

Designing with Dante and AES67 / SMPTE ST 2110

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Design Principles (and Why We Recommend Them)

Clarify Common Misunderstandings About Dante (and PTPv2).

Highlights of the Dante update with SMPTE ST 2110 support.

Dante Domain Manager™

Dante Domain Manager is a key part to Audinate's ST 2110 solution. If you are not familiar with this product, watch this video to learn its role and core functions.





In Depth Tour of Dante Domain Manager https://youtu.be/xCY3JNpCu_k





History: Audio Networks Solutions and Open Standards



CobraNet^{**}



https://en.wikipedia.org/wiki/Microsoft_Office_97



QoS Quality of Service (Prioritization of Time Sensitive Data)

ZeroConfig Automatic Peer-to-Peer Configuration (No need for Static or DHCP config) PTP Precision Time Protocol (Network Synchronization)

IGMP Snoping Internet Group Management Protocol (Reduces Impact of Multicast Distribution)



Best Practices for Audio Networks

AES AESTD1003V1 - June 6, 2009 http://www.aes.org/technical/documents/AESTD1003V1.pdf

"Audio networking systems are characterized by the transport of uncompressed audio in PCM format, which in principle could be reformatted as requested.



In practice, there are several issues for compatibility between formats that should be addressed and solved with specific implementations."



What is AES67?

Media Networking Alliance - October, 2015 http://medianetworkingalliance.com/faq/

"AES67 enables audio-over-IP streaming interoperability between audio networking [solutions] currently available, such as Dante, Livewire, Q-LAN, Ravenna [and WheatNet IP]. <u>It is not a new technology,</u> <u>but a bridging compliance mode.</u>"

Network Solution vs Interoperability Standard:



Network Solution: Dante, Livewire+, Ravenna, Q-LAN, WheatNet IP

Interoperability Standard: AES67, SMPTE 2110

Network Solution vs Interoperability Standard:



Dante, AES67 & AVB Aiden Williams, Audinate - March 16, 2015

https://www.youtube.com/watch?v=E_E5JEVGH9c

By remaining pragmatic in scope – by narrowly focusing on the transport piece – AES67 was neither too weighty to get agreement from the various solutions nor too complex to be implemented.



EBU

Production Infrastructure Strategic Programme - IBC 2018

Partially available

Rarely available

Build a Team on a Network Solution like Dante





Build a Team on a Network Solution like Dante





Build a Team on a Network Solution like Dante





For video teams, ST 2110 represents the move from point-to-point SDI to networked flexibility and smashing the bandwidth cap.

For audio, ST 2110 is old hat - an improved AES67 representing compatibility with video teams on the network.































- Replaces:
- -SDI
- -MADI
- -AES3





Even Video Tends to Stick to a "Family of Products"

BBC Cardiff Central Square

Oct 6, 2019 https://youtu.be/3GwuGyOmzxM

Key Takeaway:

Sticking to families of products gains advantage of improved control.



Control Methodology

- Separate & secured control network
- BNCS control interface to GV Convergent using NP0017
- GV Convergent controls the network fabric via Cisco DCNM
- GV Convergent connects GV edge devices via GV APIs
- Intention was to integrate NMOS IS-04 and IS-05 but the necessary level of interop has not matured in time.



BBG cymru wales

https://youtu.be/3GwuGyOmzxM

IP SHOWCASE THEATRE AT IBC2019 : 13-17 SEPT 2019

Send/Receive Dante and Open Standards Simultaneously



Build on Dante, Connect to Other Teams with ST 2110















Does Dante Support PTPv2? Yes.



PTPv1 and PTPv2:



Dante Is 75% of the Open Standards Market



Source: 2018 Roland Hemming Networked Audio Products http://www.rhconsulting.eu/blog/files/NetworkedAudio.html













This is not a "format war" – the streams in each mode are compatible.

As of v4.2 firmware for Dante devices, AES67 streams now use a fixed time offset value matching SMPTE 2110 requirements.

The question is, "How do you want to work?"



Simple, Automatic Config

Manual Configuration



Clocking Architecture SDP Customization



AES67 Stream Configuration

🙋 Create Multicast Flow			×	
Dante-Demonstr to 64 ch	Dante-Demonstration-Device supports up to 64 channels per flow.			
RTP flows for AES67 have	a maximum of 8	channels per flo	w.	
Select one or more transmit ch	annels to be pla	ced in multicast	flows.	
Flow Config (Optional) —			_	
🔿 Dante		AES67		
Destination Address: O Auto Manual				
IP Address: 239 . 69 . 21 . 101				
Port				
Port.	5004			
Channel		Add to	•	
Name		- New Fi	ow	
Right				
Digital L				
Digital R				
05				
06				
07		\checkmark		
08		\checkmark		
Create	Cancel			



SAP Announcement

Fixed SDP Profile

v=0

o=- 13769332640 13769332640 IN IP4 192.168.1.169 s=Dante-Demonstration-Device : 32 i=2 channels: 07, 08 c=IN IP4 239.69.8.15/32 t=0 0 a=keywds:Dante a=recvonly m=audio 5004 RTP/AVP 97 a=rtpmap:97 L24/48000/2 a=ptime:1 a=ts-refclk:ptp=IEEE1588-2008:00-1D-C1-FF-FE-10-29-96:0 a=mediaclk:direct=0

AES67 Stream Configuration

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Select one or more transmit channels to b	e placed in multicast flows.		
Flow Config (Optional)			
O Dante	AES67		
Destination Address: () Auto	Manual		
IP Address: 239 . 69 . 21 . 101			
Port: 5004			
Channel Name	Add to New Flow		
Left			
Right			
Digital L			
Digital R			
05			
06			
07			
08			
Create Cancel			

• Manual Stream IP/Port Addressing

ow Config (Optional) —	
ODante	AES67
Destination Address:	🔿 Auto 💿 Manual
IP Address:	239 . 69 . 21 . 101
Port:	5004
	ow Config (Optional) – Dante Destination Address: IP Address: Port:



Simple, Automatic Config

Manual Configuration



Clocking Architecture SDP Customization





Dante Domain Manager™

Optional

Required

Which ST 2110 parameters can be controlled?

- **PTP v1 Multicast**: Disabling PTP V1 multicast can prevent instability in non-Dante SMPTE devices requires all Dante devices 4.2 firmware.
- **PTP v2 Domain Number**: Can be set to any value between 0 and 127
- **PTP v2 Priority 1 / 2**: Determines which devices in a SMPTE clock domain will be automatically elected as clock master
- PTP v2 Sync Interval: Time interval between two successive PTP V2
 multicast sync packets
- PTP v2 Announce Interval: Time interval between two successive PTP v2 multicast announce packets.
- **PTP v2 Multicast TTL**: The range over which a PTP V2 multicast packet is propagated in your network
- **PTP Slave Only**: Devices in domain will not be elected as clock master

Advanced Settings			
Broadcast Studio			
Advanced settings can be used to configure interoperability, site-based clocking partitioning and unicast clocking device selection. Warning! Changing settings may interrupt audio. Audio/Clocking Parameters SAVE CHANGES CANCEL EDITING			
MODE	SMPTE		\$
PTP V1 MULTICAST			
PTP V2 DOMAIN NUMBER	127		0
PTP V2 PRIORITY 1	128		٢
PTP V2 PRIORITY 2	128		٢
PTP V2 SYNC INTERVAL	-3		٢
PTP V2 ANNOUNCE INTERVAL	-2		٢
PTP V2 MULTICAST TTL	1		٢
PTP SLAVE ONLY			×
RTP TRANSMIT PORT	5004		٢
SYSTEM PACKET TIME	1ms		\$
RX LATENCY	2ms		\$
RTP PREFIX V4	69		٢

Which ST 2110 Standards Used?

- SMPTE 2110 is a suite of standards Dante products support:
 - 2110-10 Defines the timing relationships between all components in the system
 - 2110-30 RTP Audio transport, similar to AES67
 - 2059-2 PTP Media Profile (IEEE 1588-2008)
 - 2022-7 Redundancy

Dante Domain Manager Propagates Settings For You



ST 2110 Stream Configuration

🧟 Create Multicast Flow	×		
Dante-Demonstration-Device supports up to 64 channels per flow.			
Select one or more transmit chan	nels to be placed in multicast flows.		
Flow Config (Optional)			
🔵 Dante 💿 SMPTE A	◯ SMPTE B ◯ SMPTE C		
Packet time: 1	msec		
Destination Address: () Auto (Manual		
Primary IP Address:	239 . 69 . 21 . 101		
Primary Port:	5004		
Secondary IP Address: 239 . 169 . 21 . 101			
Secondary Port:	5004		
Channel Name	Add to New Flow		
Left			
Right			
Digital L			
Digital R			
05			
06			
07			
08			
Create	Cancel		

- Manual Stream IP/Port Addressing
- SMPTE Audio Levels A, B & C

Г	Flow Config (Optional)	
	◯ Dante	TE A O SMPTE B O SMPTE C
4	Packet time:	1 msec
	Destination Address:	O Auto Manual
	Primary IP Address:	239 . 69 . 21 . 101
	Primary Port:	5004
	Secondary IP Address:	239 . 169 . 21 . 101
	Secondary Port:	5004
U		

ST 2110 Stream Configuration

🧕 Create Multicast Flow		×	
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Packet time:	1 msec		
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Primary Port:	5004		
Secondary IP Address:	239 . 169 . 🔅	21.101	
Secondary Port:	5004		
Channel		- Add to	
Name		New Flow	
Left			
Right			
Digital L			
Digital R			
05			
06			
07			
00			
Creat	e Cancel		

ST 2110-3x "Audio Levels"

	Description	Packet Time	
A:	1 - 8 channels	1.000msec	
B:	1 - 8 channels	0.125msec	
C:	1 - 64 channels	0.125msec	

(All streams are 24-bit, 48kHz)



Simple, Automatic Config

Manual Configuration



Clocking Architecture SDP Customization







Dante Latency



Latency Definitions:



Rx at Interface







Wrap-Up









Build on Dante, Connect to Other Teams with ST 2110













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PTP V2 SYNC INTERVAL	-3		٢	
PTP V2 ANNOUNCE INTERVAL	-2		٢	
PTP V2 MULTICAST TTL	1		٢	
PTP SLAVE ONLY			×	
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