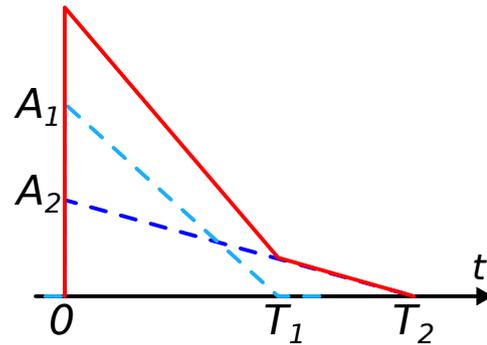
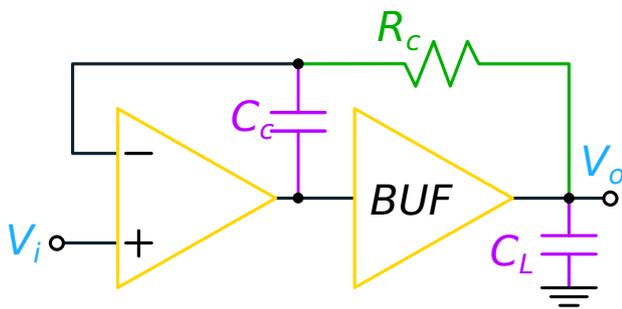


For a correct evaluation, please write your answers in a readable way; thank you



Solving six points correctly gives you 30/30

**Problem 1**

The scheme in the left figure is a power buffer for capacitive loads. The OA has  $A_0 = 126$  dB and  $GBWP = 20$  MHz. The buffer stage (BUF) has unity gain and output resistance  $R_o = 6 \Omega$ . Other parameters are  $R_c = 2$  k $\Omega$ ,  $C_c = 1$  nF.

1. Consider  $R_o = 0$  and compute the ideal and loop gains.
2. Evaluate the maximum value of  $C_L$  that grants a phase margin larger than  $45^\circ$  when  $R_o$  is not zero.
3. Compute the output rms noise voltage considering the equivalent noise sources of the OA and BUF  $\sqrt{S_V} = 20$  nV/ $\sqrt{\text{Hz}}$ ,  $\sqrt{S_I} = 10$  pA/ $\sqrt{\text{Hz}}$ . Consider again  $R_o = 0$  for simplicity.
4. Would it be convenient to use a simpler scheme without  $R_c$  and  $C_c$ , having the OA connected as follower and BUF driving the load? Why?

**Problem 2**

Because of different electron and hole mobilities, a semiconductor sensor outputs a signal made up of two triangular pulses with different amplitudes ( $A_1, A_2$ ) and durations ( $T_1, T_2$ , with  $T_2 > T_1$ ).  $T_1$  and  $T_2$  are due to the sensor design and are hence known. We want to measure the amplitude  $A_1$  in presence of white noise with bilateral PSD  $\lambda$ . Note that  $A_2$  is not known.

1. Consider two GIs plus basic electronics (sum, difference, etc.). Propose a measurement scheme with its parameter values and sketch the resulting weighting function.
2. Evaluate  $S/N$ .
3. Consider now a shot-like white noise with PSD  $\lambda$  proportional to the signal, *i.e.*, with two triangular components with PSD amplitudes  $\lambda_1$  and  $\lambda_2$ . Compute the new output noise.
4. Consider a repetitive signal. Would it be convenient to employ boxcar averagers in the non-stationary white noise case of #2.3? If yes, would the improvement in  $S/N$  follow the  $\sqrt{N_{eq}}$  law even in this case, or not? Justify your answers.

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

# Solution

## Problem 1

### 1.1

As the BUF gain is one and  $R_o = 0$ ,  $V_o$  is the voltage at the OA output. Hence, there can be no current flowing into  $R_c$  and  $C_c$ , meaning that  $V_i = V_o$ . The ideal gain is therefore one.

To compute  $G_{loop}$ , we cut the loop at the OA output and apply a test voltage  $V_s$ . Once again, no current can flow into the feedback network, and we obtain  $V^- = V_s$ , *i.e.*

$$G_{loop} = -A(s).$$

### 1.2

The scheme for  $G_{loop}$  calculation is shown in Fig. 1 (left). We label  $V_1$  the voltage drop on  $C_L$  and write (remember that the BUF output is also at  $V_s$ ):

$$\begin{aligned} \frac{V_s - V_1}{R_o} &= sC_L V_1 + \frac{V_1 - V^-}{R_c} & V_1 \left( \frac{1}{R_c} + \frac{1}{R_o} + sC_L \right) &= \frac{V_s}{R_o} + \frac{V^-}{R_c} \\ &\Rightarrow & \frac{V_1 - V^-}{R_c} &= sC_c(V^- - V_s) & \frac{V_1}{R_c} &= V^- \left( \frac{1}{R_c} + sC_c \right) - sC_c V_s \end{aligned}$$

from which, after some algebra

$$\frac{V^-}{V_s} = \frac{1 + sC_c(R_c + R_o) + s^2 C_c C_L R_c R_o}{1 + sC_c(R_c + R_o) + sC_L R_o + s^2 C_c C_L R_c R_o} \approx \frac{1 + sC_c R_c + s^2 C_c C_L R_c R_o}{1 + s(C_c R_c + C_L R_o) + s^2 C_c C_L R_c R_o},$$

that means

$$G_{loop} = -A(s) \frac{1 + sC_c R_c + s^2 C_c C_L R_c R_o}{1 + s(C_c R_c + C_L R_o) + s^2 C_c C_L R_c R_o} = -A(s) \frac{1 + s\tau_c + s^2 \tau_c \tau_L}{1 + s(\tau_c + \tau_L) + s^2 \tau_c \tau_L},$$

where we set for simplicity  $\tau_c = C_c R_c$  e  $\tau_L = C_L R_o$ . If  $\tau_L \ll \tau_c$ , *i.e.*,  $C_L \ll C_c R_c / R_o = 0.33 \mu\text{F}$ ,  $G_{loop} \approx -A(s)$  and stability is ensured. We then consider the other extreme, where  $\tau_L \gg \tau_c$  and

$$G_{loop} \approx -A(s) \frac{1 + s\tau_c + s^2 \tau_c \tau_L}{1 + s\tau_L + s^2 \tau_c \tau_L}.$$

Approximate pole positions are  $f_1 = 1/2\pi\tau_L$  and  $f_2 = 1/2\pi\tau_c \approx 80 \text{ kHz}$ , while zeros are complex at a frequency  $f_z = 1/(2\pi\sqrt{\tau_L\tau_c})$ . This means that for frequencies higher than  $f_2$ , we still have  $G_{loop} \approx -A(s)$ , and the system remains stable for any capacitor value. Loop gain is shown in Fig. 1 (right). Further discussion is in the Appendix.

### 1.3

The OA voltage noise is transferred with unity gain, like the signal, while its NI current noise gives no contribution. So does the current noise of the BUF, leaving the OA I input current noise and BUF voltage noise. With simple calculations we get

$$S_{V_o} = S_{V_{OA}} + S_{I_{OA}} \left| \frac{R}{1 + sC_c R_c} \right|^2 + S_{V_{BUF}} \left| \frac{sC_c R_c}{1 + sC_c R_c} \right|^2.$$

Considering that the zero-dB frequency of  $G_{loop}$  is equal to the  $GBWP$ , we have

$$\begin{aligned} \overline{V_o^2} &= S_V \frac{\pi}{2} GBWP + S_I R_c^2 \frac{1}{4C_c R_c} + S_V \left( \frac{\pi}{2} GBWP - \frac{1}{4R_c C_c} \right) \\ &\approx 1.26 \times 10^{-8} + 5 \times 10^{-11} + 1.26 \times 10^{-8} \approx 2.52 \times 10^{-8} \text{ V}^2 = (158 \mu\text{V})^2. \end{aligned}$$

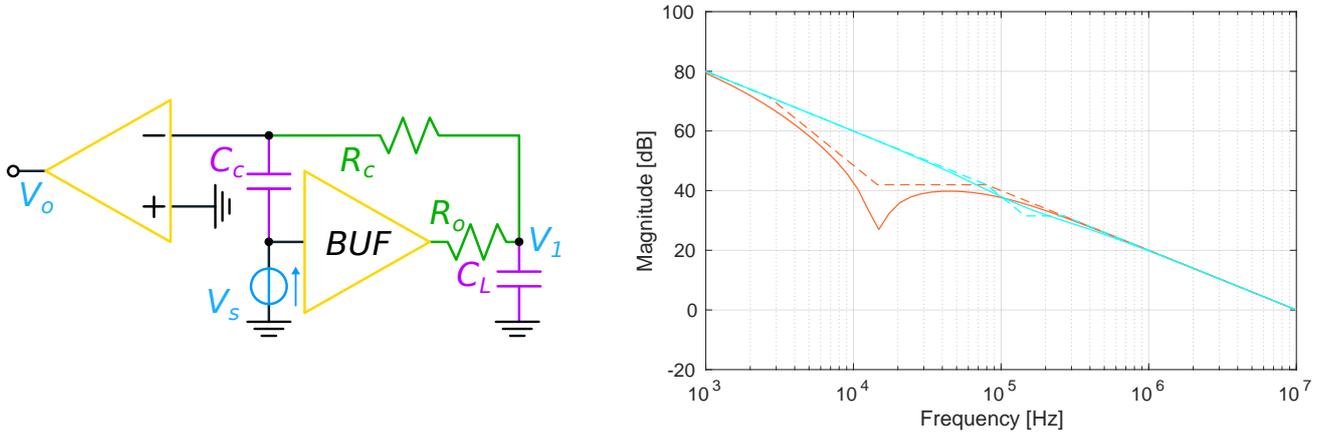


Figure 1: Left = Loop gain calculation scheme. Right = Resulting loop gain in the  $10^3 - 10^7$  Hz range for  $C_L = 100$  nF (blue) and  $10 \mu\text{F}$  (red). Solid lines = actual gain, dashes = asymptotic approximation.

## 1.4

The proposed solution has the disadvantage of having a pole set by the  $R_o C_L$  time constant, hence depending on the load conditions. Moreover, keeping BUF outside the loop means that its offset, non-linearity and other unwanted drawbacks affect the output.

## Problem 2

### 2.1

In principle, any two measurements can be used to extract  $A_1$ . However, a simple choice is to evaluate  $A_2$  from an integration in the  $T_1 - T_2$  interval (or  $2/3$  of it) and subtract this value from an integration in the  $0 - T_1$  interval. The GI outputs are

$$V_1 = A_1 G_1 \frac{T_1}{2} + A_2 G_1 T_1 \left(1 - \frac{T_1}{2T_2}\right) = A_1 \frac{T_1}{2} G_1 + A_2 \frac{T_1(2T_2 - T_1)}{2T_2} G_1$$

$$V_2 = A_2 G_2 \left(1 - \frac{T_1}{T_2}\right) \frac{T_2 - T_1}{2} = A_2 \frac{(T_2 - T_1)^2}{2T_2} G_2,$$

where  $G_1$  and  $G_2$  are the GI gains. We then set

$$G_2 = G_1 \frac{T_1(2T_2 - T_1)}{(T_2 - T_1)^2}$$

and subtract  $V_2$  from  $V_1$ . Setting  $G_1 = 2K/T_1$  returns  $V_o = KA_1$ . The WF is sketched in Fig. 2 (left).

### 2.2

The output noise is simply

$$\overline{V_0^2} = \lambda \int_0^{T_2} w^2(t, \tau) d\tau = \lambda (G_1^2 T_1 + G_2^2 (T_2 - T_1)) = \lambda G_1^2 T_1 \left(1 + \frac{T_1(2T_2 - T_1)^2}{(T_2 - T_1)^3}\right) =$$

$$\lambda G_1^2 T_1 T_2 \frac{T_2^2 + T_1 T_2 - T_1^2}{(T_2 - T_1)^3},$$

from which

$$\left(\frac{S}{N}\right) = \frac{A_1}{2\sqrt{\lambda}} \sqrt{\frac{T_1(T_2 - T_1)^3}{T_2(T_2^2 + T_1 T_2 - T_1^2)}}.$$

For example, considering  $T_1 = 1 \mu\text{s}$ ,  $T_2 = 2 \mu\text{s}$  and  $K = 1$  the noise becomes  $\overline{V_0^2} = \lambda \times 40 \times 10^6$ .

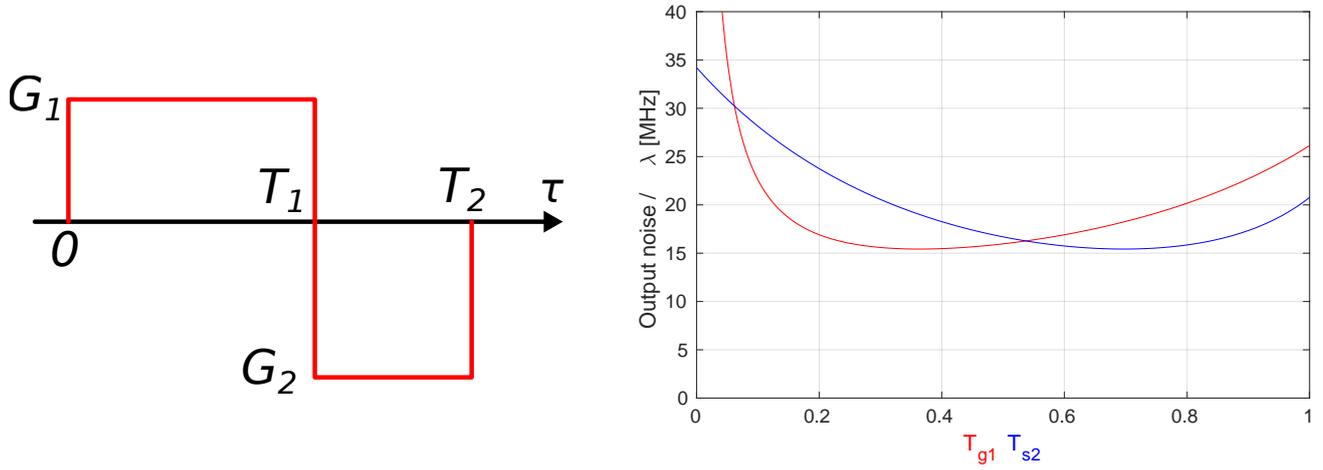


Figure 2: Left = possible weighting function for measuring  $A_1$ . Right = output noise as a function of  $T_{g1}$  (red) and  $T_{s2}$  (blue) for  $T_1 = 1 \mu\text{s}$ ,  $T_2 = 2 \mu\text{s}$  and  $K = 1$ .

### 2.3

The output mean square noise for a non-stationary white noise is (see class notes):

$$\overline{n_{out}^2(t)} = \int \lambda(\tau) w^2(t, \tau) d\tau.$$

The filter output is collected at  $t = T_2$ , where we get

$$\begin{aligned} \overline{n_{out}^2(T_2)} &= G_1^2 \int_0^{T_1} \left[ \lambda_1 \left(1 - \frac{\tau}{T_1}\right) + \lambda_2 \left(1 - \frac{\tau}{T_2}\right) \right] d\tau + G_2^2 \lambda_2 \int_{T_1}^{T_2} \left(1 - \frac{\tau}{T_2}\right) d\tau \\ &= G_1^2 \left[ \lambda_1 \frac{T_1}{2} + \lambda_2 T_1 \left(1 - \frac{T_1}{2T_2}\right) + \frac{G_2^2}{G_1^2} \lambda_2 \frac{(T_2 - T_1)^2}{2T_2} \right] = G_1^2 \frac{T_1}{2} \left[ \lambda_1 + \lambda_2 \frac{T_2(2T_2 - T_1)}{(T_2 - T_1)^2} \right]. \end{aligned}$$

### 2.4

The answer is yes to both questions. BAs improve  $S/N$  (by the square root of the number of equivalent samples) whenever the noise is not correlated over different samples, which is indeed the case here!

To conclude, we briefly discuss the optimization of the GI parameters. It makes sense to start the first integration from 0, as the signal is maximum there. Analogously, it is reasonable to stop the second integration at  $2/3$  of the  $T_1 - T_2$  interval. We are then left with the gate time of the first GI,  $T_{g1}$ , and the starting time of the second GI,  $T_{s2}$ . The output noise as a function of such parameters is shown in Fig. 2 (right) for  $T_1 = 1 \mu\text{s}$ ,  $T_2 = 2 \mu\text{s}$  and  $K = 1$ ; note that the optimum values are  $T_{g1} \approx 0.36 \mu\text{s}$  and  $T_{s2} \approx 0.7 \mu\text{s}$ .

### Appendix

The results in #1.2 do not guarantee that the circuit is actually useful for any value of  $C_L$ . In fact, if we compute the (ideal) closed-loop gain, we get (labelling  $V_1$  the OA output):

$$\begin{aligned} \frac{V_1 - V_o}{R_o} &= sC_L V_o + \frac{V_o - V_i}{R_c} \Rightarrow \frac{V_o}{V_i} = \frac{1 + sC_c(R_c + R_o)}{1 + sC_c(R_c + R_o) + s^2 C_c C_L R_o R_c} \\ \frac{V_1 - V_o}{R_c} &= sC_c(V_i - V_1), \end{aligned}$$

and the closed-loop poles might be complex. To ensure this is not the case, we must have

$$C_c^2 (R_c + R_o)^2 > 4C_c C_L R_o R_c \Rightarrow C_L < C_c \frac{(R_c + R_o)^2}{4R_o R_c} \approx C_c \frac{R_c}{4R_o} = 83 \text{ nF}.$$

If a small amount of oscillation might be tolerated (quality factor of the poles equal to one), the requirement becomes the one already discussed in #1.2, *i.e.*

$$C_L < C_c \frac{R_c}{R_o} = 0.33 \mu\text{F}.$$