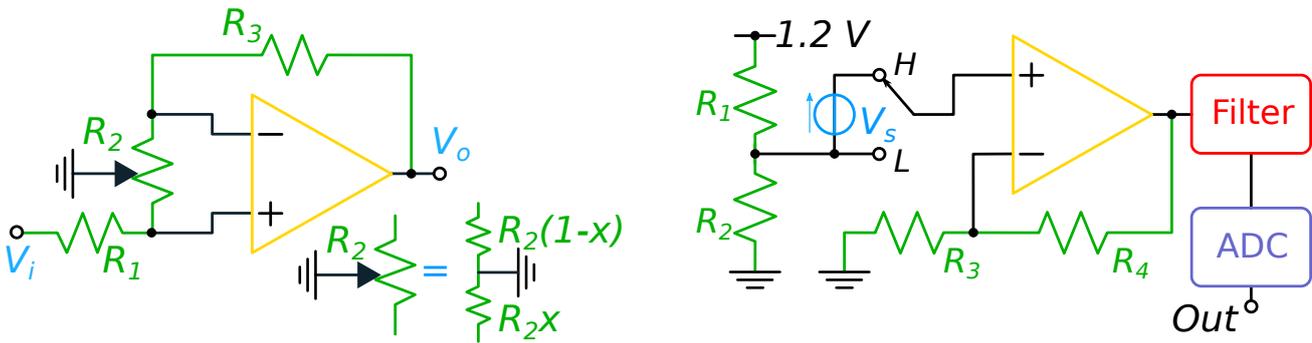


For a correct evaluation, please write your answers in a readable way; thank you



Solving six points correctly gives you 30/30

**Problem 1**

The scheme in the left figure is an audio gain stage that allows for a wide gain adjustment range and low noise, with a single potentiometer  $R_2$ . The OA has  $A_0 = 120$  dB and poles at 60 Hz and 15 MHz. Parameters are  $R_1 = 1$  k $\Omega$ ,  $R_3 = 8$  k $\Omega$ . In solving the scheme, replace the  $R_2$  potentiometer with two resistors, as shown.  $x$  changes from 0 to 1 by rotating the wiper.

1. Compute the gain.
2. Compute the loop gain and find the maximum vale of  $R_2$  that grants a phase margin larger than 45° for every  $x$ .
3. Compute the output rms noise voltage considering the equivalent noise source of the OA  $\sqrt{S_V} = 3$  nV/ $\sqrt{\text{Hz}}$  plus resistor noise ( $4k_B T \approx 1.646 \times 10^{-20}$  J). Consider  $R_2 = 2$  k $\Omega$  and just evaluate the limit cases  $x = 0$  and  $x = 1$  for simplicity.
4. The proposed scheme only gives positive gains. Propose a simple modification that can give both positive and negative values when  $x$  goes from zero to one (hint: remember the inverting stage configuration and add just one resistor!).

**Problem 2**

A temperature sensor outputs a voltage  $V_s = \alpha T$ , with  $\alpha = 40.7$   $\mu\text{V}/^\circ\text{C}$  and  $T$  expressed in  $^\circ\text{C}$ . The circuit on the right is used to measure  $T$  in the range  $[-100, +200]^\circ\text{C}$ . The ADC input range is  $[0, 1.2]$  V. The OA has  $V_{OS} = 10$  mV and input voltage noise  $\sqrt{S_V} = 200$  nV/ $\sqrt{\text{Hz}}$ . Filter  $BW$  is 30 Hz.

1. Neglect  $V_{OS}$  and find the parameter values that allows to measure  $T$  over the whole range.
2. Consider now  $V_{OS}$  and find the new parameter values. Remember that  $V_{OS}$  represents the absolute value of the offset.
3. To eliminate  $V_{OS}$ , its effect is first evaluated (switch L) and subtracted from the actual  $T$  measurement (switch H). Compute  $S/N$  for a minimum  $T$  difference of 1 $^\circ\text{C}$ .
4. The OA has a flicker noise component with corner frequency  $f_{nc} = 4$  kHz. Find the minimum reference frequency of a synchronous detection system that grants  $S/N > 10$ .

Allowed time: 2 hours 45 minutes – Do a good job!

# Solution

## Problem 1

### 1.1

The voltage at the NI input of the OA is

$$V^+ = V_i \frac{R_2 x}{R_1 + R_2 x},$$

and the output becomes (NI stage gain)

$$V_o = V^+ \left( 1 + \frac{R_3}{R_2(1-x)} \right) = V_i \frac{R_2 x}{R_1 + R_2 x} \frac{R_2(1-x) + R_3}{R_2(1-x)} = V_i \frac{x}{1+2x} \frac{10-2x}{1-x},$$

where the last expression holds for  $R_2 = 2 \text{ k}\Omega$  and is shown in Fig. 1 (left). Extreme gain values are zero and (ideally) infinite (or actually,  $A_0$ ), as can be easily seen from the scheme.

Comment on the result: The advantage of this solution is that the gain is non-linear with  $x$ , with a shape that is (loosely) linear on a  $dB$  scale (if you don't get too close to the extremes), that is what you want in an audio equipment, and all this is achieved with a single potentiometer. The standard 20 kHz audio bandwidth can be simply set by a 1 nF capacitor in parallel to  $R_3$ .

### 1.2

The calculation of the loop gain is straightforward and leads to

$$G_{loop} = -A(s) \frac{R_2(1-x)}{R_3 + R_2(1-x)},$$

from which it is clear that problems might arise at  $x = 0$ , where we have

$$G_{loop} = -A(s) \frac{R_2}{R_3 + R_2}.$$

To achieve a phase margin of  $45^\circ$ , the OA pole at 15 MHz must fall below the 0 dB. The gain of  $A(s)$  at such frequency is

$$A_0 \times 60 = G \times 15 \times 10^6 \Rightarrow G = 4,$$

so that

$$G \frac{R_2}{R_3 + R_2} < 1 \Rightarrow R_2 < \frac{R_3}{3} = 2.67 \text{ k}\Omega.$$

Comment on the result: Please note that for  $x \rightarrow 1$  we have  $|G_{loop}| \rightarrow 0$  and the OA works in open loop, which is not great (see noise calculations later).

### 1.3

We follow the suggestion and consider first  $x = 0$ , i.e., grounded NI input. The scheme is a simple NI amplifier and output noise PSD becomes:

$$S_{V_o} = S_V \left( 1 + \frac{R_3}{R_2} \right)^2 + \left( \frac{4k_B T}{R_3} + \frac{4k_B T}{R_2} \right) R_3^2 = 2.25 \times 10^{-16} + 6.58 \times 10^{-16} = 8.83 \times 10^{-16} \text{ V}^2/\text{Hz}.$$

With  $R_2 = 2 \text{ k}\Omega$  we have  $G_{loop} = -A(s)/5$ , and  $f_{0dB} = 60/5 = 12 \text{ MHz}$ , so that

$$\overline{V_o^2} = S_{V_o} \frac{\pi}{2} f_{0dB} \approx (129 \text{ }\mu\text{V})^2.$$

For  $x = 1$  the OA is in open loop, and the gain is simply  $A(s)$ . The NI input noise PSD is

$$S_V^+ = S_V + \left( \frac{4k_B T}{R_1} + \frac{4k_B T}{R_2} \right) (R_1 \parallel R_2)^2 = 9 \times 10^{-18} + 1.1 \times 10^{-17} = 2 \times 10^{-17} \text{ V}^2/\text{Hz}$$

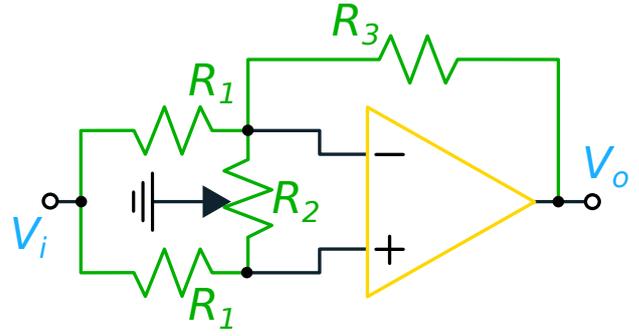
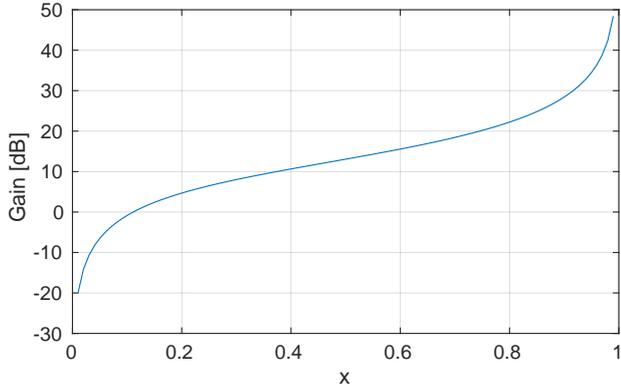


Figure 1: Left = Gain as a function of the pot position  $x$ . Right = modified scheme that provides negative gains.

and, considering  $f_p = 60$  Hz:

$$S_{V_o} = S_V^+ |A(s)|^2 \Rightarrow \overline{V_o^2} = S_V^+ A_0^2 \frac{\pi}{2} f_p \approx (43.4 \text{ mV})^2.$$

Comment on the result: This value is huge, but it should be kept in mind that comes at an unreasonable value of the gain. In reality, a small resistor placed in series to the potentiometer will limit the gain and the noise.

### 1.4

For  $x = 1$  the ideal gain is infinite. If we want to add a negative gain, we need to work around  $x = 0$ , where the OA NI input is grounded. A negative gain can be obtained by adding a resistor  $R_1$  from the input to the I input of the OA, as shown in Fig. 1, right. For  $x = 0$  the gain becomes  $-R_3/R_1$ .

## Problem 2

### 2.1

We label  $V_{ref} = 1.2 \text{ V } R_2/(R_1 + R_2)$  the voltage after the  $R_1 - R_2$  divider, and compute the minimum and maximum OA output voltages:

$$\begin{aligned} V_{min} &= (V_{ref} + \alpha T_{min})G \\ V_{max} &= (V_{ref} + \alpha T_{max})G, \end{aligned}$$

where  $G = 1 + R_4/R_3$  is the NI stage gain. We now have

$$V_{min} > 0 \Rightarrow V_{ref} + \alpha T_{min} > 0 \Rightarrow V_{ref} > -\alpha T_{min} = 4.1 \text{ mV}$$

and, setting  $V_{min} = 0$ ,

$$\begin{aligned} V_{max} < 1.2 \text{ V} &\Rightarrow (V_{ref} + \alpha T_{max})G = (V_{min} + \alpha(T_{max} - T_{min}))G < 1.2 \text{ V} \Rightarrow \\ &\alpha(T_{max} - T_{min})G < 1.2 \text{ V} \Rightarrow G < 98.3. \end{aligned}$$

Possible (ideal) resistor values are  $R_1 = 292 \text{ k}\Omega$ ,  $R_2 = 1 \text{ k}\Omega$ ,  $R_3 = 1 \text{ k}\Omega$ ,  $R_4 = 97.3 \text{ k}\Omega$ . Actual values will be a bit different because of commercial availability.

### 2.2

The worst case happens when  $V_{OS}$  lowers  $V_{min}$  and increases  $V_{max}$ . The new requirements become then

$$\begin{aligned} (V_{ref} + \alpha T_{min} - V_{OS})G &> 0 \Rightarrow V_{ref} > 14.1 \text{ mV} \\ (V_{ref} + \alpha T_{max} + V_{OS})G &< 1.2 \text{ V} \Rightarrow G < 37.3. \end{aligned}$$

Note the large change in values, consequence of the large  $V_{OS}$ : its value is equivalent to a temperature difference of  $V_{OS}/\alpha \approx 246^\circ\text{C}$ , i.e., almost equal to the entire signal range! New ideal values are then  $R_1 = 84 \text{ k}\Omega$ ,  $R_2 = 1 \text{ k}\Omega$ ,  $R_3 = 1 \text{ k}\Omega$ ,  $R_4 = 36.3 \text{ k}\Omega$ .

### 2.3

The difference operation removes the LF offset, meaning that we can switch back to the resistor values computed in #2.1. We have then for the OA output noise

$$S_{V_o} = S_V G^2 + 4k_B T (R_1 \parallel R_2) G^2 + 4k_B T R_3 (G - 1)^2 + 4k_B T R_4 \approx G^2 S_V = 3.9 \times 10^{-10} \text{ V}^2/\text{Hz},$$

and  $S/N$  becomes ( $\Delta T = 1^\circ\text{C}$ ,  $BW_n = \frac{\pi}{2} \times 30 \text{ Hz}$ ):

$$\left(\frac{S}{N}\right) = \frac{\alpha \Delta T G}{\sqrt{2S_{V_o} BW_n}} \approx \frac{\alpha \Delta T}{\sqrt{2S_V BW_n}} \approx 21,$$

where the factor 2 comes from the difference operation.

### 2.4

If the switch is driven by a square wave at frequency  $f_r$ , the OA output is also a square wave whose peak-to-peak amplitude  $A = \alpha G \Delta T$  represents the signal. If we pick a sinusoidal demodulation, this will extract the fundamental harmonic of the square wave, with amplitude  $2A/\pi$ . We then get

$$\left(\frac{S}{N}\right) = \frac{2A/\pi}{\sqrt{4S_x(f_r) G^2 BW_n}}.$$

If we demodulate below  $f_{nc}$  we have  $S_x = K/f_r$  and

$$\frac{\alpha \Delta T}{\pi} \sqrt{\frac{f_r}{K BW_n}} > 10 \Rightarrow f_r > \frac{100\pi^2 K BW_n}{(\alpha \Delta T)^2} \approx 2.86 \text{ kHz}.$$

Note that this value is close to  $f_{nc}$ , so we should also account for the WN component. A more exact calculation is obtained with  $S_x \approx G^2(K/f_r + S_V)$  and

$$\left(\frac{S}{N}\right) = \frac{A/\pi}{\sqrt{S_x(f_r) BW_n}} > 10 \Rightarrow f_r > \frac{100\pi^2 BW_n K}{(\alpha \Delta T)^2 - 100\pi^2 BW_n S_V} = 10 \text{ kHz}.$$

A final, probably unsolicited, comment is that the system in #2.3 already rejects some LF noise! Let's do some math: The sampling times difference is limited by the  $f_p = 30 \text{ Hz}$  filter, as its output must reach the steady-state before sampling. Considering 5 time constant, we have

$$t_s \geq \frac{5}{2\pi 30} = 26.5 \text{ ms},$$

meaning that the output FN contribution becomes (see lesson slide on correlated double sampling for derivation of the WF):

$$\overline{n_{FN}^2} = K G^2 \int_0^{x_p} \frac{1 - \cos x}{x} dx \approx 2.38 K G^2 = 3.8 G^2 \times 10^{-10} \text{ V}^2/\text{Hz},$$

where  $x = 2\pi f t_s$  and  $x_p = 2\pi f_p t_s = 5$ . We have then

$$\left(\frac{S}{N}\right) = \frac{\alpha \Delta T G}{\sqrt{2S_{V_o} BW + n_{FN}^2}} \approx \frac{\alpha \Delta T}{\sqrt{2S_V BW + 2.38K}} \approx 2.$$

close to the requirement, but no cigar!