

DSF66

Dual digital audio upmixer and downmixer based on SoundField® algorithms

A Synapse® product







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AUDIO PROCESSING

Block schematic & I/O panel



AES/EBU IN/OUT 1/2 \bigcirc \bigcirc \bigcirc \bigcirc AES/EBU IN/OUT 3/4 \bigcirc 0 AES/EBU IN/OUT 5/6 Ō \bigcirc AES/EBU IN/OUT 7/8 METADATA IN/OUT 0 0 0 AES/EBU IN/OUT 9/10 AES/EBU IN/OUT 11/12 AES/EBU IN/OUT 13/14 \bigcirc AES/EBU IN/OUT 15/16

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Features

The DSF66 is a dual hardware stereo-to-5.1 upmix and downmix processor, designed for HD broadcasters who use a lot of archived stereo material and wish to generate acceptable 5.1 broadcast mixes from stereo soundtracks. The DSF66 has a shared metadata input and shared physical I/O.

Software and hardware upmixing tools have existed for some years, but most of them create material for the extra three channels in a 5.1 mix by using processing, for example adding reverb or applying phase-shifts to the stereo material to create information for the rear surround channels. Instead, the DSF66 generates the material for the extra channels by closely analyzing the source stereo signal over time. Using a unique algorithm developed for the purpose, the DSF66 can detect reverberant content in the stereo signal, differentiate it from the direct sounds in the mix, and separate it out.

Users can adjust the details of the processing directly from the DSF66 GUI in Cortex, with control offered over a variety of different parameters including the level of the direct and ambient components in the front and rear channels, and the divergence of the Centre channel in the generated 5.1 mix, with options from a discrete Centre channel at one extreme to a phantom Centre at the other. Output level controls are also offered for each of the channels in the final 5.1 mix.

The I/O configuration of the DSF66 can be changed to allow for convenient connection of external inputs, outputs or a combination. In standalone mode the DSF66 has 4 AES/EBU inputs and 4 AES/EBU outputs (8 mono in and 8 mono out). In (quad speed) ADD-ON mode the unit can be configured as 8 in (16 mono in) or 8 out (16 mono out) to connect either 16 mono external source channels listen to 16 external channels in 8-out mode. If processing of embedded audio from a master card is required the unit can be used without physical I/O and all channels are routed from and to the Quad Speed Audio bus

- 3 physical I/O modes 8-in, 4-in + 4-out, or 8-out
- Output gain and delay adjustments
- 2x Upmix stereo to 5.1
- 2x Downmix from 5.1
- Cross fading between upmixed and discrete 5.1 (5.1/2.0 input auto-sensing)
- Cross fading between downmixed and discrete 2.0 (5.1/2.0 input auto-sensing)
- 8 presets for convenient storing of use cases.
- Locks to Black & Burst, AES input and Mastercard.
- Full control and status monitoring through the front panel of the SFR04/SFR08/SFR18 frame and the Ethernet port (ACP)

Applications

- High-End upmixing and downmixing of broadcast related program audio streams
 - To ensure a constant 5.1 + Stereo output regardless of input format
 - Simultaneous independent Upmixer and Downmixer (LoRo or LtRt)

Ordering information

Module:

DSF66: 16 channel processing (dual 5.1/2.0) digital audio upmixer/downmixer ADD-ON card

Standard I/O:

BPL11_DSF66: I/O panel for DSF66 with unbalanced AES/EBU in and out

Relay bypass I/O:

 BPX11_DSF66: I/O panel for DSF66 with unbalanced AES/EBU in and out and bypass relays

Specifications

AES Audio Input	
Connector	BNC 75 Ohm
Standard	AES-1992 for synchronous or asynchronous
	PCM/AES, SMPTE 276M for single ended
	synchronous or asynchronous PCM/AES
Number of Inputs	4 (8)
Sampling Rate	32 kHz to 192 kHz Synchronous 48 kHz in
	Master/ADD-On mode
Resolution	24 bits when AES inputs selected, 20 bits in
Minimum Innut (Output Delay	Master/ADD-ON mode
	75 Onms
Level	0.2V to 1V nom
AES Audio Output	
Number of Outputs	4 (8)
Connector	BNC,
Resolution	24 bits
Sampling Rate	48KHz synchronous
Minimum Input/Output delay	IBD
Maximum Input/Output offset delay	4000 ms
Peference Input through PPC	
Number of Inputs	2 on SED18 2 on SED08 and 1 on SED04
Ri-level	PAL Black Burst ITU624-4/SMPTF318 Composite
	NTSC SMPTE 170M
	1Vp-p nominal, 75 Ohms terminated through loop
Word clock	AES11-2003 Annex B, not terminated on loop
	48kHz
Miscellaneous	
Weight	Approx. 250g
Operating Temperature	$0 ^{\circ}C to +50 ^{\circ}C$
Dimensions	137 x 296 x 20 mm (HXWXD)
Flectrical	
Voltage	+24V to $+30V$
Power	-24 V 10 + 30 V < 10 Watts
	NTO Walls