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## Now I tried to build 24 bit DAC without oversampling. I found only one circuit PCM1704 from Texas Instruments which satisfy these requirements. It looks like his production now slowly ending. It is very expensive and because it has only single channel

Introduction

that there are two pieces on the board for stereo. It should be the best what can be builted for playing of 24bit records. **Circuit description** In supply part I used two bridge rectifiers. Theoretically one is enough, but I wanted back compatibility with other boards, where

## transformer winding for positive branch can be connected also to S/PDIF decoder. Because I want small temperature losts I used transformer with two 9V secondary windings. Because small negative voltage regulators are not available, that I used classic

78XX and 79XX voltage regulators. Capacity of C1 and C2 should be enough thanks to very small current consumption of circuit. Heatsink is not needed. All voltage regulators are properly blocked with 100nF ceramic capacitors and outputs are filtered with C3 to C6 electrolytic capacitors. Every convertor and operational amplifier has on the power line pins additional capacitors recommended in the datasheet. Convertor accepts data in the 24bit right justified format. I had to connect some logic on the data input. This logic counts with data in the I2S format where data are first delayed for 7 bits with shift-register for correct aligning to 24bit. On the I2S bus are data delayed for 1 bit and left-justified. Thanks to this format we can play data in 16 bit or 24 bit format without any settings change.

Next we must buffer 32 bits of data for left channel in four 74HCT164 8 bit shift-registers because we need to play both channels

simultaneously and without time-shift. Instead of five pieces of 74HCT164 can be used Xilinx CPLD circuit like in devilsound

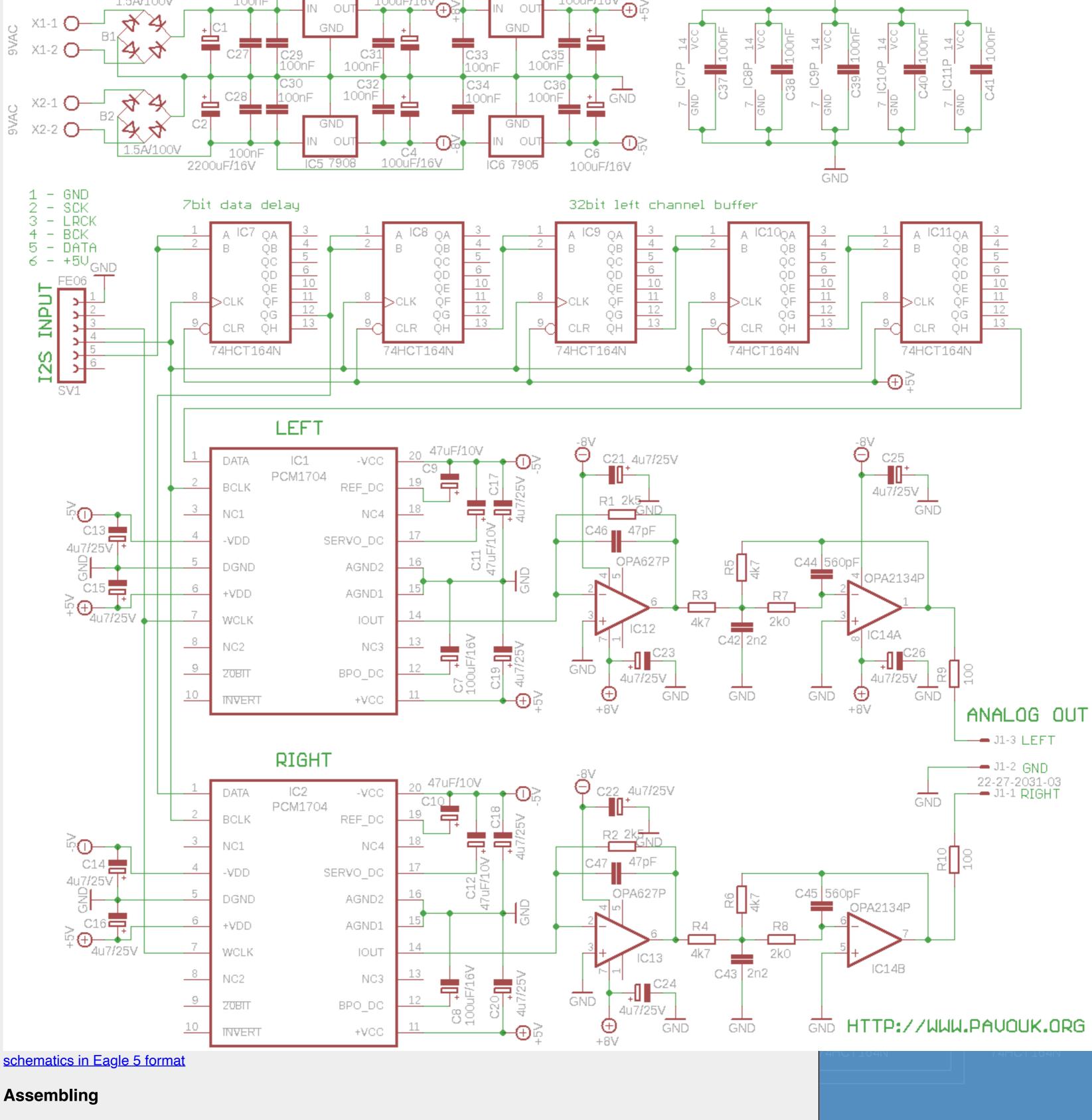
DAC, but I haven't any experience with Xilinx programming now. DACs have current output which we must convert to voltage output with I/V convertors. I used recommended circuit with lownoise Difet operational amplifiers OPA627. On their output is connected 2nd order low-pass filter with dual operational amplifier OPA2134 with very low distortion according to recommended circuit in datasheet. On the filter output is signal inverted which is OK in the music, when it is on both channels. If we want to connect DAC without output filter, that we don't assemble IC14 and capacitors C42 and C43. Resistors R3 to R6 must be replaced with zero resistors. We needn't to assemble R7, R8, C44, C45, C25 and C26. But this is not important, because

**Schematics diagram** PCM1704

C5 100uF/16V ⊕⊊ 1.5A/100V OUT

# X1-2 (

in the wiring without filter they are not used.



**Component view** 

Top side

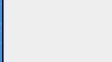
We assemble components on the board from smaller to bigger. We must begin with DACs IC1 and IC2 because later it will not be easy to install them. Next we place all SMD capacitors and resistors. We continue from the top side with integrated circuits IC7 to

IC11 and sockets for IC12 to IC14. Next we can assemble all electrolytic capacitors. Beware their polarity! On the end we

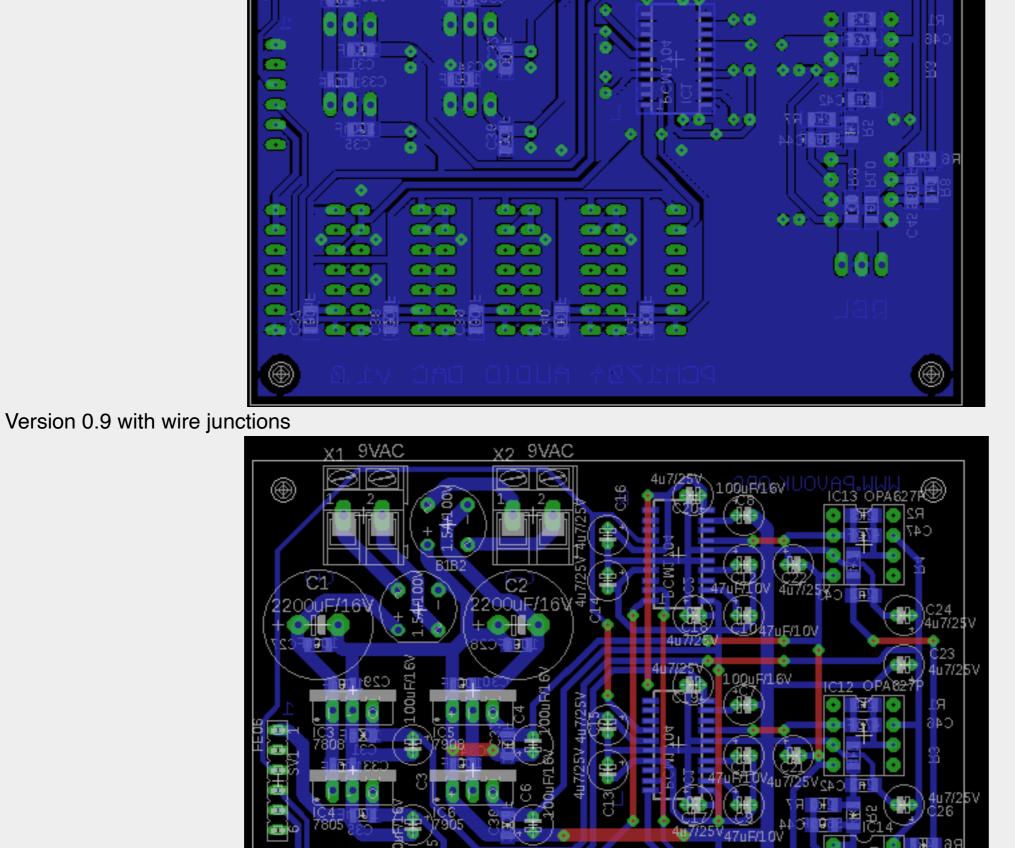
assemble voltage regulators, rectifiers and all connectors. With careful assembling circuit works on the first power-on.

X2 9VAC

X1 9VAC



Bottom side



Printed circuit board

909 00000 0000000 $\bigcirc$  $\odot$ LGR

Top side of board version 1.0 P PR SHALLAR JULY CPS PCM1704 AUDIO DAC V1.0 PDF format Eagle5 format

small number of boards. After finish of assembling I cleaned PCB from flux with Isopropyl-alcohol.

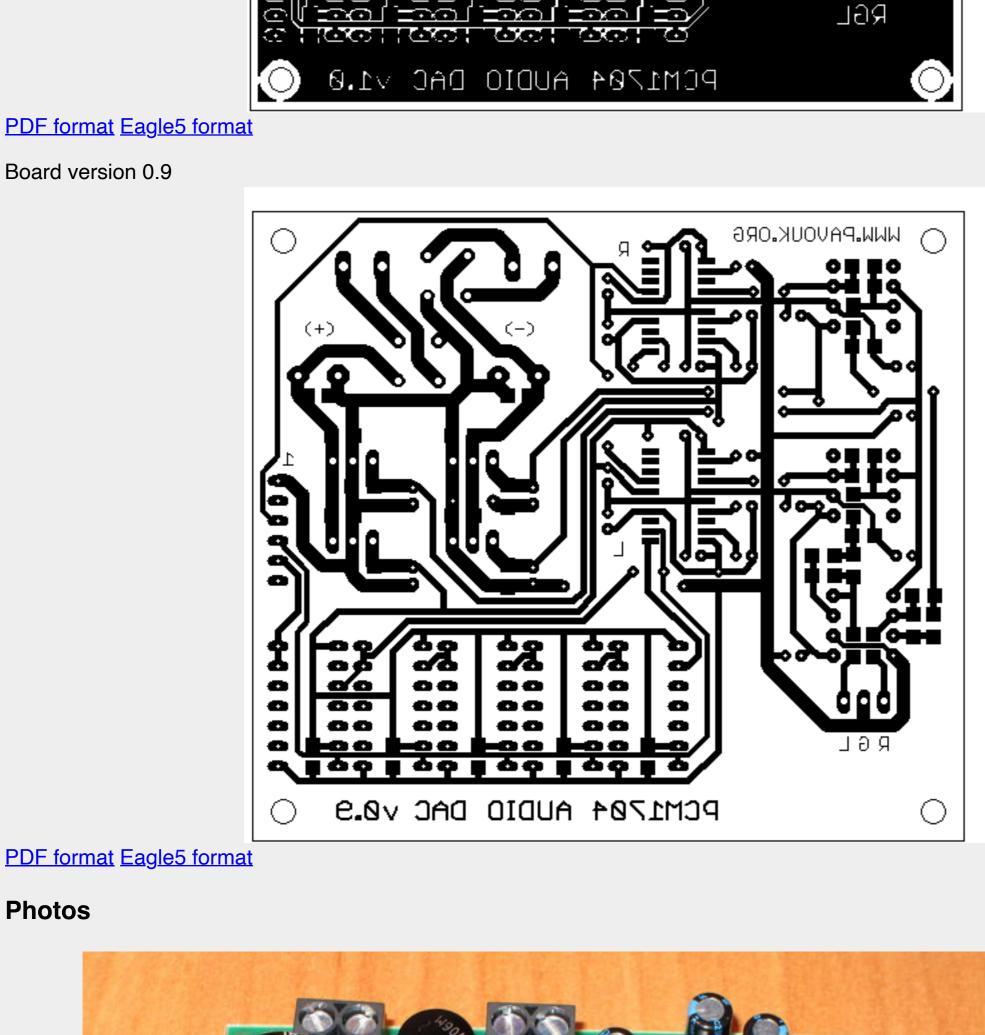
PCM1704 AUDIO DAC v0.9

Printed circuit board was this time more complicated with 13 wire junctions. In regard of high price of components I estimated that

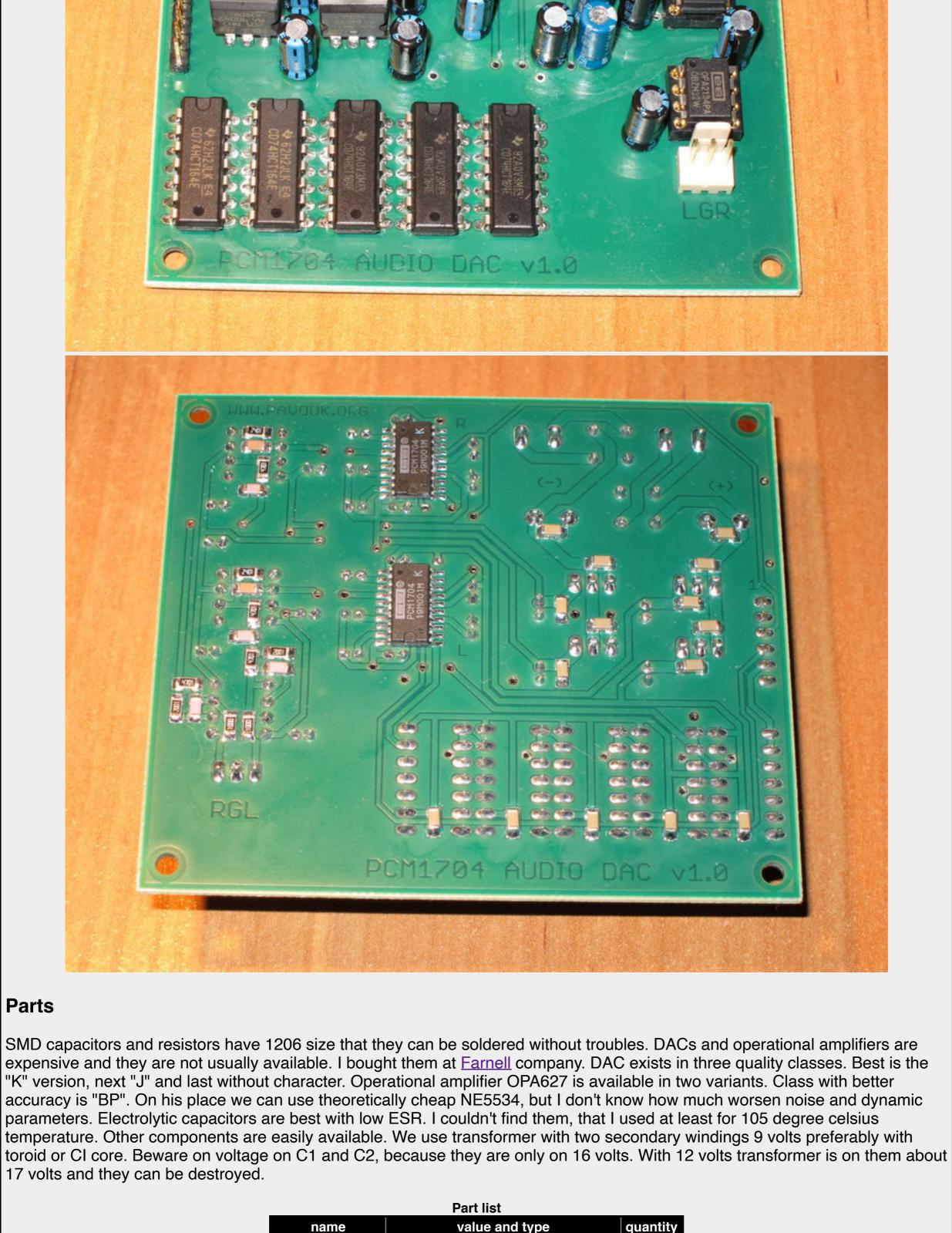
price of PCB will be minor in all costs and I decided to made professional double-sided board with ground planes and masks. However there is still available single-sided PCB version 0.9 which is in other aspects identical to version 1.0. Routes are still very wide, that worse accuracy class is enough for our PCB. In design I tried to have good separation of analog and digital part and right connection of ground and power lines for minimizing of noise and interference. Data for professional manufacturing of PCB we can generate from Eagle with CAM processor. They are not attached, because every manufacturer has a little different requirements. Usually is generated top and bottom side, board outline and top and bottom mask in GERBER\_RS274X format. Drills are generated in EXCELLON format. I ordered board in special prototype service at MEV company which is cheaper for

Board version 0.9

Bottom side of board version 1.0



**Photos** 



**Parts** 

B1, B2 Bridge rectifier 1.5A/100V 2x C1, C2 2200uF/16V low ESR electrolytic RM5 2x C3-C8 100uE/16V low ESB electrolytic BM2 6v

C3-C8	100uF/16V low ESR electrolytic RM2	6x	
C9-C12	47uF/16V low ESR electrolytic RM2	4x	
C13-C26	4u7/50V low ESR electrolytic RM2	14x	
C27-C41	100nF SMD1206 ceramic	15x	
C42, C43	2n2 SMD1206 ceramic	2x	
C44, C45	560pF SMD1206 ceramic	2x	
C46, C47	47pF SMD1206 ceramic	2x	
IC1, IC2	PCM1704U-K(U-J, U)	2x	
IC3	7808	1x	
IC4	7805	1x	
IC5	7908	1x	
IC6	7905	1x	
IC7-IC11	74HCT164(HC, ACT) DIL14	5x	
IC12, IC13	OPA627BP(AP)	2x	
IC14	OPA2134PA	1x	
J1	Molex 3pin	1x	
R1, R2	2k5 SMD1206	2x	
R3-R6	4k7 SMD1206	4x	
R7, R8	2k0 SMD1206	2x	
R9, R10	100R SMD1206	2x	
SV1	Jumper ribbon 6 pins	1x	
X1, X2	Frame terminal AK300/2	2x	
0 1 110101011	Precise DIL8 socket	3x	

### • http://www.halidedesign. http://tech.juaneda.com/e http://www.ti.com/lit/gpn/ • http://www.ti.com/lit/gpn/opa2134 OPA2134 datasheet.

End

Links

This construction of DAC with PCM1704 circuits is for the present most expensive from all my DAC constructions but it has a highest quality. I looked on the output signal with oscilloscope and I simple didn't see any interference from digital part to analog. Output sounds for me more dry than with AD1865 based DAC. Reason for this will be probably used low-pass filter. Output of DAC after I/V convertor before filter is absolutely perfect and now best what I see and hear from my constructions. Nothing is added and nothing is lost. In datasheet is written, that maximum sample frequency of PCM1704 is 96kHz with 8x oversampling. I succesfully tests DAC in 24bit/192kHz mode. According to other document from TI is maximum sample frequency 768kHz which is probably true. On the output is not present capacitor. I measured on the one output DC offset 1.6mV and 1.9mV on another, which is probably OK. It is possible to combine DAC with both types of S/PDIF decoders and with USB DAC with I2S output. Unfortunately circuit has now status NRND, that it is not recommended for a new constructions. I am a little sad, that on market wins price over quality. **Updates** 

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