

NAB 2015, Las Vegas, April 12th, 2015



Introduction to AES67 & How to get it into your Products



Agenda:

	<u>Topic</u>	<u>Speaker</u>
20 min.	Introduction to MNA	Marty Sacks (Telos Alliance) & Stefan Ledergerber (Lawo)
	AES67 – a quick Recap	Andreas Hildebrand (ALC NetworX)
50 min.	RAVENNA SoM & Reference Designs	
	AudioLAN SoMs	Arie van den Broek (Archwave)
	Digigram OEM Solution	Philippe Delacroix (Digigram)
	JADE Engine	Stephan Türkay
	Axia OEM Solution	Greg Shay (Telos Alliance)
10 min.	Lessons learned from implementing AES67	
15 min.	Q&A	all
	Round-up / AES67 on the Show Floor	Stefan Ledergerber (Lawo)



- Actively promote the adoption and standardization of AES67 as an audio interoperability standard through:
 - Marketing
 - Education
 - Training





- Provide developer support for AES67 products, and actively support those members producing AES67 compliant products, including, but not limited to:
 - AES67 System Development Kit
 - AES67 Test Tools





- Educate the business and end-user communities about the value, benefits and applications for AES67.



- The MNA is a non-profit corporation with membership available to all manufacturers, organizations, companies and individuals who are interested in the objectives and purposes of the alliance.



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AES67

**AES67-2013 Standard for
Audio Applications of Networks:
*High-performance Streaming Audio-
over-IP Interoperability***

published on September, 11th, 2013

Scope:

- **Interoperability guidelines** for professional, low-latency audio over campus and local area IP networks **using existing protocols wherever possible.**
- Excludes:
 - Non-IP networking
 - Low-bandwidth media
 - Data compression
 - Low-performance WANs and public Internet
 - Video (should provide good basis for follow-on video project)

Existing Audio-over-IP solutions / technologies / initiatives:

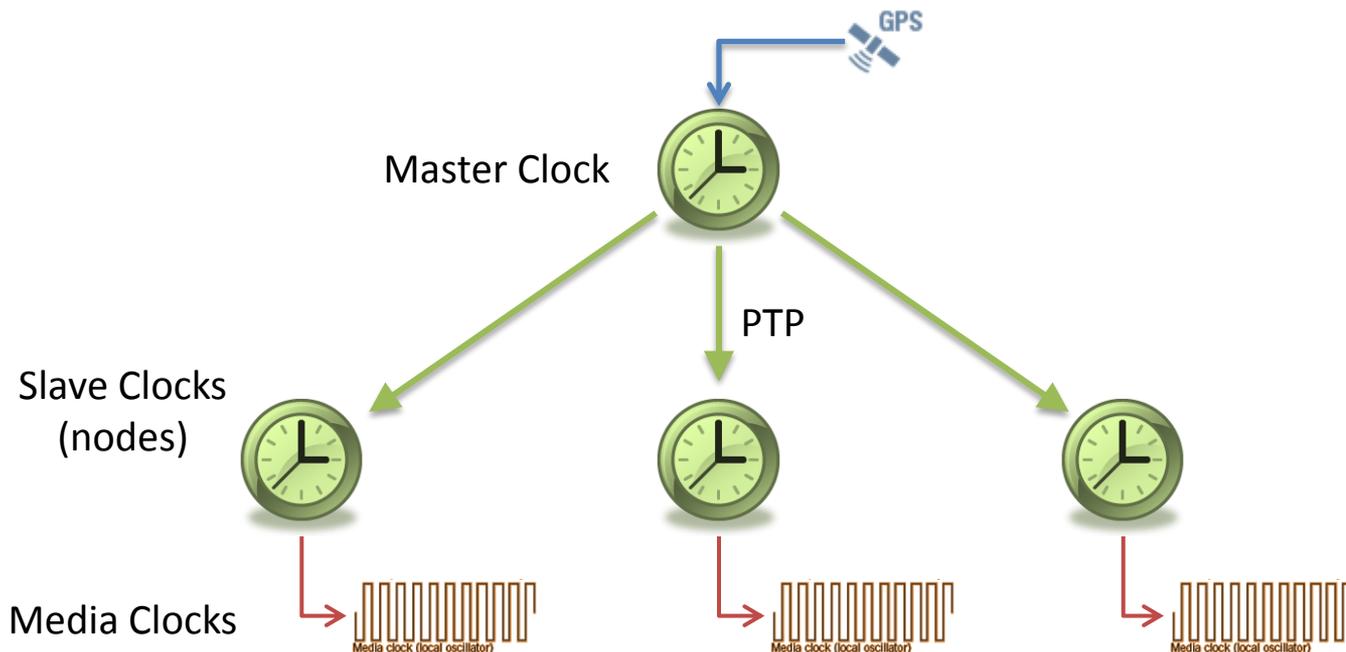
Technology	Purveyor	Date introduced	Synchronization	Transport
Livewire	Telos/Axia	2003	Proprietary	RTP
Wheatnet-IP	Wheatstone	2005	Proprietary	RTP
Dante	Audinate	2006	IEEE 1588-2002	UDP
N/ACIP	EBU	2007	Adaptive (per stream)	RTP
Q-LAN	QSC Audio Products	2009	IEEE 1588-2002	UDP
 RAVENNA	ALC NetworX	2010	IEEE 1588-2008	RTP
AVB	IEEE, AVnu	2011	IEEE 802.1AS	Ethernet, RTP

AES67 technology components:

- **Synchronization:** IEEE 1588-2008, default profile (media profile suggested)
- **local media clock generation**

AES67 technology components

– time synchronisation & media clock generation:



AES67 technology components:

- **Synchronization:** IEEE 1588-2008, default profile (media profile suggested)
- **local media clock generation**
- **Network:** IPv4 (IPv6), unicast / multicast & IGMPv2
- **Transport:** RTP/AVC (RFC 3550 & 3551) / UDP / IP

Selected solutions / technologies compared to OSI layer model:

OSI Layer	A-Net	EtherSound	CobraNet	Livewire, Dante & ...	AVB	AES67 & RAVENNA
Application						
Presentation						
Session				RTP		RTP
Transport				UDP		UDP
Network				IP		IP
Data Link		Ethernet	Ethernet	Ethernet	Ethernet	Ethernet
Physical	Copper	Copper / Fiber	Copper / Fiber	Copper / Fiber	Copper / Fiber	Copper / Fiber



AES67 technology components:

- **Synchronization:** IEEE 1588-2008, default profile (media profile suggested)
- **local media clock generation**
- **Network:** IPv4 (IPv6), unicast / multicast & IGMPv2
- **Transport:** RTP/AVC (RFC 3550 & 3551) / UDP / IP
- **Encoding:** 16 / 24 bit linear, 48 (44.1 / 96) kHz, channel count: 1..8
- **Packet Setup:** 48 samples (6 / 12 / 16 / 192), max. payload size: 1440 bytes
- **Quality of service:** DiffServ w/ 3 suggested traffic classes (DSCP)
- **Connection management:** SDP (refers to RFC 7273 - RTP Clock Source Signaling), SIP (unicast)
- **Discovery:** excluded, but several recommendations given (i.e. ZeroConf, SAP and others)

AES67 – the “O negative” of audio networking



RAVENNA

Q-SYS™



Dante™

ACIP



*AES67 – the “O negative”
of audio networking*





*AES67 – the “O negative”
of audio networking*



*AES67 – the “O negative”
of audio networking*

When will it be available?



*AES67 – the “O negative”
of audio networking*



When will it be available?



First AES67 plug-fest October 2014 @ IRT in Munich!

- 3.5 days of plugging
 - 22 participants (10 manufacturers, IRT, EBU, SR)
 - 16 products (15 were based on  RAVENNA)
 - Lots of streams – millions of packets!



First AES67 plug-fest October 2014 @ IRT in Munich!

- 3.5 days of plugging
 - 22 participants (10 manufacturers, IRT, EBU, SR)
 - 16 products (15 were based on  RAVENNA)
 - Lots of streams!

Tests included:

- Synchronization (PTP)
- Concurrent multicast streaming between all nodes w/ mandatory stream formats
- Unicast streams and SIP connection setup



AES67 interoperability successfully demonstrated!

AES TG SC-02-12-M: AES67 Development

- Outlining AES67 compliance test guidelines
- Specifying and engineering the compliance tests
- Planning and organizing plug-fests
- Improving the standard specification where necessary
- Participants:
 - anyone implementing AES67
 - parties / individuals with strong interest in AES67 interoperability

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What is **RAVENNA**?

RAVENNA Draft on
Operational Principles



Ingredients:

- 20 ml PTPv2
- 500 g RTP
- 1 pkt multicast
- 1 pinch of Bonjour

Cooking order:

1. Stew PTP to order
2. Add RTP
3. Mingle with multicast
4. Add Bonjour on top

Serve hot and Enjoy!



RAVENNA / AES67 Building Blocks from ALC NetworX

- (1) COMi.MX RAVENNA / AES67 SoM
- (2) C/C++ RAVENNA / AES67 Software Framework
- (3) COMi.MX Reference Design incl. full Source Code Package
- (4) RVSC for Windows

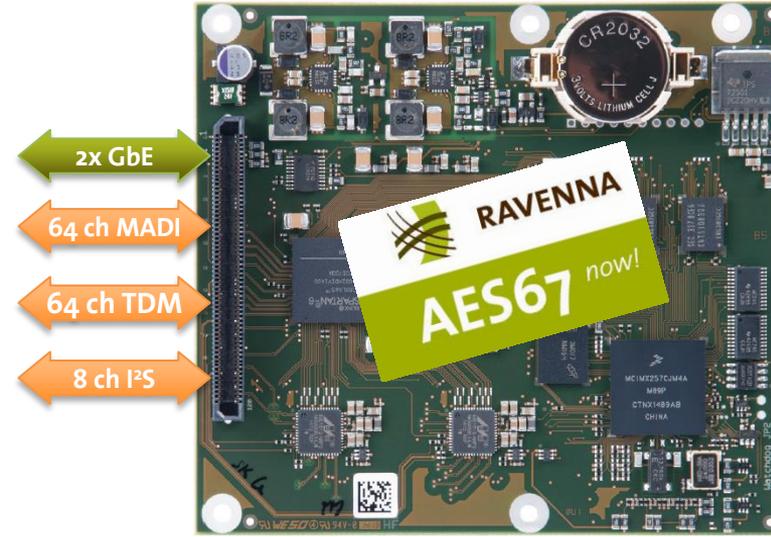
COMi.MX - RAVENNA / AES67 SoM

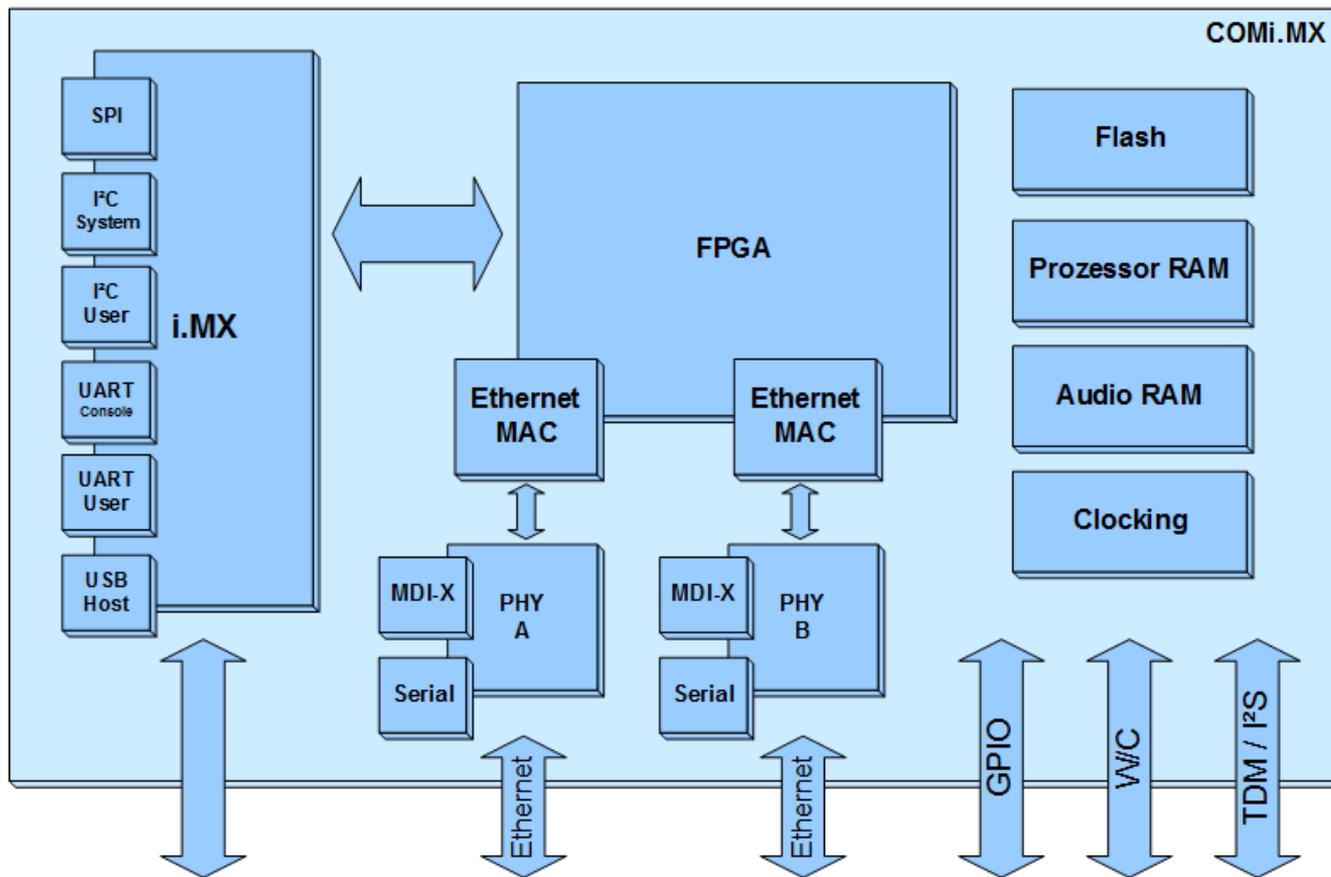
for quick & easy OEM integration



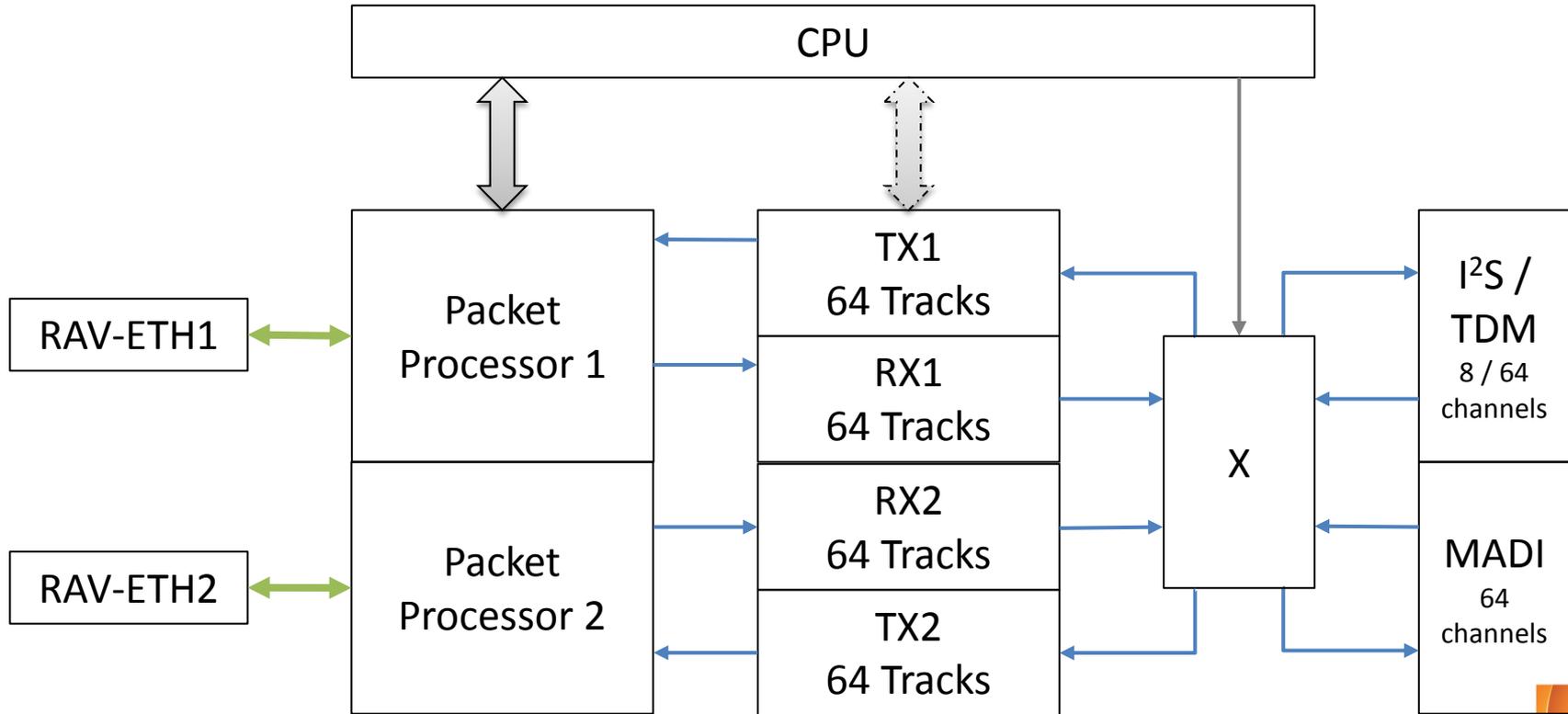
COMi.MX – RAVENNA / AES67 SoM

- Fully self-contained RAVENNA implementation
- 2x 64 channels in & out
- Up to 192 kHz sampling rate
- Full AES/EBU bit-transparent operation supported
- Lowest latency support: down to 1 sample/packet!
- Jitter / delay buffer up to 80 ms per channel
- Audio interfaces: I²S (8 ch) / TDM, MADI (64 ch)
- 4-tier 256 x 256 audio matrix
- Load balancing support through 2 GbE NICs (redundancy w/ later firmware)
- Full AES67 support (w/ firmware update)

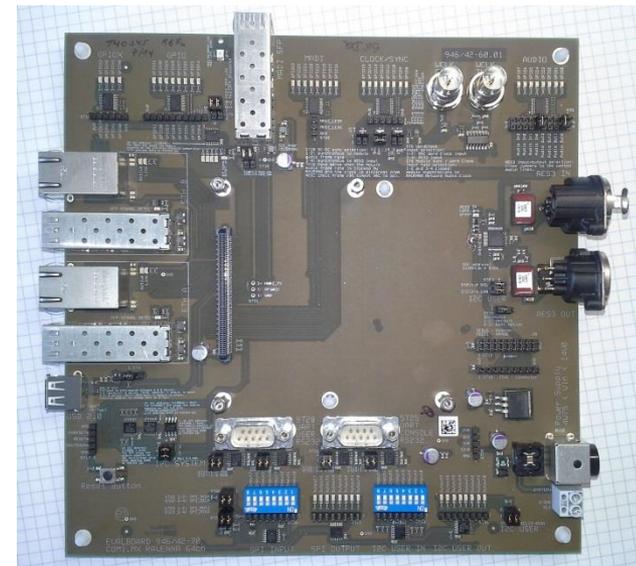
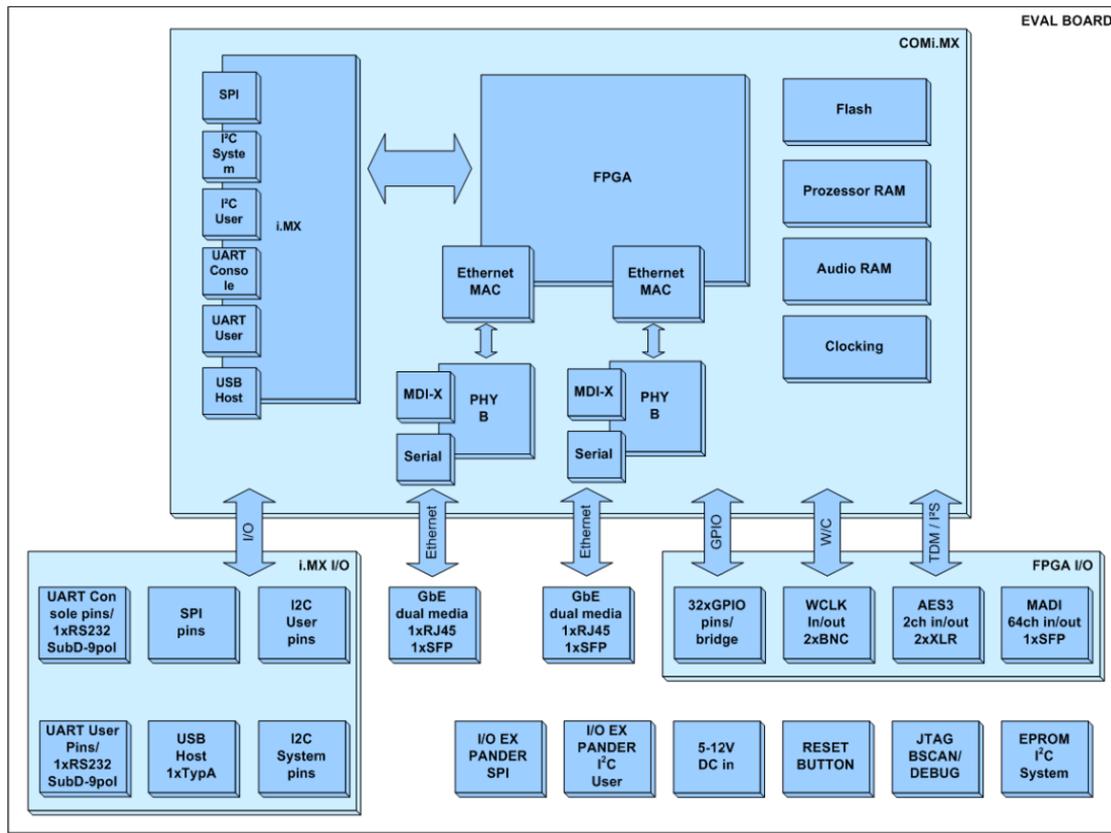




COMi.MX – data flow (parallel operation):

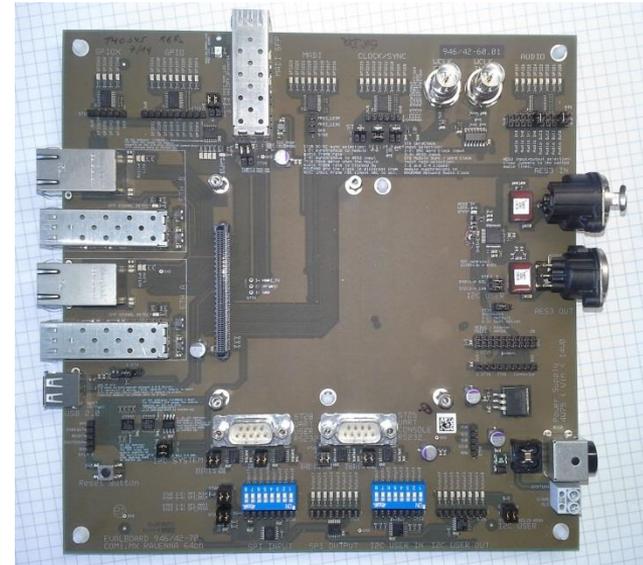


COMi.MX – Eval Board



COMi.MX – Eval Board

- Test and evaluation board for COMi.MX
- Single voltage power input (4.75 – 15 V DC)
- AES3 in / out (2 x XLR)
- 64 channel MADi in / out (1 x SFP)
- WCLCK in / out (2 x BNC)
- Dual GbE NICs (2 x RJ45 / 2 x SFP)
- 1 x UART system console (RS-232 SubD-9)
- JTAG Debug Port
- Access to: I²C (system + user), UART user, USB 2.0 (A/B), SPI, all GPIO lines



RAVENNA/AES67 Software Framework

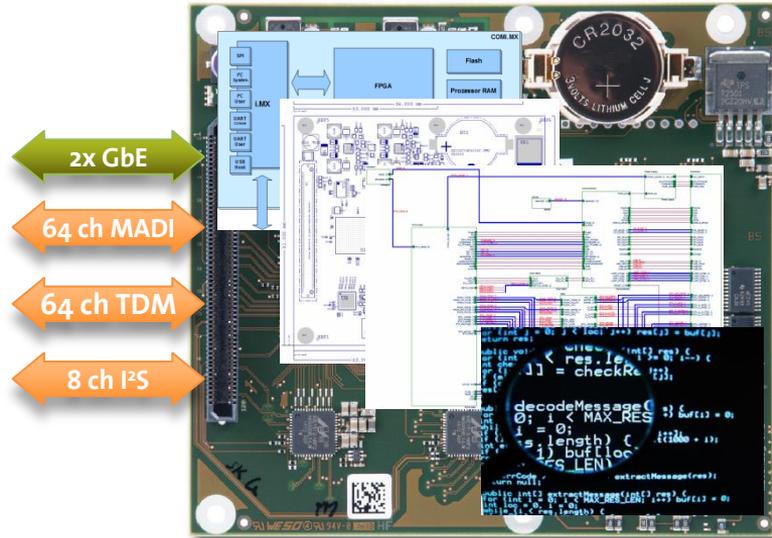
- C/C++ source code:
 - IEEE1588 clock protocol stack (based on ptp4L)
 - RAVENNA/AES67 streaming engine
 - Configuration & connection management
 - Advertising & discovery
 - Ember+ support
- Audio interface driver needs to be supplied by manufacturer (platform-specific)
- Technical documentation
- Complete Linux environment & development tool chain
- Reference platforms: COM8313 and Linux-PC



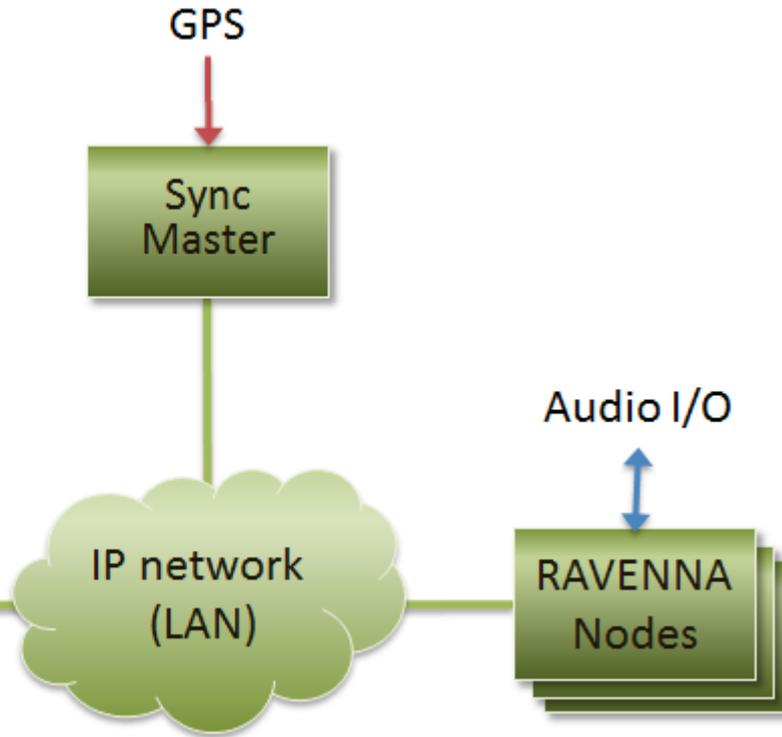
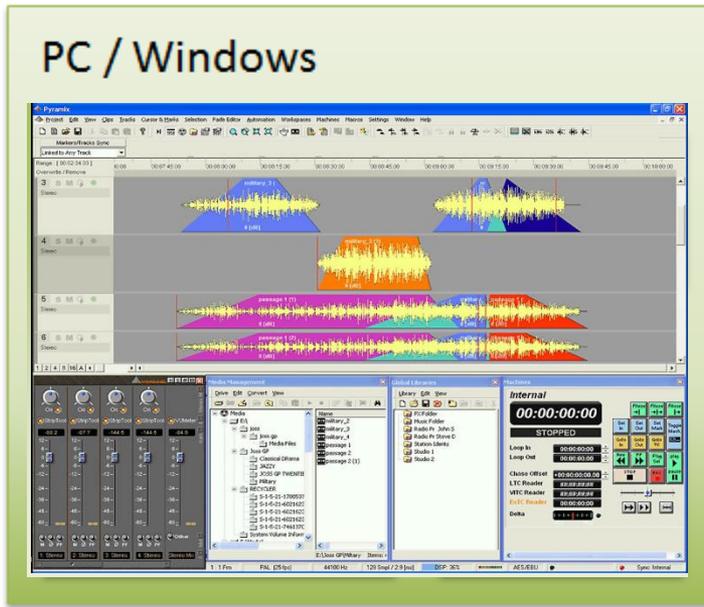
COMi.MX – Reference Design

SoM design & source code package:

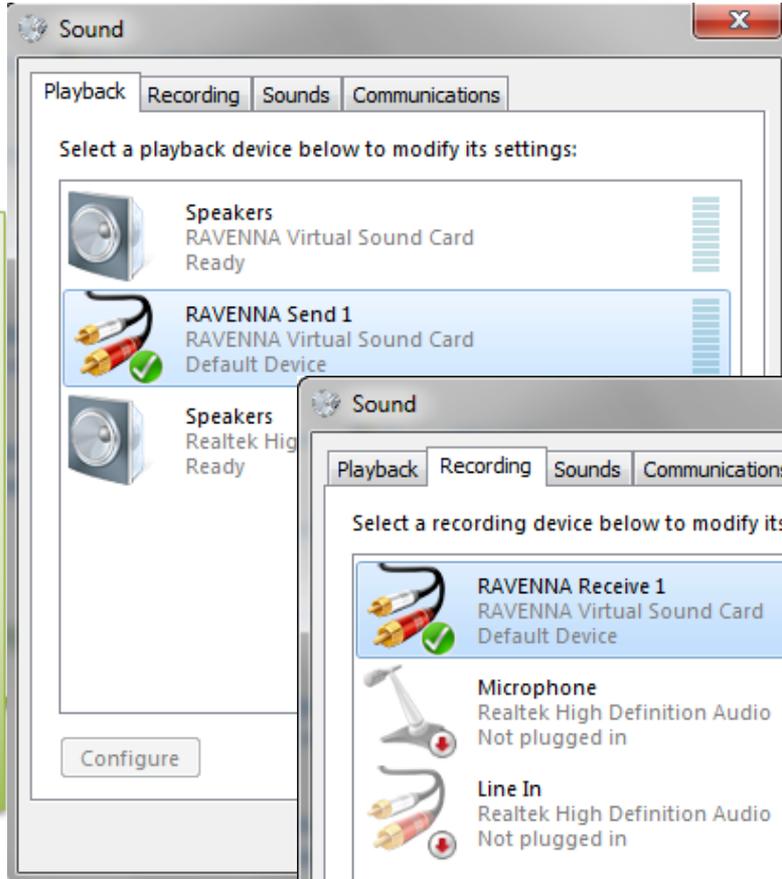
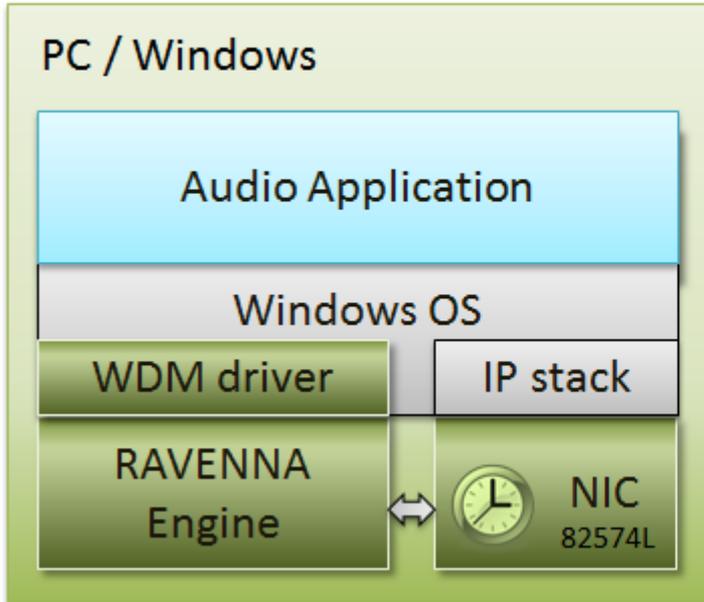
- Functional VHDL project w/ full VHDL source code for Xilinx Spartan6 on COMi.MX
- Schematic diagrams of COMi.MX board
- RAVENNA/AES67 software framework
 - C/C++ source code
 - Complete Linux environment & development tool chain
- Technical documentation
- Basic technical support



RVSC – RAVENNA Virtual Sound Card for Windows

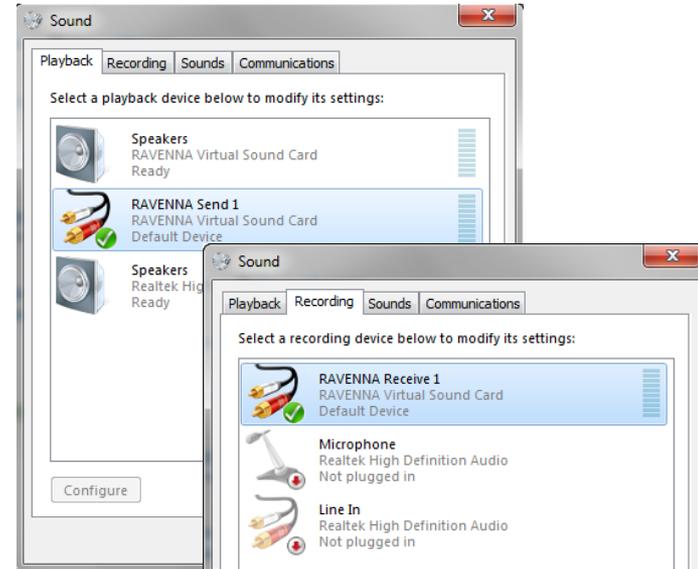


RVSC – RAVENNA Virtual Sound Card for Windows



RVSC – RAVENNA Virtual Sound Card for Windows

- Available for Windows 7 & 8
- Supports WDM driver model
- Up to 64 channels playback / record
- Typ. processing latency: ~ 15 ms
- HW PTP support through Intel 82574L or i350
- AES67 support (multicast-only)
- Free version w/ 2 playback & 1 record WDM device
- Full version available to RAVENNA partners for OEM integration



Thank you for your attention!

**RAVENNA booth in Central Hall
C2218**

Contact information:

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How to get it into your Products:

ARCHWAVE
connecting audio

Arie van den Broek
CEO Archwave



AES67

Distinct Networked Audio market segments:



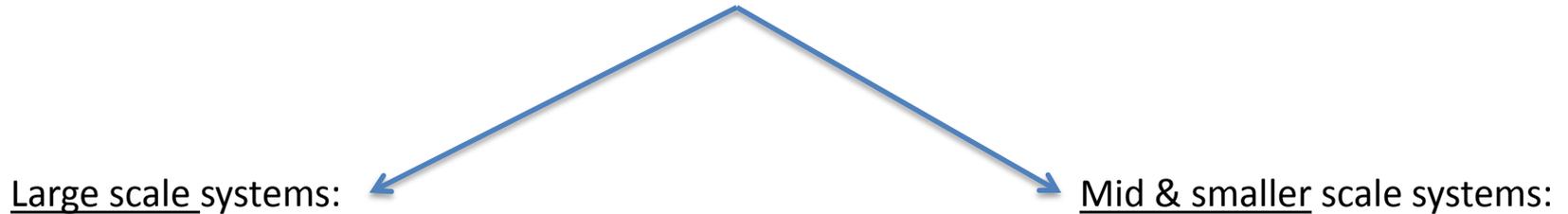
Large scale systems:

- Targeting Broadcast, Fixed install
- High node & channel count
- Knowledgeable audience
- Key decision parameters:
 - Technology
 - Interoperability

Mid & smaller scale systems:

- Targeting MI & Sound Reinforcement
- Lower node & channel count (<64ch/<25)
- More 'Novice' audience & user base
- Key decision parameters
 - Ease-of-Use / Simplicity
 - Cost, cost & cost...

Distinct Networked Audio market segments:



Mass-adaption requires tailored solutions
for each segment

Our Approach:

Distinct Networked Audio market segments:

Large scale systems:



Mid & smaller scale systems:



ARCHWAVE
connecting audio

Integrated Solutions



RAVENNA

AES67 *now!*

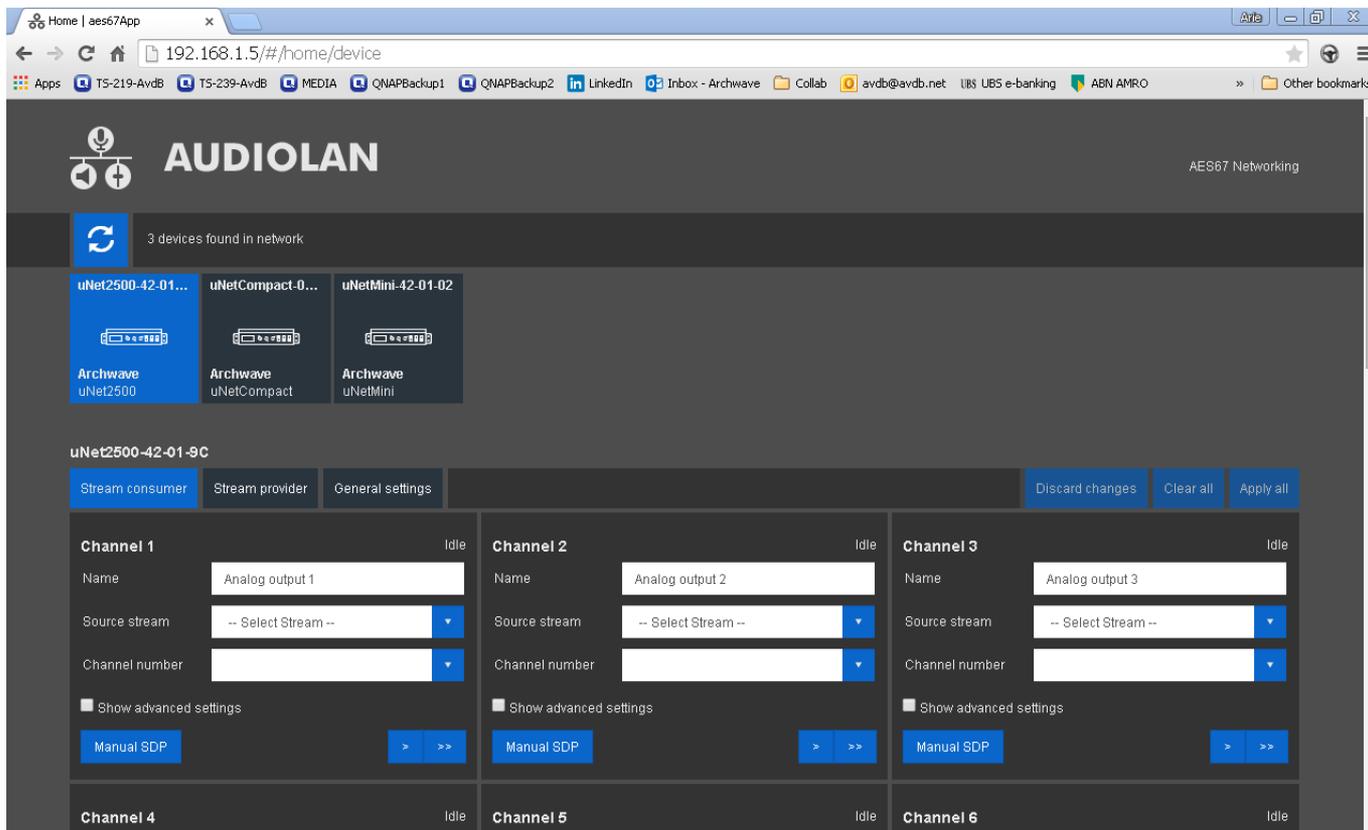
Modular solutions for Large Scale systems:

- **uRec Standard:**
 - Up to 32 channels of networked audio
 - Up to 32x32 USB-streaming
 - Up to 32 ch direct to USB media recording
 - 3-way USB-Network-AVport Bridging (*new!*)
- **uNet Compact:** Up to 16ch of networked Audio
- **uNet MINI:** Up to 4ch of networked Audio (*June*)
- **uNet Center:** Up to 64ch of networked Audio (*end 2015*)

Integrated user-friendly web interface

Targeting end-node implementations

Web interface



The screenshot shows a web browser window with the URL `192.168.1.5/#/home/device`. The page title is "AUDIOLAN" and it indicates "AES67 Networking". A status bar shows "3 devices found in network".

Three device cards are visible:

- uNet2500-42-01-...**: Archwave uNet2500
- uNetCompact-0...**: Archwave uNetCompact
- uNetMini-42-01-02**: Archwave uNetMini

The configuration page for **uNet2500-42-01-9C** is shown, with tabs for "Stream consumer", "Stream provider", and "General settings". The "Stream consumer" tab is active, showing settings for six channels:

Channel	Status	Name	Source stream	Channel number	Advanced settings	Manual SDP
Channel 1	Idle	Analog output 1	-- Select Stream --		<input type="checkbox"/>	> >>
Channel 2	Idle	Analog output 2	-- Select Stream --		<input type="checkbox"/>	> >>
Channel 3	Idle	Analog output 3	-- Select Stream --		<input type="checkbox"/>	> >>
Channel 4	Idle					
Channel 5	Idle					
Channel 6	Idle					



ARCHWAVE
connecting audio



Modular solutions for small to mid-scale systems:

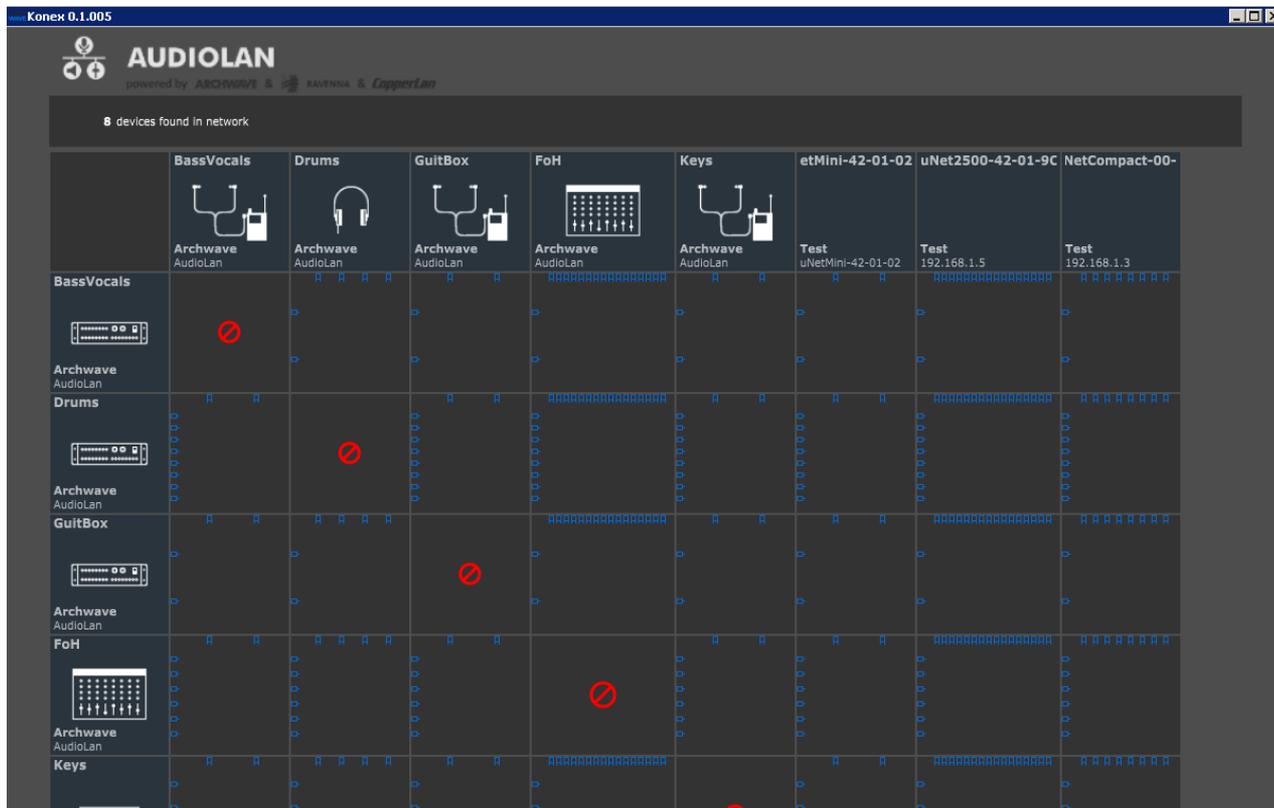
- Modular hardware solutions from 2 to 32 audio channels
- Integrated command & control based on CopperLan providing
 - Unique 3-way command & control:
 - ❖ Ravenna/AES67 compliant web interface
 - ❖ User-friendly audio matrix software (Win/OSX/IOS/Android)
 - ❖ uLink hardware-only configuration
- Full MIDI integration
- targeting complete system implementations up to 64 channels



Konnex Manager

Main page

Win/OSX/IOS/Android



Konnex Manager interface showing 8 devices found in network:

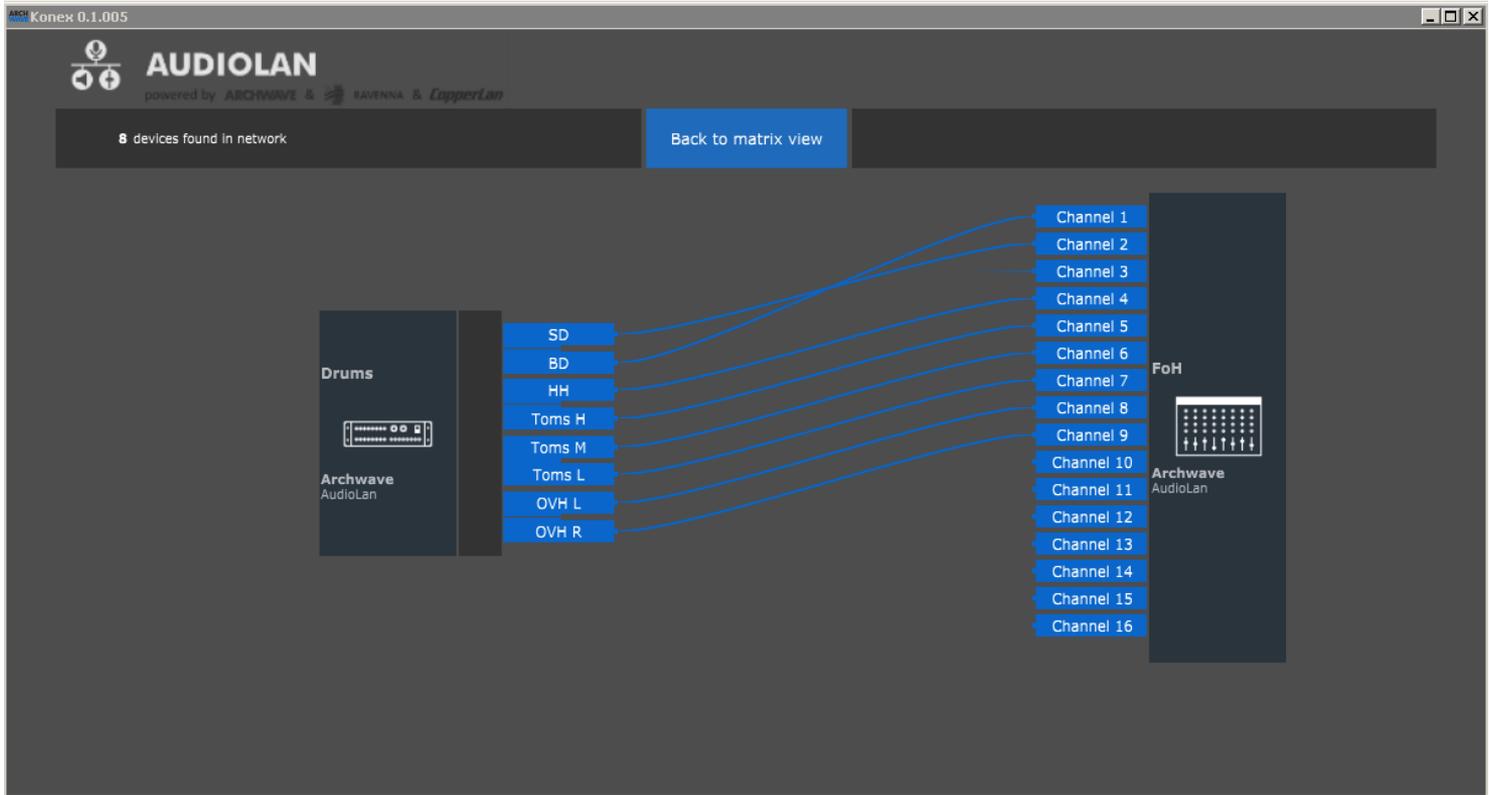
Device	BassVocals	Drums	GuitBox	FoH	Keys	etMini-42-01-02	uNet2500-42-01-9C	NetCompact-00-
BassVocals	Archwave AudioLan	Test uNetMini-42-01-02	Test 192.168.1.5	Test 192.168.1.3				
Archwave AudioLan	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗
Drums	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗
Archwave AudioLan	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗
GuitBox	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗
Archwave AudioLan	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗
FoH	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗
Archwave AudioLan	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗
Keys	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗
Archwave AudioLan	⊗	⊗	⊗	⊗	⊗	⊗	⊗	⊗



Konnex Manager

Connection page

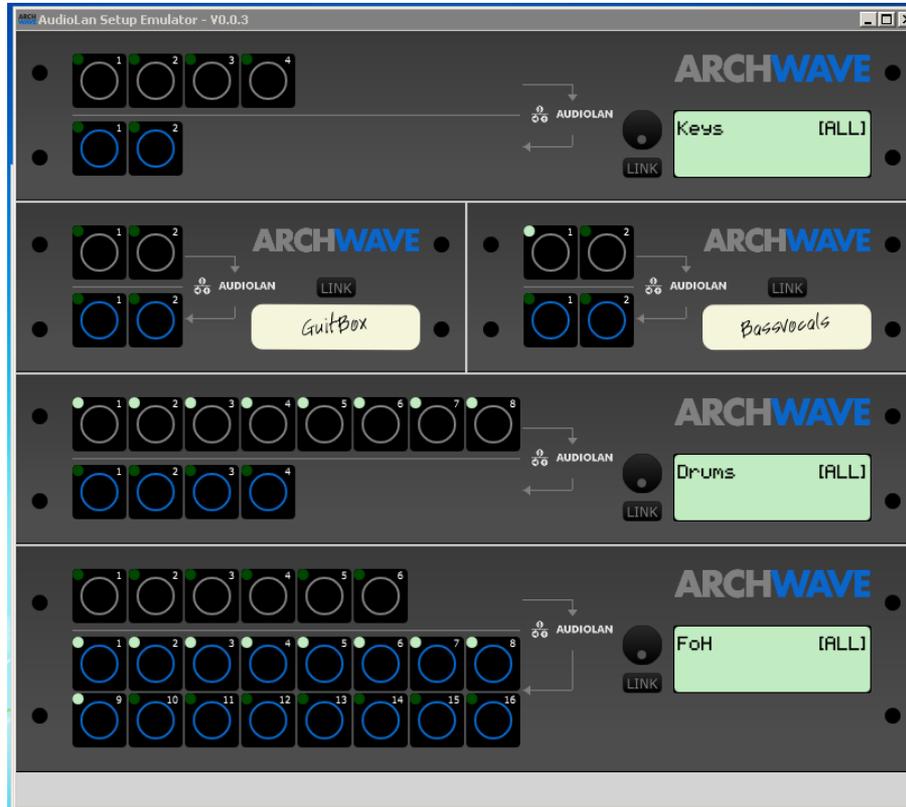
Win/OSX/IOS/Android



ARCHWAVE
connecting audio

uLink concept

Hardware only
configuration



Our Approach:

Distinct Networked Audio solutions:

Large scale systems:



Mid & smaller scale systems:



Our Approach:

Distinct Networked Audio solutions:

Large scale systems:



Range of modular network solutions:

- from 2 to 32 audio channels
- USB-Network Bridging (new!)
- targeting node implementations

Mid & smaller scale systems:



Integrated modular solutions consisting of:

- modules from 2 to 32 audio channels
- Integrated command & control
- Integrated, user-friendly matrix software (Win/OSX/IOS/Android)
- MIDI
- targeting complete system implementations up to 64 channels



Thank you !

Contact Details

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Profile

For more than 30 years, through constant innovation and the development of fruitful partnerships, Digigram has been influential in energizing the industry, raising standards, and pushing forward technological development

Positioned at the convergence of professional audio and video with IT, Digigram offers solutions that enable users worldwide to increase their competitiveness through change

Sound cards | Audio over IP codecs | Video over IP codecs | Services



Audio Networking Solutions

2001 : EtherSound©

- Proprietary ultra low latency on LAN Ethernet Level 2

2006 : Visblu Networking Audio Operating System

- Including FluidIP smart audio streaming engine on WAN-IP

2007 : Full ACIP interoperability

- EBU Tech 3326 recommendations on WAN-IP for point to point

2011 : Full commitment to RAVENNA

- Low latency synchronous audio network over LAN-IP



RAVENNA AES67 now!





AoIP through the workflow





digigram LX-IP RAVENNA / AES67 Sound Card

- Full RAVENNA and AES67 compliance
- Up to 256 RAVENNA channels In / Out from multiple streams
- Ultra-low latency: down to 1 audio sample per packet => as MADI !
- 64 / 64 MADI I/Os for seamless transition to RAVENNA (option)
- Full routing capability
- Windows and Linux drivers
- PCI express bus





ACIP ↔ RAVENNA / AES67 IQOYA AoIP codecs

- IQOYA *CALL Program Contribution codec
- IQOYA *LINK Program Distribution codec
- Fully ACIP Tech 3326 compliant stereo codecs
 - PCM 16/24 bits, ISO MPEG 1/2 layer I,II & III, Fraunhofer® MPEG-4AAC, AAC-LD, HE-AACv2, AAC-ELD, OPUS Music and Speech, e-APTX
- AES/Analog I/Os, RS232 & GPIOs tunneling
- Multiple Fail-over and redundancy schemes
- Embedded Linux/ARM pure software processing
- Now RAVENNA/AES67 I/Os

=> Perfect bridge between studio AoIP (LAN-RAVENNA/AE67) and remote Broadcasting and transmission (WAN-ACIP)

ACIP



RAVENNA AES67 now!



ACIP



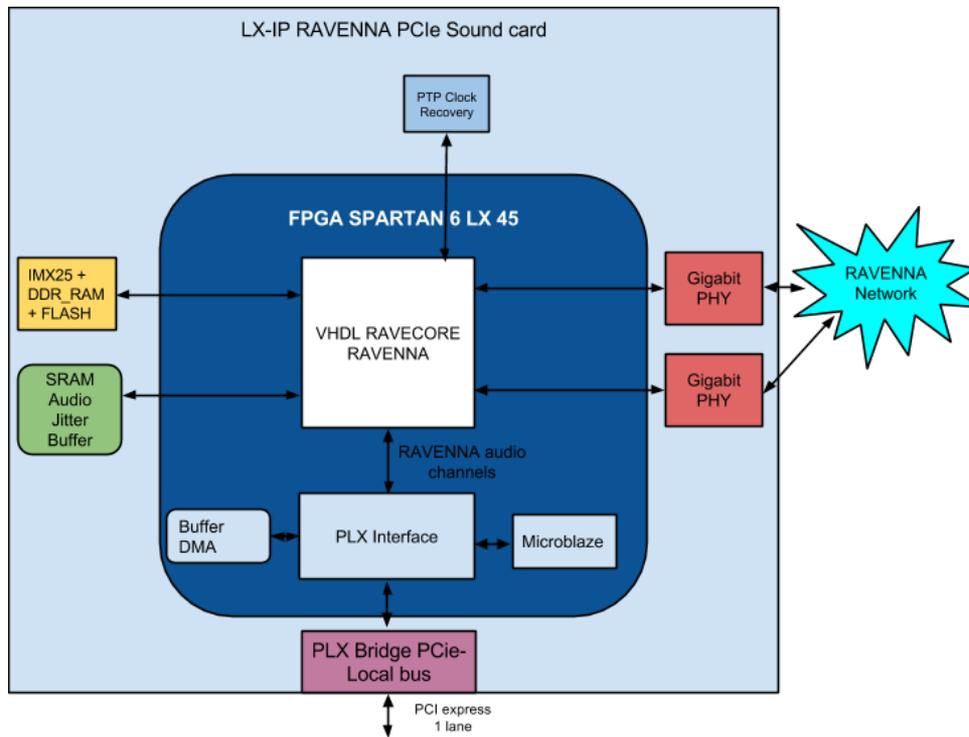
digigram

Customised AES67 OEM Solutions

- Technology derived from the Digigram products
- LX-IP RAVENNA OEM board with specific drivers and/or firmware
- Fully custom hardware design based on LX-IP reference design
- Pure embedded software solutions under Linux derived from IQOYA visiblue middleware
- Flexible business models : NRE, custom product, OEM modules, royalties ..

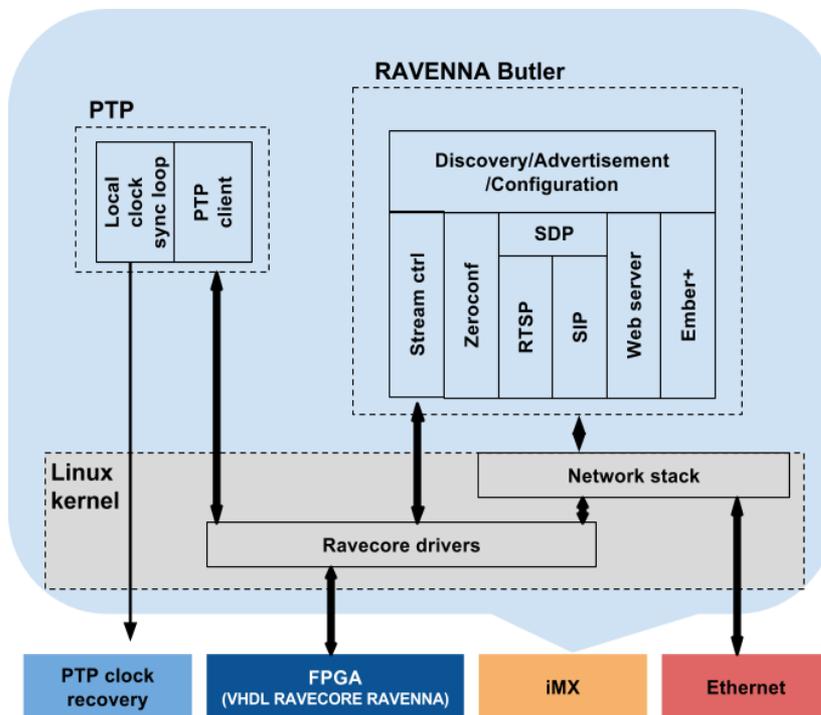


LX-IP Hardware architecture





LX-IP Software architecture





Questions ?

Thank you !!

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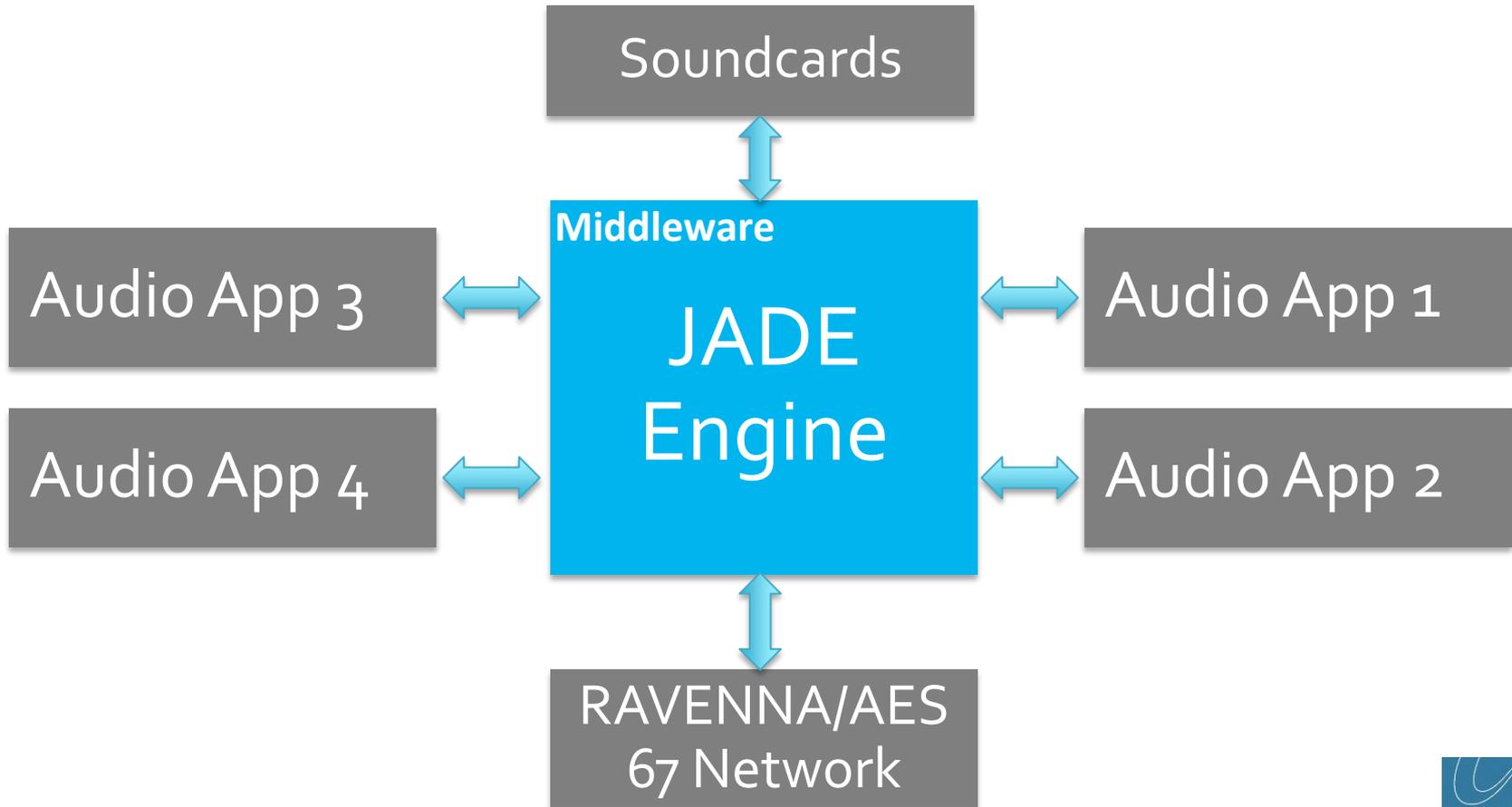
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The screenshot displays the Lawo JADE Engine software interface. The main window is titled 'Lawo JADE Engine' and features a routing console with a grid of audio components. On the left, there are several 'WDM Driver' units (Client 1-4) and 'Analogs' (1+2) with various parameters like frequency and sample rate. A 'Lawo Compressor' window is visible in the top right, showing a compressor curve and parameters like 'ATTACK' and 'LAHLY'. Below it, a 'LAWO Loudness Metering' window shows a spectrum plot and various metering parameters. At the bottom, another 'LAWO Loudness Metering' window displays six vertical meters for different audio sources: PC Audio (-20 LUFS), VLC Player (-28 LUFS), RAVENNA 1 (-29 LUFS), RAVENNA 2 (-24 LUFS), Codec (-12 LUFS), and Microphone (-20 LUFS).

JADE Engine AES67/RAVENNA for PCs

Stephan Türkay – Product Manager Radio/OnAir – Lawo AG



what is in the box ?

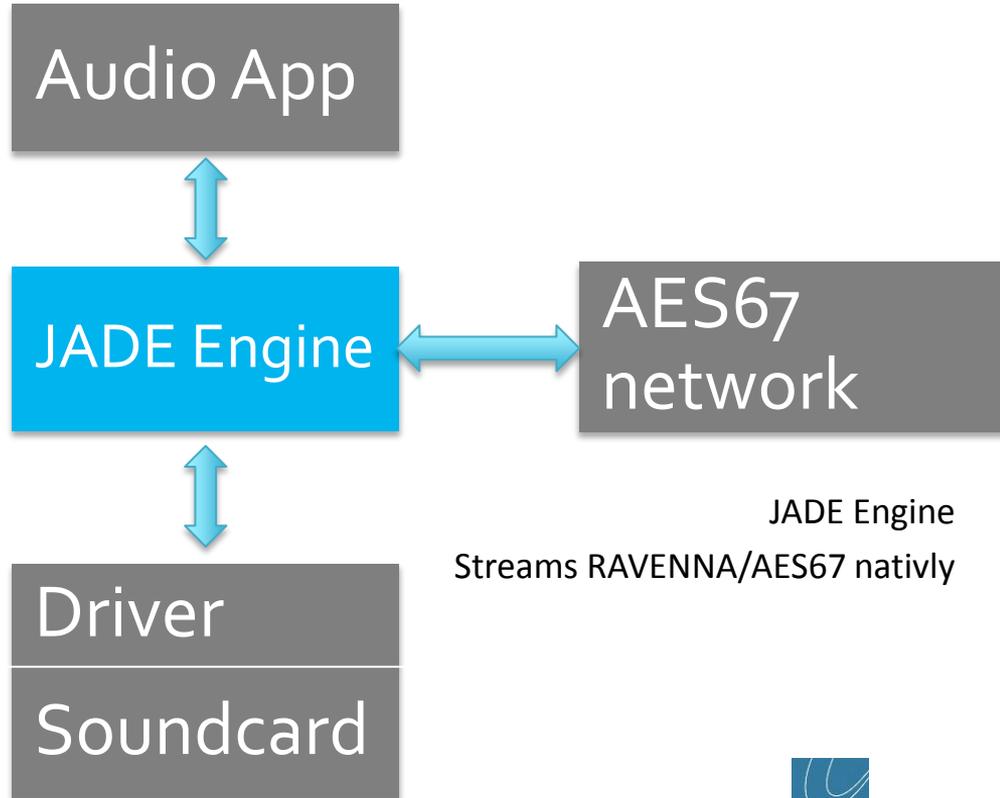
- mix and route pc audio signals
- native RAVENNA/AES67 streaming
- VST plugin host
- extensive signal metering
- save and recall complete environments
- ember+ remote control
- scripting interface



getting you connected

JADE Engine emulates multiple virtual soundcards /
hosts VST plugins

JADE Engine is able to load soundcard drivers



JADE Engine
Streams RAVENNA/AES67 natively

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THANK YOU

Agenda:

	<u>Topic</u>	<u>Speaker</u>
20 min.	Introduction to MNA	Marty Sacks (Telos Alliance) & Stefan Ledergerber (Lawo)
	AES67 – a quick Recap	Andreas Hildebrand (ALC NetworX)
50 min.	RAVENNA SoM & Reference Designs	
	AudioLAN SoMs	Arie van den Broek (Archwave)
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	Round-up / AES67 on the Show Floor	Marty Sacks (Telos Alliance) & Stefan Ledergerber (Lawo)

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AES67
Livewire+

OEM Livewire+ AES67 from Telos Alliance

Greg Shay
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The Telos Alliance
Cleveland, Ohio

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Background:

- Axia Livewire Partner Program, very successful past 10 years
- Nominal cost
- Was a reference design, with all sources, hardware schematics, FPGA, firmware, software source code.
- Linux driver source code
- However, no direct support, no design support. Online forum for answering questions about the reference design.
- The intention was the partner would digest and internalize the principles of the Livewire interface design, and make it their own.



New: OEM Livewire+ AES67

- This is a new announcement at NAB 2015 !
- The details and specifics of the deliverables and the license will be worked out in the next 6 months.
- Planned are 2 tiers, using the model of the open source industry
- Tier 1 : Reference design source files provided, very nominal cost (close to free), but no support or design assistance.
- Tier 2 : NRE cost, and generously flexible implementation royalty terms, for which we will provide implementation support and verification testing.



New: OEM Livewire+ AES67

- Livewire+ is a suite of solutions over IP, for making whole studios and facilities.
- Includes audio, control, discovery and management, all over IP.
- Includes industry standard AES67 Audio over IP.
- Includes legacy Livewire audio over IP.



AES67 intellectual property

- The Telos Alliance will indemnify partners using the OEM Livewire+ AES67 technology against any intellectual property claims.

Why are we doing this?

- We are confident AES67 is safe.
- One year ago we made this public statement.
- Nevertheless, this solution takes away any remaining shred of doubt, and allows the Telos Alliance to share its confidence with our OEM partners to proceed now with industry standard AES67.



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Sharing Experience of Implementing AES67

Greg Shay
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Cleveland, Ohio

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What does it take to implement AES67?

First by comparison:

If you have implemented Livewire, AES67 is the same basic methods, a number of details are different:

- IEEE-1588 sync is used instead of Livewire Sync, but even both of these used timestamps in clock packets.
- The clock sync PLL daemon 'xsync' was rewritten to use the new packet format, and the lower clock packet rate adjusted PLL control loop parameters.
- Packet timestamps used to phase align audio blocks between channels, instead of individual stream buffer initialization logic.





AES67 comparison to Livewire, cont.

- AES67 unicast mode requires the addition of SIP protocol to initiate audio connections (*Session Initiation Protocol*).
- Open source, high quality, proven SIP stacks available (example 'PJSIP' from www.pjsip.org)
- AES67 multicast configuration can remain simple, just like Livewire channel numbering, but use full 239.192.. multicast IP addresses.
- The AES67 mandatory 1ms packet mode, becomes a third packet size, between 250usec Livestreams and 4ms standard streams.
- Use of SDP text descriptions for stream type and packet size description actually simplifies and clarifies knowledge of types.





The fundamentals of implementing AES67:

- Software, Firmware or FPGA implementations possible, all with their own advantages and tradeoffs.
- IEEE-1588 being an industry standard, is supported in hardware in almost all of the big name modern processors and Ethernet interfaces. Freescale, Marvell, Intel, Xilinx, Altera, and others..
- AES67 shares the use of IEEE-1588 with AVB (and future SMPTE), so many 'AVB media' features can be directly used for AES67. From this point of view, and AES67 implementation can be viewed as an AVB implementation but with full IP stack headers on the packets, and without the need to use special switches for network bandwidth reservation.
- IEEE-1588 timestamping can be done directly in FPGA, too, but usually is worth choosing a component with hardware ready.





The fundamentals of implementing AES67:

- Many motherboard and server Ethernet interfaces have IEEE-1588 hardware support. Intel has been very good on this. Most of the time the OS/driver features must be turned on.





Software vs Firmware vs FPGA

- Software implementations are limited by the real time latency of the OS. Windows is typically ~ 10ms – 20ms. A software implementation may be able to take advantage of motherboard IEEE-1588 sync and drivers, and produce and consume 1ms packets, but the 1ms packets will be in groups, or bunches.
- This will require the AES67 'Link Offset' of the next device to be set higher, preventing low latency operation even with 1ms packets.





Software vs Firmware vs FPGA

Firmware:

- Many good RTOS (Real Time OS) available, suitable for implementing low latency 1ms packets and below.
- IEEE-1588 features in hardware in many microprocessors (ex: Freescale iMX6 family).
- Eventually the audio channel count may become limited by the amount of CPU consumed formatting packets, headers, moving data, and buffering many streams audio.



Software vs Firmware vs FPGA

FPGA:

- Hardware – like speed and no latency problems.
- Excellent for super low latency modes and high channel counts.
- Timestamping packets for 1588 sync can be done in FPGA.
- More difficulty implementing protocol stacks and 1588 sync control loop logic.

A combination may be optimal, and most open for different needs:

For further reference:

NAB 2013 paper, by the author:

“Taking the ‘Sting’ Out of Evolving Digital Audio Networks”

Available on www.telosalliance.com website, or on request.

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**AES67
Livewire+**

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AES67 at the NAB show:

- ALC NetworX – C2218
- Archwave – C2218
- Digigram – C1858
- DirectOut – Josephs Electronics C6948
- GatesAir – C3107
- Genelec – C3258 or C2218
- (Jutel – C2218)
- Lawo – C2218
- Merging Technologies – C3239
- Neumann – C2055 or C2218
- Orban – N5007
- Riedel – C4937
- Sound4 – C549
- Studer – C2851
- Telos Alliance – C549
- Wheatstone – C755





The screenshot shows the Media Networking Alliance website. At the top left is the logo. A navigation bar contains links for HOME, JOIN THE MNA, ABOUT US, EVENTS, NEWS, FAQ, and AES67. The main content area features a large red box with the text 'PROMOTING THE ADOPTION OF THE AES67 STANDARD ON BEHALF OF THE PRO AUDIO INDUSTRY' and a 'JOIN US' button. Below this are logos for various member companies: LAWO, BOSCH, YAMAHA, QSC, ARCHWAVE, DirectOut Technologies, THE TELOS ALLIANCE, SHURE, and ATTEROTECH. At the bottom are logos for RIEDEL, Focusrite, HARMAN, RAVENNA, and digigram. The website URL 'medianetworkingalliance.com' is displayed at the bottom right.

Media Networking Alliance

HOME JOIN THE MNA ABOUT US EVENTS NEWS FAQ AES67

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