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### Versatile Audio Network Module for RAVENNA/AES67 Ecosystems

The ZMAN family of modules is first and foremost designed to provide a high performance media transport that is tightly synchronized (ultra-low jitter and wander), deterministic, and low latency for Networked I/O end-points. What sets them apart from other offerings is the additional processing capabilities that are built-in, with 2 x ARM CPUs and serious FPGA real-estate.

Merging is strongly committed to foster the adoption of AES67 networking capability by making available, as a simple to integrate module, all required functionalities for OEMs to take advantage of this rapidly evolving market.

#### FEATURES

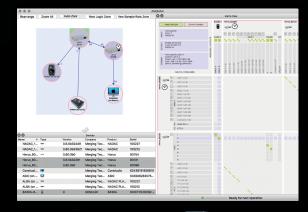
- ▲ High resolution sample rates up to 384kHz PCM, DXD and DSD256
- ▲ Up to 64 RAVENNA/AES67 streams, 256 audio channels
- AES3 I/O up to 192kHz
- ▲ Channel based audio routing (512 x 512 Matrix)
- ▲ Ultra-low latency on-chip mixing engine (128 inputs x 32 outputs)
- ▲ 28 freely assignable EQ bands
- On-board ultra-low jitter and ultra-low phase noise clock
- ▲ Fully compatible RAVENNA/AES67 protocols over Gigabit Ethernet
- Web-based network remote control

#### BENEFITS

- Compact Mezzanine board, small form factor 59.6 x 44.5 mm
- ▲ Ultra-low phase noise clock
- PTPv2 Master or Slave, IEEE-1588-2008 standard
- Support for industry standard SMPTE 2110
- Automatic device network discovery (Bonjour)
- Channel based internal routing
- Built-in ARM processors + abundant FPGA-based DSP in a single chip
- Compatible with any RAVENNA/AES67 device on network
- SDK and API for easy integration
- Use case examples through configuration scripts
- ▲ Simple firmware update via network
- Optimized compatibility with Merging Technologies products portfolio

#### SOFTWARE TOOLS & DRIVERS

- Windows® ASIO driver 🔺
- Mac® OSX CoreAudio driver 🔺
  - Linux ALSA drivers 🔺
- ANEMAN (Audio Network Manager) full support 🔺



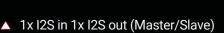


#### AUDIO SPECIFICATIONS

- Sample Rate 44.1kHz/48kHz, 88.2kHz/96kHz, 176.4kHz/192kHz
- Optional high resolution support for DXD, 384kHz, DSD64, DSD128 and DSD256
- ▲ Word lengths 16, 24, or 32 bits per sample
- ▲ I2S/TDM Audio format
- Network input audio buffer up to 16k samples
- Up to 64 RAVENNA/AES67 I/O streams, up to 256 network audio channels
- Word clock IO for synchronization

#### HARDWARE SPECIFICATIONS

- Single Power supply 3.3V, under 5W 🔺
  - Xilinx SoC Zyng based design 🔺
- Dual core ARM Cortex A9 processors, ARMv7-A architecture
  - DDR3 Memory (512 MB) 🛆
    - NOR Flash (128 MB)
- Standardized RGMII interface for Gigabit Ethernet switches or PHY, IEEE Std 802.3
  - High quality, on board clock management 🔺
    - Mezzanine connectors (3 x 80 pins) 🔺
      - Card edge test header (Mini-PCI) 🔺



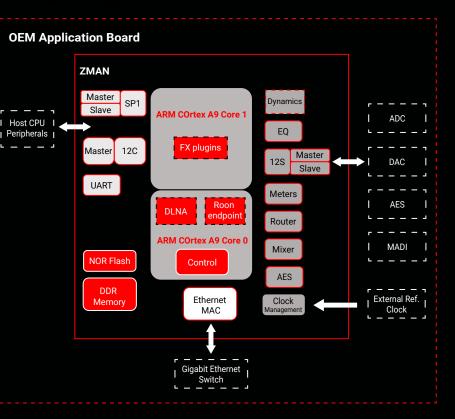
INTERFACE SPECIFICATIONS

- A 1x I2S in 1x I2S out (Master/Slave) with 8 data lines each
- 🔺 1 x I2C Master
- ▲ 1 x SPI Slave and 1 x SPI Master
- 🔺 1 x RGMII
- 1 x UART (up to 921'600 bauds)
- 🔺 GPIO
- Differential clock input to support OCXO, Atomic clock or GPS disciplined oscillator

#### **PRICE-PERFORMANCE OPTIONS**

#### ZMAN 010

- ▲ 64 x 64 I/0@1FS
- 80 DSP slices (FPGA)
  - ZMAN 020
- 🔺 256 x 256 l/O@1FS
- 220 DSP slices (FPGA)



## ZMAN + ZOEM

#### Audio Networking and Processing development kit. The quick-start Platform for initial Evaluation

#### EVALUATION KIT ZOEM FOR ZMAN

- Single power supply
- Easy mezzanine prototyping
- Fully exposed GPIO and audio I/O
- A On board Gigabit Ethernet switch and PHY
- ▲ Dual Ethernet copper ports & SFP Fiber port
- ▲ 1 stereo AES-EBU input and 1 stereo AES-EBU output
- Debug console available on standard micro USB port
- Out of the box DAC-like Web remote UI

