SmartStream + 2 (UDP)	SmartStream + 3 (UDP)
1	9002	9000	9001	9002	9003
2	9002	9010	9011	9012	9013
3	9002	9020	9021	9022	9023
4	9002	9030	9031	9032	9033
5	9002	9040	9041	9042	9043
6	9002	9050	9051	9052	9053
7	9002	9060	9061	9062	9063
8	9002	9070	9071	9072	9073
9	9002	9080	9081	9082	9083
10	9002	9090	9091	9092	9093
11	9002	9100	9101	9102	9103
12	9002	9110	9111	9112	9113

13	9002	9120	9121	9122	9123
14	9002	9130	9131	9132	9133
15	9002	9140	9141	9142	9143
16	9002	9150	9151	9152	9153

Configure a Static IPv4 Address in Gateway

Static IP addresses are fixed addresses which are recommended for studio installations. Using a static IP address ensures remote codecs can connect reliably using the same IP address over time.

- 1. Press the **SETTINGS** button.
- 2. Select **Transports** and press the

 substitute button.
- 3. Use the down navigation button to select LAN1, LAN2 or a VLAN interface.
- 4. Select **Configuration** and then **Usage** and then the appropriate control and/or streaming mode for the connection. Next, press the substant.
- 5. Select IPv4 Mode and press the wobutton.
- 6. Select **Static** and press the would button.
- 7. Navigate to **IPv4 Address** and enter the IP address (using the keypad or onscreen keyboard), then press the button (or select **Enter** using the onscreen keyboard).
- 8. Navigate to IPv4 Subnet and enter the Subnet Mask, then press the subtrant
- 9. Navigate to IPv4 Gateway and enter the Gateway details, then press the solution.
- 10. Press the **Return** button, then select **Save** in the confirmation dialog and press the button to confirm the new settings.
- 11. Check the **Unit Details** menu to ensure the new static IP address has been entered correctly.

Configuring a Static Public or Private IP Address in Genie, Merlin and Bridge-IT (v.2.x firmware) Codecs

- 1. Press the **SETTINGS** button.
- 2. Select **Network** and press the button.
- 3. Use the down mavigation button to select LAN1, LAN2, Wi-Fi or a VLAN interface.
- 4. Select **Config** and then **Usage** and then the appropriate control and/or streaming mode for the connection. Next, press the button.
- 5. Select **IPv4 Mode** and press the or button.
- 6. Select **Static** and press the button.
- 7. Navigate to IPv4 Static and enter the IP address, then press the button.
- 8. Navigate to **IPv4 Subnet** and enter the Subnet Mask, then press the button.
- 9. Navigate to IPv4 Gateway and enter the Gateway details, then press the button.
- 10. Use the up ___ navigation button to scroll to the top of the menu and select **Apply Setting**, then press the ___ button to confirm the new settings.
- 11. Check the **Unit Details** menu to ensure the new static IP address has been entered correctly.

Configuring a Static IP Address in Commander G3 and i-Mix G3 Codecs

To set up a static IP address in Commander G3 and i-Mix G3 codecs select Menu > Configuration > Advanced > LAN settings > IP Setup > Setup > Static > IP Address > [enter IP address] > press OK > Subnet Mask [enter Subnet Mask] > press OK > Gateway [enter Gateway] > press OK > reboot the codec.

Record IP Address Details

IPv4 Static IP Address							
IP Address							
Subnet Mask							
Default Gateway							
IPv6 Mode: Manual	(Not	availab	le in lega	cy Comn	nander a	nd i-Mix	codecs)
IP Address		:	:	:	:	:	:
	<u> </u> :						
IP√6 Prefix Size							
IPv6 Gateway		:	:	:	:	:	:
	:						

Getting Connected

Once the studio codec is configured you are now ready to receive an incoming call from the remote codec over the internet. Always dial from the field codec to the studio codec over the internet unless the remote codec is assigned a public static IP address and you know this address. If you dial the studio using a cell-phone data network at the remote site you will not normally experience any firewall or port blocking issues at the remote end of the link using default Tieline ports.

Troubleshooting: How to Determine Where Firewall Port Blocking is Occurring

If you find you are unable to either send or receive audio between the studio and remote codecs you can use Tieline's Link Quality reading to diagnose where ports are being blocked. LQ can be displayed on the front **LCD SCREEN** of Tieline's Bridge-IT, Merlin and Genie codecs by selecting **Cxns**, then select the connection you want to view and press the **OK** button. LQ readings are also displayed on the home screen of all Commander and i-Mix G3 codecs.

Link Quality (LQ) Readings

Send and Return LQ numbers help you to determine if a problem is occurring at either end of a connection. For example, on an IP connection the Return LQ reading represents the audio being downloaded from the network locally (i.e. audio data is being sent by the remote codec). Conversely, the Send LQ reading represents the audio data being sent by the local codec (i.e. being downloaded by the remote codec). To ensure a stable connection, try to maintain a reliable reading of 80 or higher for both the **Send** and **Return** LQ reading.

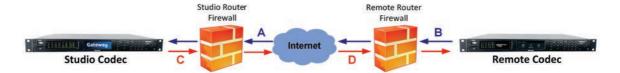


Important Note:

- The **Return** link quality reading is the same as the Local (**L**) setting displayed on a
- The Send link quality reading is the same as the Remote (R) setting displayed on a G3 codec.

Diagnosing Port Blocking via the Studio Codec LQ

If the studio codec **Return** LQ reading is **01** then incoming audio from the remote codec is being blocked by a firewall at either point A or B in the following diagram. If the studio codec **Send** LQ reading is **01** then outgoing audio from the studio is being blocked by a firewall at either point C or D in the following diagram.



Diagnosing Port Blocking via the Remote Codec LQ

If you attach your Tieline codec at the remote site to a LAN with access to the internet you can often dial and connect to the studio without any problem. It is less likely that a firewall will block outgoing TCP and UDP ports. However, if there is a firewall at the remote site it may block incoming data packets from the studio.

The principle is the same at the remote codec for diagnosing blocked ports. If the remote codec **Return** LQ reading is 01 then incoming audio from the studio codec is being blocked by a firewall at either point C or D in the preceding diagram. If the remote codec **Send** LQ reading is 01 then the outgoing audio from the remote codec is being blocked by a firewall at either point A or B in the preceding diagram.

Troubleshooting TCP Port Blocking

Error messages on the codec screen can help to diagnose TCP port blocking.

- 1. "Connection Refused" usually means that the firewall is configured correctly but the codec is not using the expected port. For example, the firewall is set up to forward via port 9002 but codec is 'listening' to port 10,000. "Connection Refused" is not normally shown if the firewall is not configured correctly because a firewall will by design silently drop any forwarding requests to ports that it doesn't have open (see next point). Note: "Connection Refused" will also be displayed if the Commander G3 or i-Mix G3 codec you are calling is already connected.
- 2. "Connection Timeout" can mean one of two things:
 - The firewall is not configured correctly and the attempted codec connection is being silently dropped, e.g. a remote codec is dialing to port 9002 but the studio firewall port forwarding is not configured.
 - The UDP port is not port forwarded correctly. Tieline codecs send test data during connection establishment to make sure that the audio path is configured correctly; if this process fails then it will also result in a "Connection Timeout".

How do I determine which end is blocking data flow?

Tieline test codec firewalls have the default Tieline TCP and UDP ports open. You can dial into these test codecs (or other codecs you know are configured correctly) from your recently configured studio and remote codecs and use the LQ readings to diagnose whether your studio or remote codec firewall is blocking your data packets. If one codec connects ok and the other one doesn't, then you will know which end is likely to be causing the problem. As an example:

- 1. Dial from site 1 to a Tieline test codec.
- 2. Dial from site 2 to Tieline test codec.

If both of these connect successfully then the "outbound" TCP path for session data is OK, and the inbound UDP audio path is OK.

- 3. Dial to site 1 from a codec you know is configured correctly.
- 4. Dial to site 2 from a codec you know is configured correctly.

If either of these calls fail then TCP and/or UDP inbound data is being blocked on the failed connection (see "Troubleshooting TCP Port Blocking" above).

Testing your Codec

- Visit www.tieline.com or contact Tieline for a list of test IP codec addresses you can use to verify your codec is configured correctly.
- See Testing IP Network Connections for more IP test information.