

TDA1541A Balanced Driver

What it does

This circuit takes I2S data (with any number of bits per frame) and splits up the stream so that it can feed 2 x TDA1541As running in simultaneous data mode. The output of the circuit drives a channel of one TDA1541A with L+ and the other channel with L-, so that the DAC is producing L +/- balanced output. The other TDA1541A is operated in a same manner, but with R +/- balanced output.

Some things to consider:

- The output current swing of the DAC, the circuit here is splitting the phase in the digital domain.
- The digital representation of audio, remember that both offset binary and 2' complement are asymmetrical in terms of the largest positive and largest negative integer they can represent. Unlike sign-magnitude, there is no -0. To invert the phase of the digital audio, you DO NOT add 1 after the bit inversion.
- All output from this circuit is delayed by exactly one cycle of the input bit clock BCK. This is why you see `data_l_n` and `data_r_n` in phase with `data` .

How it does it

This is just taken from my previous posts on diyAudio, and may give some hints, it does not necessarily apply to this circuit, although the principal of operation hasn't change:

"When input OB/TWC input is connected to VDD1 the two channels of data (L/R) are input simultaneously via DATA L and DATA R, accompanied with BCK and a latch-enable input (LE). With this mode selected the data must be in offset binary. "

If you look at the timing data below, you'll see that LE goes high after the LSB has been clocked in, and loads the data from the shift register used to receive it internally. Now from I2S, we are able to select left/right data based on WS. So for example if WS is low, we select left, and feed I2S - BCK to LCK (the burst wave). With it goes the data. If the LCK clock isn't ticking, it won't load any data in, that's why it remains low (after the falling clock-in edge), at this time the right one takes over. Now with I2S, WS changes before the LSB is clocked in (rising edge) , with the simultaneous mode we want the latch to open after the LSB has been clocked in (on the falling edge), so the WS signal is delayed appropriately.

The timing chart below shows that LE must close before the data for the next word is clocked in (starting with the MSB), and this is fine, cause the clock LCK/RCK don't tick in the period where RLE /LLE is high (latch open) and no data is clocked in. Before the clock starts up again, LLE or RLE close, so that the data may be loaded.

As for the two's complement to offset binary conversion, this can simply be done by inverting the MSB of each word. When WS on the I2S bus changes, the next rising edge of BCK is the LSB of the pervious word, and on the rising edge after that, is the MSB of the new word. We use WS and a few D-F/F to find the MSB, ie. a signal that goes high when the MSB of a word is present in the stream. And, pass the DATA and the find-MSB signal through a XOR gate. The XOR gate will invert the DATA when the find signal is high, therefore inverting the MSB, doing tow's complement to offset binary conversion.

The final stage uses a counter to count in the first 16 bits from each channel, and this must be done to stop the remaining bits from flushing out the data. Different sources can have different frame widths.

Disclaimer

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