# Engineering Advisory 70568 Polycom<sup>®</sup> UC Software PTT/Group Paging Audio Packet Format



This engineering advisory provides details about the format of the packets used in the Push-to-Talk (PTT) and Group Paging features available in Polycom<sup>®</sup> UC Software 4.0.x.

This engineering advisory applies to administrators or product developers who want to interoperate their products with the Polycom Multicast PTT/Group Paging feature.

The PTT and Group Paging features work by multicasting packets on a certain channel to an IP address and port set by an administrator. By default, packets are multicast to the IP address 224.0.1.116 using UDP and port 5001. Each packet consists of either a header, or a header and additional audio, depending on the packet type. The header of each packet is 20 bytes and consists of the 5 fields shown next in Table 1: Header Fields and Size.

Table 1: Header Fields and Size

Op Code	Channel Number	Host Serial Number	Caller ID Length	Caller ID
1 byte	1 byte	4 bytes	1 byte	13 bytes

The header network byte order begins with the Op Code field and ends with the Caller ID field as highlighted in the following Wireshark capture.

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C	Data: 1	011fz	333f	6a00	d303	030346	632	333	333	663	661	000	000a0	C8.					
	[Length		Section Section																
0020	01 74		-	89 66	02		<u>6f</u> 33	A COLORED	11	f2	33 00	3f	6a 00				03 3f6a.		
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0070 0080	and the second sec	ff 7 7e f		7e 7e				1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1			7e ff		7e 7c				} }1.		
0090	7d 7d						And and	7e			ff			Contraction of the local sectors of the local secto	1~.~			·~	
🔵 Dati	a (data.da	ata), 506	bytes				Pac	kets:	609	Displ	ayed	: 135	Marke	d: 0 Dr	opped:	0			

# **Header Fields**

This section describes each of the 5 fields found in the header.

# Op Code

The Op Code field is 1 byte and provides information about the packet type. There are three packet types: *PTT Alert, PTT Transmit,* and *PTT End of Transmit*. Use Table 2: Op Codes to match an Op Code to the corresponding packet type and to understand the function of each packet type.

#### Table 2: Op Codes

Op Code	Packet Type	Packet Purpose
OF	PTT Alert	This packet signals all phones listening on the current channel that a phone is about to begin broadcasting.
10	PTT Transmit	This packet is used to transfer audio data and is the only packet type which contains audio frames.
FF	PTT End of Transmit	This packet signals all phones that the broadcasting phone has completed its broadcast.

# **Channel Number**

The channel number field is 1 byte and represents the channel that the packet is transmitted on. The channels range from 1-50, with channels 1-25 for PTT, and channels 26-50 for paging. The PTT/Paging feature enables users to broadcast messages with a certain priority level: Normal, Priority, or Emergency. By default, the PTT feature treats channel 24 as a Priority channel and channel 25 as an Emergency channel while the Paging feature treats channel 49 as the Priority channel and channel 50 as the Emergency channel. The Priority and Emergency channels can be changed by administrators.

# **Host Serial Number**

The host serial number field is 4 bytes and represents the last 4 bytes of the serial number/MAC address of the broadcasting phone. This field is used for contention resolution – when multiple phones begin broadcasting on the same channel at the same time, the phone with the lowest serial number continues to broadcast and all other phones will stop broadcasting. Any 32-bit number can be used in place of the serial number as long as its value is guaranteed to be unique among the multicast participants.

# **Caller ID Length**

The caller ID length field is 1 byte and represents the number of bytes in the caller ID field. Although the packet includes the caller ID length, the encoded length and length of the caller ID string are fixed at 13.

# Caller ID

The caller ID field is 13 bytes and consists of a text string (a phone's extension for example) that identifies the broadcasting phone. If this string is less than 13 bytes, it is terminated with a null. Otherwise, if this field is null, the value from reg.1.displayName (found in the **reg-basic.cfg** file) will be used. If that too is null, the phone's MAC address will be used. A receiving phone displays the caller ID on its screen.

# Audio Data

Audio data is only present in a *PTT Transmit* packet. There are two codecs which can be used to send the audio data:

- 1 G.722 Typical audio payload is 240 bytes (30ms)
- 2 G.711u Typical audio payload is 240 bytes (30ms)

Audio data consists of a 6 byte audio header followed by two frames of audio data. The first frame is a redundant frame—it contains a copy of the audio from the previous packet. The second frame contains the current audio. The only exception is the first PTT Transmit packet, which will not contain a redundant audio frame. An example Audio Header is shown next in Table 3: Audio Header Example

### Table 3: Audio Header Example

Number of Bytes	Description	Notes
1	Codec Type	0x00 means G.711μ 0x09 means G.722
1	Flags Byte	Not applicable
4	Sample Count	RTP timestamp for the second audio frame (except for the first PTT transmit packet, then it's for the first and only audio frame)

# **PTT/Page Session**

A PTT or Page is initiated by sending 31 PTT Alert packets at approximately 30 millisecond intervals, followed by the transmission of the audio data in PTT Transmit packets. Upon completion of the Page, after a 50 millisecond delay, 12 PTT End of Transmit packets are sent at approximately 30 millisecond intervals completing the Page.

# **Example Page Session**

The following example shows a Wireshark capture of a short paging session, specifically a PTT session, using the G.711µ codec with a 20 msec sample size (resulting in an audio frame of 160 bytes). A different PTT session will contain a different number of bytes (and packets).

The following tables (Tables 4 to 12) provide packet details of the entire audio frame (187 packets). Included is:

- Packet number and type
- Transmit time in seconds
- Source and destination IP addresses
- Protocol used
- VLAN formation
- Packet contents—Highlighted contents are explained in detail

In some instances, the packet contents are a repeat of previous packets. These are noted in the following tables.

#### Table 4: First PTT Alert Packet

Pkt No.	Time (seconds)	Source IP Address	Destination IP Address	Protocol	VLAN Info
1	0.000000	192.168.1.103	224.0.1.116	UDP	Source port: complex-link Destination port: complex-link
		0 01 74 00 04 f2 11 15 1 00 00 40 11 c0 c8 c0		E. @g	Alert packet
		9 13 89 00 1c 90 63 <b>0f</b> 5c 6f 64 79 20 4d 65 73		e dy Meserv	Highlighted contents described in table below

#### Table 5: Contents of PTT Alert Packet

Field Value	Number of Bytes	Field Name	Notes
Of	1	Op Code	PTT Alert
1a	1	Channel Number	26 (first channel in paging range)

Field Value	Number of Bytes	Field Name	Notes
f2 11 15 11	4	Host Serial Number	Last four bytes of phones MAC address (004f2111511)
0d	1	Caller ID Length	13
4d 65 6c 6f 64 79 20 4d 65 73 65 72 76	13	Caller ID	Melody Meserv

### Table 6: Remainder of PTT Alert Packets

Pkt Nos.	Time (seconds)	Source IP Address	Destination IP Address	Protocol	VLAN Info
2 - 31	every 0.030 (approx.)	192.168.1.103	224.0.1.116	UDP	Source port: complex-link Destination port: complex-link
	0010 00 30 16 71 0020 01 74 13 89	01 74 00 04 f2 11 15 1 00 00 40 11 c0 c8 c0 a 13 89 00 1c 90 63 0f 1 6f 64 79 20 4d 65 73 6	8 01 67 e0 00 .0.q a f2 11 15 11 .t	@g	Repeat of Packet 1

After the 32 PTT Alert packets, the actual data transmission starts with the PTT Transmit packets.

#### Table 7: First PTT Transmit Packet

Pkt No.	Time (seconds)	Source IP Address	Destination IP Address	Protocol	VLAN Info
32	0.969281	192.168.1.103	224.0.1.116	UDP	Source port: complex-link Destination port: complex-link

Pkt No.	Time (seconds)	Source IP Address	Destination IP Address	Protocol	VLAN Info
	0010 00 d6 16 9 0020 01 74 13 8 0030 0d 4d 65 0 0040 6f ca 7b ff 0050 f5 f8 ef fb 0050 79 f6 dd f 0070 de 59 f8 ff 0080 b8 f9 5d f 0080 b9 f9 f9 00a0 f9 b9 f9 f9 00b0 df 75 76 9 00c0 f3 30 f4 f7	5 5e 7a f7 70 f4 7a 5e 5 5c 6d b1 9f b9 9d b9 4 9f df fb f2 b3 fb 76 8 b2 fa dc dc fb df 91 9 df f9 9d f5 f9 f7 bb 9 fb fb f2 f9 79 f9 b7 f 9 5f 76 b9 f6 b9 6e ea 9 c d7 fa ba 5d da 7e 5 7 b8 de df f4 73 bb 76 la 5e 5f bc 9c f7 bc 7	c0 a8 01 67 e0 00      0         10 1a f2 11 15 11       .t         73 65 72 76 09 00       .Mela         c db f2 5e d7 dc       o. {.^.         o b3 f3 9d f9 f3      m         f6 ba d7 fb b8       y         o 5f f9 d8 dd b7       Y         79 f7 9d f5 75      v         a8 6d f1 f3 9d      v         o 78 fa da 99 f9       0	E. g ev ody Meserv z.p.z^^ v v yu y yu y yu y s.~x s.~x	1 <sup>st</sup> PTT Transmit packet Highlighted contents described in table below

## Table 8: Contents of First PTT Transmit Packet

Field Value	Number of Bytes	Field Name	Notes
10	1	Op Code	PTT Transmit
1a	1	Channel Number	26 (first channel in paging range)
f2 11 15 11	4	Host Serial Number	Last four bytes of phones MAC address (004f2111511)
0d	1	Caller ID Length	13
4d 65 6c 6f 64 79 20 4d 65 73 65 72 76 09	13	Caller ID	Melody Meserv
00	1	Codec	G.711 μ
6f	1	Flags	Not applicable
ca 7c 95 5e	4	RTP Sequence number	
7a f7 70 f4 7a	160	Audio frame	

	Time (seconds)	Source IP Address	Destination IP Address	Protocol	VLAN Info
	0.989055	192.168.1.103	224.0.1.116	UDP	Source port: complex-link
					Destination port: complex-link
	0000 01 00 5 0	0.01.74.00.04.50.11.15	5 11 08 00 45 00^t.		2 <sup>nd</sup> PTT Transmit packet
		0 01 74 00 04 f2 11 15			
		0 00 00 40 11 c0 03 c0			
		9 13 89 00 c2 65 76 1			
		5c 6f 64 79 20 4d 65 7.			
		5 5e 7a f7 70 f4 7a 5e			
		5c 6d b1 9f b9 9d b9			
		4 9f df fb f2 b3 fb 76 f			
		8 b2 fa dc dc fb df 9b			
		9 df f9 9d f5 f9 f7 bb 7		yu	
		fb fb f2 f9 79 f9 b7 fb	y		
		) 5f 76 b9 f6 b9 6e ea a c d7 fa ba 5d da 7e 57	· · · · · · · · · · · · · · · · · · ·	nm	
		' b8 de df f4 73 bb 7e '		].~W	
		a 5e 5f bc 9c f7 bc 78		.s.~x	
		a 56 51 bc 9c 17 bc 78 d7 5f 7b 5a ba f8 be		x.y.m.	
		78 71 9f b9 71 f5 9d 5		Zy	
		78 71 91 09 71 13 90 3 lb 7a 5b b0 f6 f0 df 77		q∼ wx	
		3 bc 9f f9 f9 f6 57 fc 7	- · · · · · · · ·	wx W.y	
		fc db f5 f7 9d 76 fe fa	···· /_··	vv. y .vx~	
		8 d9 fb f3 de b8 f6 38			
		b 7c b8 fa de fe de f3			
		e 75 75 f7 f8 fe be 5c	-	v ∖u{z	
		6 f8 fb dd f1 f5 fb fb f2			Uteb tebas deservations and the state
			f8 75 b7 78 b4		Highlighted contents described in
	0180 5c fa f1 fa				table below

#### Table 9: Second PTT Transmit Packet

#### Table 10: Contents of Second PTT Transmit Packet

Field Value	Number of Bytes	Field Name	Notes
10	1	Op Code	PTT Transmit
1a	1	Channel Number	26 (first channel in paging range)
f2 11 15 11	4	Host Serial Number	Last four bytes of phones MAC address (004f2111511)
0d	1	Caller ID Length	13

Field Value	Number of Bytes	Field Name	Notes
4d 65 6c 6f 64 79 20 4d 65 73 65 72 76 09	13	Caller ID	Melody Meserv
00	1	Codec	G.711 μ
6f	1	Flags	Not applicable
ca 7c 95 5e	4	RTP Sequence number	
7a f7 70 f4 7a	160	Audio frame	Redundant frame (duplicate of last frame in previous packet)
d7 5f 7b 5a ba	160	Audio frame	New audio frame

The remainder of the PTT Transmit packets (34 - 174) are transmitted every 30 milliseconds. Each packet will contain two audio frames:

- A duplicate of the last audio frame from the previous packet
- A new audio frame

The data transmission is complete. The end of the broadcast is signaled by 12 PTT End of Transmit packets.

#### Table 11: End of PTT Transmission Packets

Pkt No.	Time (seconds)	Source IP Address	Destination IP Address	Protocol	VLAN Info
175- 187	3.824994	192.168.1.103	224.0.1.116	UDP	Source port: complex-link Destination port: complex-link
	0010 00 30 17 1e	01 74 00 04 f2 11 15 1 00 00 40 11 c0 1b c0 a	18 01 67 e0 00 .0@	E. 9g	End of Transmission packet
		13 89 00 1c a0 62 ff 1 6f 64 79 20 4d 65 73 0		b dy Meserv	Highlighted contents described in table below

#### Table 12: Contents of End of PTT Transmission Packet

Field Value	Number of Bytes	Field Name	Notes
ff	1	Op Code	PTT End of Transmit

Field Value	Number of Bytes	Field Name	Notes
1a	1	Channel Number	26 (first channel in paging range)
f2 11 15 11	4	Host Serial Number	Last four bytes of phones MAC address (004f2111511)
0d	1	Caller ID Length	13
4d 65 6c 6f 64 79 20 4d 65 73 65 72 76	13	Caller ID	Melody Meserv

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