

# **Portable Compact Mixer-Recorder**



- Inputs

- o 8 Total analog inputs: 6 mic-line inputs, 2 on returns

o Com Rtn (TA3, 3.5 mm) balanced, 1-channel, 8k ohm input impedance

o External Slate Mic (TA5): balanced, 8k ohm input impedance, menu-selectable 12 V phantom

- Maximum Input Level o Mic: +8 dBu (2.0 Vrms)
- Aux: +18 dBu (6.2 Vrms)

- 6 Buses (L, R, 1-4)

Limiters

**Buses** 

• Analog first stage, all subsequent stages digital Attack time: adjustable 1 to 200 ms

• Limiters available at all channels, buses, headphones, for all sample rates

- Compressors
- Compressors available at all channels (pre or post-fade) and buses for all sample rates

• Threshold: adjustable, 0 dBFS to -40 dBFS

• Attack time: adjustable, 1 to 200 ms • Release time: adjustable, 50 ms to 1000 ms

## • Selectable ratio: adjustable, 1:1 to 20:1 · Knee: soft, hard

- Delay
- Channel Adjustable 0-50 ms

Fader stage: 16 dB Bus stage: 16 dB

• Trim stage (mic input): 76 dB Trim stage (line input): 50 dB

- Mic-to-Headphone: 112 dB
- Outputs

Maximum Output Level (all into 10k load)

Line: +20 dBu (7.8 Vrms)

 XLR (L, R) active-balanced, 250/3.2k/120 ohms (mic/-10/line) TA3 (X1/X2) active-balanced, 250/3.2k/120 ohms (mic/-10/line) o 3.5 mm (X3/X4): unbalanced, stereo, 1.8k ohms

## o "-10": +6 dBu (1.5 Vrms) Mic: -20 dBu (0.078 Vrms)

Outputs

- o X3/X4 Out: +6 dBu (1.5 Vrms) o Headphone outputs (¼", TA-5): +14 dBu (4.0 Vrms)
- ¼", 3.5 mm • TA5 (along with mic input pins) for single connection to headset + mic • High output, 4 ohm output impedance, 400 mW + 400 mW at each connector, all individually driven

**Headphone Outputs** 

Recording

· Compatible with headphones of any impedance

## Bit Depth 0 16, 24 Recording

exFAT formatting

**Automatic Mixing** 

• Attenuation range: 0-20 dB

o 12 tracks (8 iso channels, 4 buses)

o AAC 2 track at 48 kHz, selectable bit rate 32, 64, 128, 192, 256 kbps

• Dugan Automixer up to 8 channels on Left and Right Mix bus

o 64-bit WAV (RF64) monophonic and polyphonic; support for files > 4 G

o Broadcast WAV monophonic and polyphonic file format

• NoiseAssist operates with sampling rates of 44.1 kHz to 48.048 kHz. • CEDAR sdnx operates with sampling rates of 44.1 kHz to 96 kHz.

• CEDAR sdnx audio path latency: 0.27 ms @ 48kHz, 0.14ms @ 96kHz

• NoiseAssist audio path latency: 0.77 ms @ 48kHz

# • USB-C 2-in/2-out USB audio interface · HUSB-A host for keyboard, external controller, external USB hubs supported for connecting multiple devices.

**USB** 

Timecode and Sync · Modes Supported: Off, Rec Run, Free Run, 24h Run, External, including External Auto-Record and Continuous modes. • Frame Rates: 23.98\*, 24, 25, 29.97 DF, 29.97 ND, 30 DF, 30 ND

• Sample/Timecode Accuracy: 0.1 ppm (0.25 frames per 24 hours)

• Timecode Input: 20k ohm impedance, 0.3 V - 3.0 V p-p (-17 dBu - +3 dBu)

• Word Clock Input: 10k/75 ohm selectable impedance, 1-5 V p-p input sensitivity

- SD-Remote Android Phone app via Bluetooth LE • SD-Remote iPad and iPhone app via Bluetooth LE · USB Keyboard • External Timecode Record Trigger

• SD-Remote Android Tablet app via USB or Bluetooth LE

- Dual rear-mount Sony-style L-mount batteries with chargers
- o All mic preamps off: 730 mA o All mic preamps on: 920 mA
- 5.1 cm x 22 cm x 17 cm;
- Power • External: 10-18 V input on locking TA4 connector, (pin 4 positive, pin 1 ground), supports Smart Battery telemetry
- · Current Draw, at 12 V no battery charging
  - Storage: -40° C to 85° C Dimensions (H x W x D)

LCD

- Operating: -20° C to 60° C, 0 to 90% relative humidity (non-condensing)
- 2.0 in. x 8.7 in. x 6.6 in
- 1.25 kg (unpackaged, without batteries)
- Weight
- 2.75 lbs (unpackaged, without batteries)

· MixAssist up to 8-channels on Left and Right Mix bus

- **Remote Control** • Sound Devices CL-12 Linear Fader Controller
- 320×240, Transflective, excellent sunlight visibility • Larger touchscreen display available via USB-connected SD-Remote app
- **Environmental**

- Equivalent Input Noise

- **Processing Engine** • Highly extensible, full FPGA-based audio processing, 3 FPGAs
- o -131 dBV (-129 dBu) max (mic in, A-weighting, 76 dB gain, 150 ohm source impedance)
- Six-way ARM multiprocessor system · 64-bit audio processing precision
- Mic/Line inputs o 6 total, all fully featured; 3 on full-size XLR, 3 on TA3
  - o Mic-level inputs: (XLR, TA3): Class-A, discrete differential long-tail pair, 4k ohm input impedance o Line-level inputs: (XLR, TA3,): active-balanced, 4k ohm input impedance o 48 V phantom: full 10 mA to all 6 inputs simultaneously
  - o AES3 or AES42 available on XLR input 1 o AES42: +10 V, 250 mA available, mode-1, auto-ASRC o USB Audio: 2 Inputs o Aux (3.5 mm: unbalanced 2-channel, 4k ohm input impedance
  - o Line: +28 dBu (19.5 Vrms) o Com Rtn: +24 dBu (12.3 Vrms) External Slate Mic: +12 dBu (3.2 Vrms)
- Buses 1-4 can receive pre-fade, post-fade, or independent send level from isolated channels, Return, and Com Return. **High-Pass Filters** • Adjustable 10 Hz to 320 Hz, 18 dB/oct. 1st stage analog (before preamp), 2nd stage digital.

Left and Right Mix Bus receives post-fade isolated channels. Optional NoiseAssist plugin instances can be applied to any bus.

- Release time: adjustable, 50 ms to 1000 ms • Threshold: adjustable, -2 dBFS to -12 dBFS Selectable ratio: inf:1, 20:1, 18:1, 16:1, 14:1, 12:1, 10:1 · Knee: soft, hard
- Output Adjustable 0-500 ms **Maximum Gain**
- Headphone stage: 20 dB Mic-to-Line: 108 dB TA5 (along with mic input pins) for single connection to headset + mic High output, 4 ohm output impedance, 400 mW + 400 mW at each connector, all individually driven Compatible with headphones of any impedance
- Digital Outputs · AES3 transformer-balanced, in pairs; 1-2 (XLR-L), 3-4 (XLR-R), 110 ohm, 2 V p-p, AES and S/PDIF compatible

Two, four, or eight instances of Noise Suppression can run on any combination of isolated channels (excluding 17-32 on Scorpio), or buses.

• A/D converters |32-bit, 120 dB, A-weighted dynamic range typical

A/D converters |Sampling rates 44.1 kHz, 47.952 kHz, 48 kHz, 48.048 kHz, 96 kHz, 192 kHz

- o Internal 256 GB SSD; two removable SD Cards, 10% over-provisioned for optimum performance o Simultaneous recording to internal SSD and the two SD cards
- **Noise Suppression** Via optional paid Sound Devices NoiseAssist or CEDAR sdnx Plugins
- USB-C (USB 3.1 type 1) for file transfer of internal SSD, both SD Cards.

• Word Clock Output: 75 ohm impedance, 5 V p-p output, at SR

• USB MIDI MCU Control - supported 3rd party fader controllers

• Timecode Output: 75 ohm impedance, 5 V p-p (+7 dBu)

- **File Delivery to Cloud** • Compatible with Viviana Cloud
- o All mic preamps on, 192 kHz sample rate, recording to internal SSD and 2 SD Cards: 1.07 A · Intelligent power-down of unused mic preamps and other internal circuits

**Specifications Analog Inputs** • Frequency Response  $\circ$  10 Hz to 80 kHz  $\pm$  0.5 dB (192 kHz sample rate, re 1 kHz) o 0.005% max (mic in, 1 kHz, 22 Hz-22 kHz BW, trim at 20, fader at 0, -10 dBu in)