PROJECT REPORT

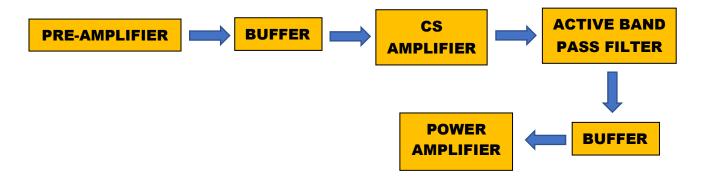
DESIGN AND CONSTRUCTION OF AN AMPLIFIER

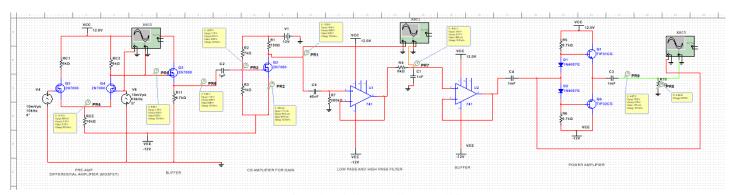
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BLOCK DIAGRAM:





The full circuit for Audio Amplifier

SPECIFICATIONS:

• **GAIN**: 500

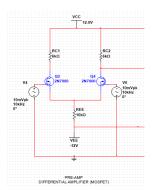
• **SUPPORTED FREQUENCY RANGE**: 20Hz-20kHz

• **FINAL POWER:** 2.5 Watts

DESCRIPTION OF SUB-BLOCKS:

PRE-AMPLIFIER:

CIRCUIT TOPOLOGY USED: Differential Amplifier using MOSFETs



A microphone is a transducer that converts vibrations from air molecules to electrical signals, but since the vibrations are small, the electrical signals are weak. A pre-amplifier is used to boost the low-voltage weak input coming from the microphone called a mic-level signal and to convert it into line-level so that it can be further processed for filtering and amplifying because all of the processing can be done only for strong signals because weak signals might get dominated by the stronger noise. Thus, to amplify the mic input at the initial level, we use a pre-amp and a differential amplifier design which serves the purpose of amplifying by cancelling the external noise. It takes a differential input from 2 sides as +vin and -vin. Due to the noise added, the new inputs are:

$$v_1 = v_{in} + v_{noise}$$

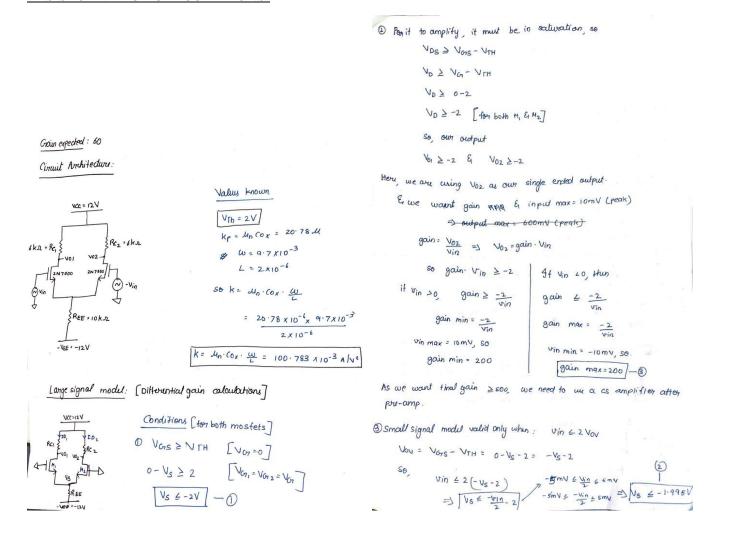
$$v_2 = -v_{in} + v_{noise}$$

Let Outputs of Q3 and Q4 be:

$$V_{OUT1} = V_{amplified} + V_{noise_amplified} \\ V_{OUT2} = -V_{amplified} + V_{noise_amplified} \\ thus \\ V_{OUT1} - V_{OUT2} = 2. \\ V_{amplified} \\ thus \\ V_{OUT1} - V_{OUT2} = 2. \\ V_{amplified} \\ thus \\ V_{OUT1} - V_{OUT2} = 2. \\ V_{amplified} \\ thus \\ V_{OUT1} - V_{OUT2} = 2. \\ V_{amplified} \\ thus \\ V_{OUT2} = 2. \\ V_{amplified} \\ thus \\ V_{OUT3} - V_{OUT3} = 2. \\ V_{amplified} \\ thus \\ V_{amplified} \\$$

Thus, we remove the noise which is called as common mode input gain and also amplify the weak signal (differential mode input)

RESISTOR VALUE CALCULATIONS:



From
$$\emptyset \ \ \Theta \ \ \ \bigvee_{S} = -2 \cdot 1$$

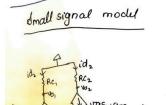
$$(4)$$

gm.
$$(vg_1-v_S+vg_2-v_S)=\frac{v_S}{ReE}$$

$$gm(-2v_S) - \frac{v_S}{REE} = 0 \implies v_S = 0$$

now, required gain = 60

similarly doing tan



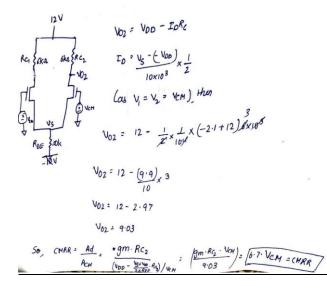
REE = 10ks

$$\frac{vo_2-0}{Rc_2}=-id_2$$

$$vo_2=-vgs_2\cdot gm\cdot Rc_2$$

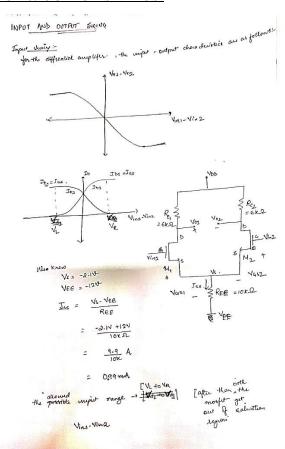
Ugs = ug - us = - uin - us

CMRR (COMMON MODE REJECTION RATIO:



As we see from the equation, we want the common mode rejection ratio to be high so that it rejects most of the common mode gain. Now, from The given equations and values, we can see that for a high CMRR, we must have an R_{EE} value such that the denominator is positive and so, a high R_{EE} would work.

INPUT & OUTPUT SWINGS:



Vint - Vingo VR is when Idi = Iss Pda 20 Vs = Mm - Vassi Va= Vina - Vasz Vins - Nosa = Vina - Vasa Vina - VIna = Vasi - Vasa Vinz = -Vinz : 2NIn = YGSI-YGSD V452 = Vth When Ido 20 Ida = 1 Uncox (W) (Vost-V+h)2 2261 2755 Ancox(WL)

2×0-99 W 100.743 XIO-8 Vin = 1 x 0.14 VIn = 0.07 9.

Vin = 70mV

Max unput signed should have a peak of Form beyond that the moster you out of voluntion region [it could be not region]

Voupput surly = Vos -Voz You = - Vos Vod = Vos-Voz = 240 Vos = VDD - IDS RDD No = Vab - Iba RDD Voz-Vog = - Ab (Jos - Jo2) Varian(E) + You - Varian(E) - You VIII(01/W) [VIII - VIDE] [Do + Ibo -2 \ Ib Fb2 山の大学 (Vins - VIn 2) = 2 Incention [Iss & 2 \ Ib 1 Inc Q (ID) ID) = - 1 Molox 10 (Viot - Yin2)2 + Iss 4 Do. 20 = [Iss - Juncon 10 (Ving Vin 2)]2 (Intrio)2. (Por 200)2 = [Tss - 1 union to (Ving Ving)] 7 - (20.102)2 = /55 = 1 wo com 10 x Tos (Vin 2 - Vin 2)2 + (Vina - Vinz) + (lun lox 10)

Luncon w (Vin-Ving) - (Vin-Vinz) + Mucos w 2

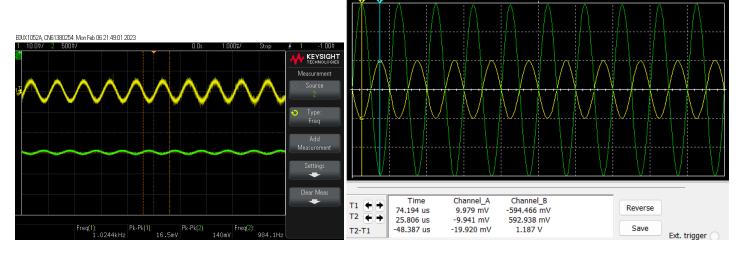
(IDIO-ID2) 2: 1 [uncor w) (Vinz-Vint) (15cs (Vinr VIn2)) IDI-JO2 = Lunconto (Vin - Vinz) / 47rs - (Vinz - Vinz) - (Vinz - Vinz) ← Ving - Vinz = QVin = QX0.08 = 0.100-2 1 x 100 x 87.0 x 87.0 x 0.14 x 0.14. D.987 X10-3 A

= 0.987 mA € Ib1-Ib2

> 2 Vo = (-RD (Ib1-962) No = 1- 6x105 x 0.987 x 10. Vo = 40960

May output suring me can get = 4.9612

RESULTS:



HARDWARE RESULTS

MULTISIM RESULTS

BUFFER:

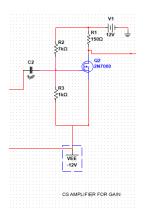
CIRCUIT TOPOLOGY USED: (Source Follower)



A buffer is basically used for impedance matching. The output impedance of a pre-amp must be matched with the input impedance of the CS-amplifier. So, when we use a buffer between pre-amp and CS-amp, the output impedance of pre-amp becomes infinite and the input impedance of CS-amp is also infinite. Thus, we use a source follower as a buffer which is built using a MOSFET here

COMON SOURCE (CS) -AMPLIFIER:

CIRCUIT TOPOLOGY USED:



We use a CS-amplifier to provide the rest of the gain required. We provided a gain of 60 in pre-amp and so here, we are aiming for a gain of 17. We provided only 60 gain in pre-amp because if we give high gain here, then input for CS-amplifier will become very high and we will not be able to do small signal analysis for that.

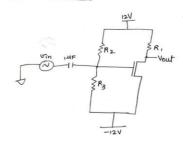
The condition to do small signal analysis is:

$$v_{in} \le 2 * v_{overdrive}$$

RESISTOR VALUE CALCULATIONS:

Grain Expected: 17

Circuit Anchitechturu:



R2 E1 R3 are used to give the required biasing

Laura Cianal model -

-12V

Large Signal model

Conditions:

VGS 2 VTH VG-VS ≥ 2 VG = 2+ VS VG > -10 / -0

2 Condition for Saturation:

now, as gain = 17, & max input we ranget: 10 m x 60 = 600 mV, SO

Vo = gain · Vin

so, Vo & 10.2 V, so

From 0 & @]

$$\frac{12-V_{G_1}}{R_2} = \frac{V_{G_1}+12}{R_3}$$

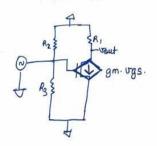
$$7R_3 = R_2$$

$$= 2 R_2 = 7kx$$

$$R_3 = 1kx$$

$$V_{03} = -9V$$

Small Signal



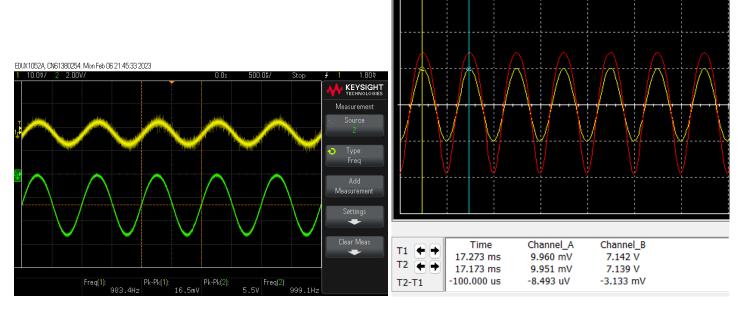
Vout = - gm · vgs · R;

= 100.783×10-3x (-9+12-2) gm = 100.783mS Forom 3 & G 17- -gm. R1. -17 = -100.783.R, => |R1 = 168.6

R, & 17052

gm = k · (Vors - VTH)

SU, we want 17 gain,



HARDWARE RESULTS

MULTISIM RESULTS

FILTER:

For the next block we built a filter to filter out high frequency noise and tune our signal according to our requirements. We built a band pass filter that allows frequency in the range of 20Hz and 20kHz to pass without attenuation and it attenuates the other frequency. To build this band pass filter we need to build a low pass filter with a cut-off frequency 20kHz and a high pass filter with a cut-off frequency of 20Hz. To decide the values for low pass and high pass filters we do the following calculation.

$$f_c = \frac{1}{2\pi RC}$$

$$RC = \frac{1}{2\pi f_c}$$

LOW PASS FILTER:

$$f_c = 20kHz$$

$$RC = \frac{1}{2\pi \times 20k}$$

$$RC = 7.96 \times 10^{-6}$$

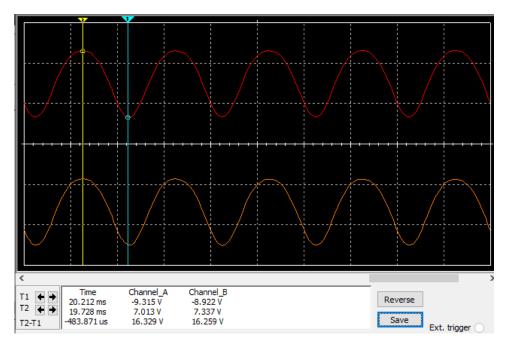
$$R = 8k\Omega \quad C = 1nF$$

HIGH PASS FILTER:

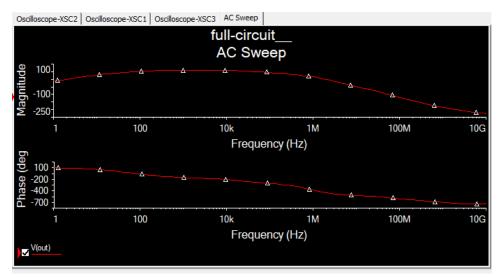
$$f_c = 20Hz$$
 $RC = \frac{1}{2\pi \times 20}$ $RC = 7.96 \times 10^{-3}$ $R = 200k\Omega$ $C = 40nF$

We use the above calculated values of resistor and capacitor to build the low pass filter. We also use a opamp buffer to connect the high pass filter with low pass filter as opamp provides an infinite input impedance. We again use this opamp buffer to connect the output from the low pass filter to the next stage, i.e., Power Amplifier, for impedance matching.

MULTISIM RESULTS:



Red waveform denotes the signal given as an input the filter block and Blue waveform denotes the output signal from the filter block

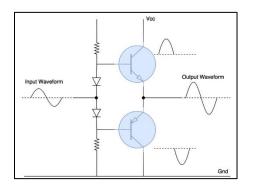


Frequency Response of the filter block

POWER AMPLIFIER:

Power amplifier is an electronic amplifier designed to increase the magnitude of power of a given input signal. The power of the input signal is increased to a level high enough to drive loads of output device, here speaker. It is the final block of in this amplifier chain to drive the speaker directly. The input signal to a power amplifier needs to be above a certain threshold. So, instead of directly feeding the input from the mic to the power amplifier, it is first pre-amplified using differential amplifier and CS amplifier and is sent as input to the power amplifier.

For this power amplifier, we use the AB type topology as shown below.



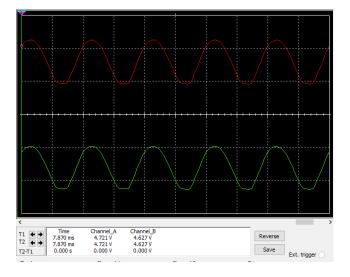
This class of amplifier is a combination of Class A and Class B amplifiers. This class of power amplifier is efficient (unlike class A) and reduces the problem of distortion of the signal at the crossover region (as seen in class B amplifier).

For this topology, we use two transistors, one npn type and other pnp type. Two forward biased diodes are connected in series to control the variation of V_{BE} (emitter-base voltage).

The resistors are connected in series with each diode. This type of circuit operates even for small power inputs. Because the class A amplifier operated for small current outputs and Class B operates for high current outputs. This is achieved by prebiasing the two transistors in the output stage of the amplifier circuit (by use of diodes). For high input currents one of the two transistors is switched on and the other is off (for positive input cycle npn transistor is on and for negative input cycle pnp is switched on). Ans for the low input signal both the transistors are on due to pre-biasing of the transistor.

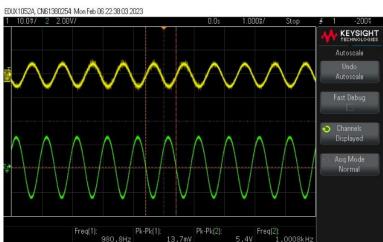
We feed the output from the filter block to this power amplifier after passing it through a coupling capacitor to remove any dc biasing if any. The output of this power amplifier is applied to the speaker which has a input impedance of approximately 10ohms. The average power obtained at the output load is approximately 2.5W which is enough to drive the speaker.

MULTISIM RESULTS:



Red waveform is the output from the filter (input given to the power amplifier) and the green waveform is the final output from the power amplifier across the speaker load

HARDWARE RESULTS:



Yellow waveform is the input signal given (10mVpp) to this audio amplifier and the green waveform is the output signal received from the power amplifier across the speaker.