TWO-STAGE DUAL-CHANNEL HEADPHONE AMPLIFIER

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ABSTRACT

We developed a two-stage, dual-channel audio amplifier targeted for the Sennheiser HD650 headphones with 300Ω impedance and 98 mW/dB sensitivity. HSPICE testing demonstrates that it meets the expected design specifications for gain, 3dB midband, input & output impedance, noise, and power output. Measurements made on a soldered version of our design match up with expectations.

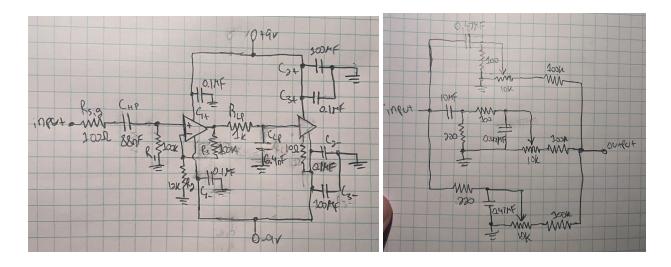
DESIGN

Our functional design is a two-stage amplifier consisting of a voltage gain stage and a unity-gain buffer stage. The voltage gain stage uses an OPA1656 op-amp and a negative feedback loop to achieve a closed loop voltage gain of 11.8. The second stage utilizes a BUF634A to decrease output impedance and increase current to the load. We implemented a high pass filter between the input and the op-amp, ensuring the midband starts at 18Hz, and a low pass filter between the op-amp and the buffer, capping the midband at 25Khz. Out design has capacitors in the positive and negative power supply lines for both chips to filter out high-frequency noise from the power supply and help stabilize the circuit from fluctuations that occur in real-life operation.

We chose the OPA1656 for our voltage gain stage due to its dual channels, low noise, and low distortion. For our unity-gain buffer we use the BUF634A, which is high-speed, high-current, and has a resistor-adjustable bandwidth. It has a low output impedance of $<5\Omega$ and a wide-bandwidth mode (210MHz) which we enabled. Both chips have a power supply range from $\pm 2.25 \text{V}$ to $\pm 18 \text{V}$ which accommodates for the 9v bolt batteries we used for the positive and negative lines.

We soldered the functional design to a perfboard for measurements. However, we also designed a three-band equalizer in HSPICE that we were unable to implement in real life due to time constraints. The EQ design allows for the adjustment of the bass (<72Hz), mid (72Hz - 3.4KHz), and treble (>3.4KHz). The equalizer works by utilizing a low pass, high pass, and band-pass filter to allow attenuation of each band, thus changing the ratio of each band in the frequency response.

SCHEMATICS



NETLISTS

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.SUBCKT EQ IN OUT BASS=0.5 MID=0.5 TREBLE=0.5
* Include our op-amps
                                 * Include the potentiometer
.include 'OPA1656.hsp'
                                 .INCLUDE 'POT.hsp'
.include 'buf634a a.hsp'
                                 * High pass filter (Treble)
                                 C6 IN E1 0.47u
Rsig in N1 100
                                 R4 E1 0 100
R1 N2 0 100k
                                 X1 0 E1 E2 POT VALUE=10k SET='TREBLE'
R2 N3 Ø 12k
                                 R2 E2 OUT 100k
R3 N3 N41 100k
Rlp N41 N42 1k
                                 * Band-Pass filter (Mid)
                                 C7 IN E3 10u
* Capacitors
Chp N1 N2 88n
                                R5 E3 0 220
                                R8 E3 E4 100
C1plus vdd 0 0.1u
C2plus vdd 0 100u
                                C9 E4 0 0.47u
C3plus vdd 0 0.1u
                                X2 0 E4 E5 POT VALUE=10k SET='MID'
Clp N42 0 6.4n
                                R1 E5 OUT 100k
C1minus vss 0 0.1u
                                 * Low Pass Filter (BASS)
C2minus vss 0 100u
                                 R6 IN E6 220
C3minus vss 0 0.1u
                                 C8 E6 0 10u
                                                                                      * Define the potentiometer subcircuit
                                X3 0 E6 E7 POT VALUE=10k SET='BASS'
* Op-Amps
                                                                                      .SUBCKT POT 1 T 2 VALUE=10K SET=0.5
                                R10 E7 OUT 100k
X1 N2 N3 vdd vss N41 OPA1650
                                                                                      RT 1 T 'VALUE*(1-SET)+.001'
X2 N42 out vdd vss BUF634A
                                                                                      RB T 2 'VALUE*SET+.001'
                                 .ENDS EQ
                                                                                      .ENDS POT
.ENDS DUT
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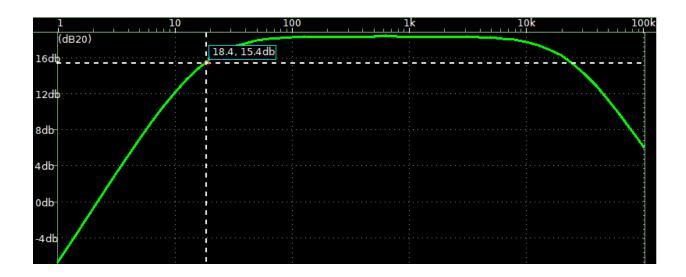
Fig 1. Amplifier Netlist

Fig 2. Equalizer (EQ) Netlist

Fig 3. Potentiometer Netlist (in EQ)

SIMULATION ANALYSIS

Test 1a yielded the following results: gain_mid= 9.2505 at= 100.0000k which indicates an appropriate gain from the first stage. Test 1b demonstrates the 3dB bandwidth of 18Hz-25KHz. Midband gain is 8.29.

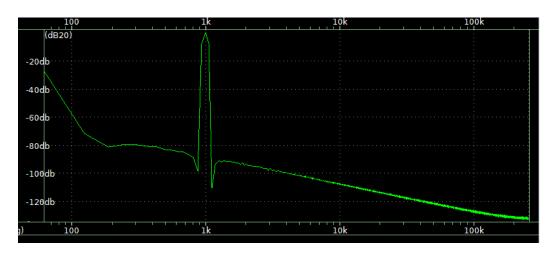


Tests 2 indicates that the input impedance, $100k\Omega$, exceeds the desired value of $10k\Omega$. Test 3 indicates that the output impedance, 3.9Ω , is below the desired threshold of 5Ω . Test 4 results show that output power (188mW) is above the required 100mW. However, the I_{SS} peak is slightly above the desired value.

			vouc_peak-	3.3343	at-	1.23331		
				from=	1.0000m	to=	2.0000m	
			vout_pp=	6.9542	from=	1.0000m	to=	2.0000m
			pout_rms=	188.9099m				
zin_1k= 100.1185k	zout 1k=	3.9288	idd_avg= -	-41.6653m	from=	1.0000m	to=	2.0000m
			iss avg=	38.0003m	from=	1.0000m	to=	2.0000m
zin_5k= 100.1034k	_		idd peak=-					
zin_10k= 100.1075k	zout_10k=	3.9386		from=	1.0000m	to=	2.0000m	
zin 15k= 100.1075k	zout_15k=	3.9386	iss_peak=	110.2651m	at=	1.7536m		
zin 20k= 100.1114k				from=	1.0000m	to=	2.0000m	

Test 5 shows that our circuit operates with low total harmonic distortion of -89dB. This is lower

than the maximum threshold of -60dB. $thd_db = -89.5445$ up to the harmonic: 40 $thd_db = -89.5445$ up to the harmonic: 40



MEASURED RESULTS

When we tested our soldered design using 9v batteries, a signal generator, and an oscilloscope, we recorded the following results. 3dB bandwidth: 18Hz - 25KHz with a midband gain of 8.1. The input impedance was $101\text{k}\Omega$ and the output impedance was 3.95Ω . The output power reached 171mW RMS. Our THD was measured to be -82dB. The discrepancy in output power is likely because our simulated I_{SS} exceeds the value achievable by 9v batteries, leading to attenuated output power in real life. The THD is also slightly greater than simulated, likely due to human error in soldering and imperfections in the audio cable and jacks.