

# Designing aapsdkaspokd

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## 0.1 Analog Input Stage Design

### 0.1.1 Input AC Coupling

The TLV320AIC series codec has a configurable input impedance of 10 k $\Omega$ , 20 k $\Omega$ , and 40 k $\Omega$  **slaa557**. To minimize loading on the source, the maximum available value of 40 k $\Omega$  is chosen.

The AC-coupling capacitor together with the input impedance of the codec ADC forms a simple RC high-pass filter, whose cutoff frequency can be calculated as

$$f_c = \frac{1}{2\pi R_{eq} C_c}$$

The resulting cutoff frequency  $f_c$  should be well below the audible range, typically under 20 Hz, to preserve bass integrity and prevent phase shifts that could degrade the perceived low-frequency response **audio phase shift**.

With a coupling capacitor of  $C_c = 2.2 \mu\text{F}$  and  $R_{eq} = 40 \text{ k}\Omega$ , the cutoff frequency is

$$f_c = \frac{1}{2\pi \cdot 40 \text{ k}\Omega \cdot 2.2 \mu\text{F}} \approx 1.81 \text{ Hz}$$

### 0.1.2 Input Scaling

The TLV datasheet specifies a maximum input voltage on the analog pins of  $0.5 V_{\text{RMS}}$  with a common-mode voltage of  $0.9 \text{ V}$  **slaa557**. To map Eurorack audio levels into this range, the signal must be attenuated appropriately. Typical Eurorack signals reach  $\pm 5 \text{ V}$ , i.e.,  $10 V_{\text{pp}}$  **doepfer a110-2**.

The target ADC peak-to-peak voltage is calculated from the RMS specification:

$$V_{\text{ADC,peak}} = V_{\text{ADC,RMS}} \cdot \sqrt{2} \approx 0.707 \text{ V}$$

$$V_{\text{ADC,pp}} = 2 \cdot V_{\text{ADC,peak}} \approx 1.414 \text{ V}$$

The required attenuation factor is then

$$a = \frac{V_{\text{ADC,pp}}}{V_{\text{Eurorack}}} = \frac{1.414}{5} \approx 0.283$$

To satisfy the input impedance requirements while providing this attenuation, a **unity-gain buffer op-amp** is used before a passive resistor divider. The buffer presents an input impedance equal to the op-amp's own input impedance, which is typically greater than 200 k $\Omega$ , thereby easily meeting the desired 100 k $\Omega$  input impedance at the module. The resistor divider can then be designed to provide the necessary attenuation without significantly loading the signal source.

### 0.1.3 Design Rationale

The chosen signal-conditioning approach uses a unity-gain buffer followed by a passive resistor divider. This decision is based on the constraints of the system and the electrical design conclusions of other commonly used amplification circuits. An inverting amplifier topology is unsuitable because its input impedance is determined directly by the input resistor  $R_{\text{in}}$ . Since the module requires an input impedance on the order of 100 k $\Omega$ , the corresponding feedback resistor  $R_{\text{f}}$  would also need to be of similar magnitude in order to achieve the desired gain. Such large resistor values introduce significant thermal noise, as described by the Johnson noise expression

$$v_n = \sqrt{4kTR},$$

which grows with increasing resistance and would negatively affect the signal-to-noise ratio. Furthermore, due to space and component constraints, adding a dedicated high-impedance buffer stage in front of the inverting configuration is not feasible.

A non-inverting amplifier configuration is also unsuitable because it cannot realize an attenuation. Its closed-loop gain is defined as

$$A_{\text{v,noninv}} = 1 + \frac{R_{\text{f}}}{R_{\text{g}}},$$

and therefore satisfies

$$A_{v,\text{noninv}} \geq 1,$$

which prohibits any gain smaller than unity. Since the ADC input range requires attenuation of the incoming signal, a non-inverting topology cannot be used.

For these reasons, the most practical and robust solution is to use a unity-gain buffer to ensure high input impedance and to isolate the source from the following attenuation network, and then apply a passive resistor divider to scale the signal to the required amplitude. The divider consists of  $R_{\text{top}} = 10\text{ k}\Omega$  and  $R_{\text{bot}} = 1.8\text{ k}\Omega$ , resulting in a total resistance of  $11.8\text{ k}\Omega$ . The associated divider current is given by

$$I = \frac{V_{\text{in}}}{R_{\text{top}} + R_{\text{bot}}}.$$

Thus, for an input voltage of  $5\text{ V}$  the current is  $I = 0.424\text{ mA}$ , and for  $12\text{ V}$  it increases to  $I = 1.017\text{ mA}$ . This power consumption is acceptable within the system's constraints, and the simplicity and predictability of the passive attenuation stage outweigh the modest inefficiency introduced by the resistor divider.

#### 0.1.4 Comparison of Input Stage Circuits

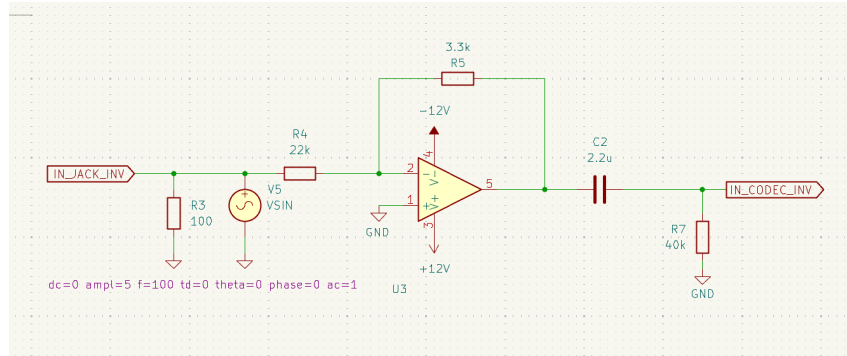
I ran SPICE simulations to compare two input stage approaches:

1. Inverted Feedback Attenuator
2. Buffer to Voltage Divider

### 0.2 Circuit Schematics

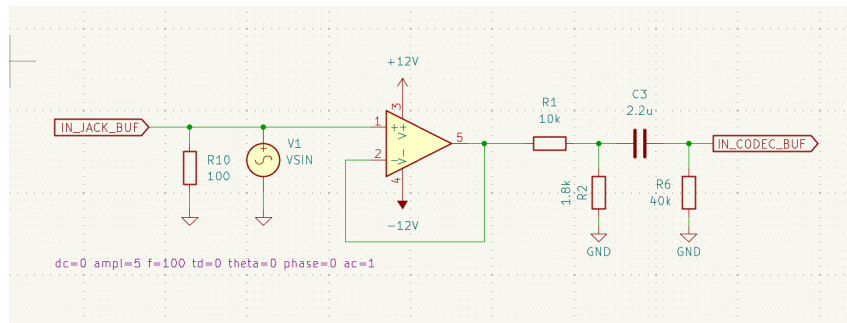
ADC input impedance is  $40\text{ k}\Omega$  and the signal source output impedance is  $100\Omega$ . The attenuation factor  $a$  is slightly different because of resistor values.

#### Inverted Feedback Attenuator

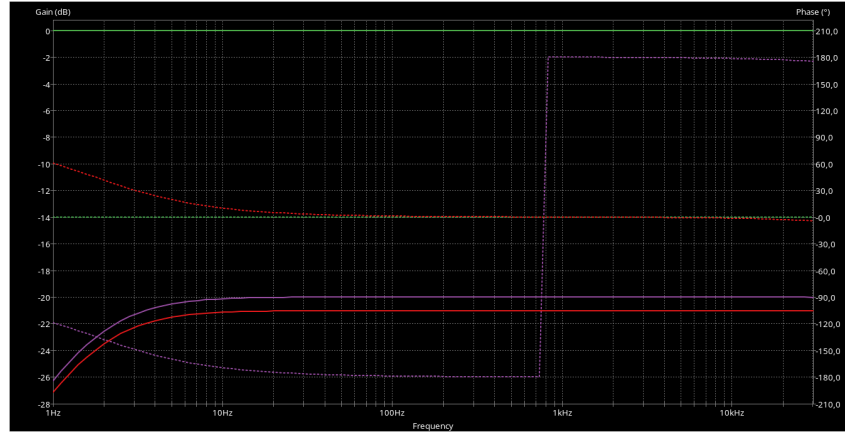


This circuit is used in most Mutable Instruments modules. It only offers  $22\text{ k}\Omega$  input impedance, and probably has higher thermal noise due to the resistors in the feedback path.

#### Buffer to Voltage Divider



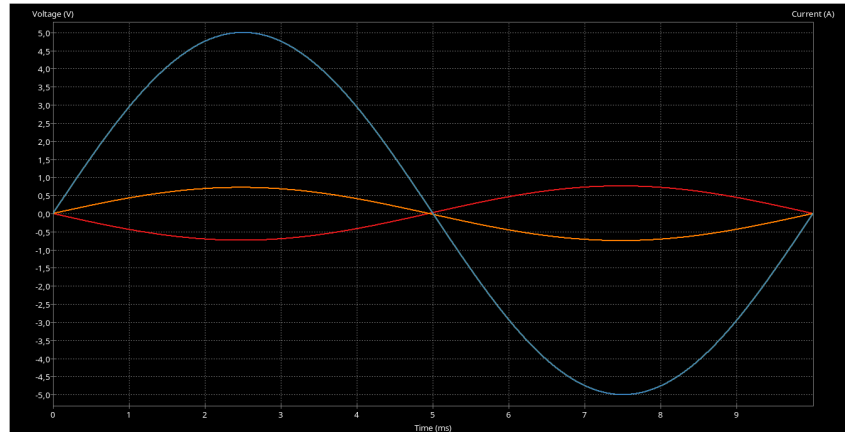
**AC Analysis – Bode Plot** The Bode plot shows the frequency response of both circuits in audio range ( $1\text{Hz} - 30\text{kHz}$ ):



- **Green:** Input signal at  $0\text{dB}$
- **Red:** Output of buffered circuit
- **Purple:** Output of inverted feedback circuit

**Observation:** Both circuits perform similarly, but the inverted feedback has a strong phase shift at  $700\text{--}800\text{Hz}$ . The phase shift in the bass frequencies caused by the AC-coupling capacitor is below  $10^\circ$  at  $20\text{Hz}$ .

### Transient Analysis – 100 Hz Signal



- **Blue:** Input signal ( $10V_{pp}$ ,  $100\text{Hz}$ )
- **Red:** Output of inverted feedback circuit ( $\sim 1.5V_{pp}$ )
- **Orange/Red:** Output of buffered circuit ( $\sim 1.47V_{pp}$ )

#### 0.2.1 ADC Absolute Input Limits:

The input pins of the ADC must remain within  $-0.3\text{ V}$  to  $(A_{VDD} + 0.3\text{ V})$ . Since  $A_{VDD} = 1.8\text{ V}$ , generated by the internal LDO of the TLV320AIC, the maximum allowed input voltage is  $2.1\text{ V}$ . To protect the ADC, a voltage divider with  $R_1 = 1.8\text{ k}\Omega$  and  $R_2 = 12\text{ k}\Omega$  is used, resulting in an attenuation factor of  $0.15$ . For a maximum input signal of  $\pm 10\text{ V}$ , the attenuated voltage reaches approximately  $2.11\text{ V}$ , which is just within the ADC limits. Such high input levels are uncommon in practice, as typical Eurorack audio signals are within  $\pm 5\text{ V}$ .

### 0.2.2 Noise Analysis

**Inverted Input circuit** In order to analyse the noise behaviour of an amplifier circuit, the respective noise components have to be calculated. In an inverted input op-amp circuit, the noise components contain of:

- Resistor Noise (Johnson Noise)
- op-amp voltage input noise
- op-amp current input noise

First the resistor noise aspects are analysed.

All resistors have a voltage noise given by:

$$V_{NR} = \sqrt{4kTBR}$$

where:

- $T$  = absolute temperature in Kelvin,  $T = T(^{\circ}\text{C}) + 273.15$
- $B$  = bandwidth in Hz
- $k$  = Boltzmann's constant,  $k = 1.38 \times 10^{-23}$  J/K
- $R$  = resistance in Ohms

With values in the circuit:  $R_{in} = 22\text{k}\Omega$  and  $R_f = 3.3\text{k}\Omega$ .

This yield following Johnson's noise densities:

For  $R_{in}$

$$V_{Rin} = 155.846\text{nV}/\sqrt{\text{Hz}}$$

and for the feedback resistor  $R_f$

$$V_{Rf} = 60.359\text{nV}/\sqrt{\text{Hz}}$$

at  $T(^{\circ}\text{C}) = 25^{\circ}\text{C}$  and a bandwidth of 20kHz

### 0.3 Crystal

#### 0.3.1 Calculation to Validate the Crystal Compatibility with STM32

This section follows the procedure and calculations derived from [1].

To ensure the crystal provides enough energy to excite and sustain oscillation, the following condition must be satisfied:

$$\text{Gain Margin} = \frac{g_m}{g_{m,\text{crit}}} \geq 5$$

or equivalently:

$$\text{Gain Margin} = \frac{g_{m,\text{crit max}} \cdot 5}{g_{m,\text{crit}}} \geq 5$$

where  $g_m$  is the oscillator transconductance of the product. Each STM32 device specifies  $g_m$  in the datasheet. The critical transconductance of the crystal is:

$$g_{m,\text{crit}} = 4 \cdot ESR \cdot (2\pi F)^2 \cdot (C_0 + C_L)^2$$

with

$$ESR = 100 \, \Omega, \quad F = 24 \, \text{MHz}, \quad C_0 = 7 \, \text{pF}, \quad C_L = 10 \, \text{pF}$$

For the STM32H7:

$$g_{m,\text{crit max}} = 1.5 \, \text{mA/V}$$

[2, p. 107]

Calculating the critical transconductance:

$$g_{m,\text{crit}} = 1.0254 \, \text{mA/V}$$

and the gain margin:

$$\text{Gain Margin} = \frac{g_{m,\text{crit max}} \cdot 5}{g_{m,\text{crit}}} = \frac{1.5 \cdot 5}{1.0254} \approx 7.31$$

Since the gain margin is greater than 5, the crystal is compatible with the STM32H7.

#### 0.3.2 Calculation of Capacitors $C_{L1}$ and $C_{L2}$

Since the chosen crystal is compatible with the STM32H7 MCU, the appropriate external load capacitor values need to be determined.

The relation between the crystal's load and shunt capacitance and the external load capacitors is given by:

$$C_L - C_0 = \frac{C_{L1} \cdot C_{L2}}{C_{L1} + C_{L2}}$$

Since the external load capacitors are generally the same  $C_{L1} = C_{L2}$  results in:

$$C_L - C_s = \frac{C_{L1}}{2} = \frac{C_{L2}}{2}$$

which can be rewritten as:

$$C_{L1} = C_{L2} = 2 \cdot (C_L - C_s)$$

The stray capacitance  $C_s$  accounts for the MCU pin capacitance and parasitic PCB capacitance. It typically ranges between 2–3 pF. Many STM32 datasheets assume  $C_s = 5 \, \text{pF}$  for a conservative estimate. For this calculation, we take  $C_s = 3 \, \text{pF}$ .

For the NX2520SA-24.000000MHZ-B8 crystal with  $C_L = 10 \, \text{pF}$ , the external load capacitors are:



$$C_{L1} = C_{L2} = 2 \cdot (10pF - 3pF) = 14pF$$

This results in a practical value that is readily available on the market:

<https://www.mouser.de/ProductDetail/KYOCERA-AVX/04023U140GAT2A?qs=ZwvTZoB9S08Z%2FQAWMtnEbw%3D%3D>

## References

- [1] *How to select a compatible crystal and load capacitors for STM32 with layout guidelines*, <https://community.st.com/t5/mcus/how-to-select-a-compatible-crystal-and-load-capacitors-for-stm32/ta-p/780236>, Apr. 2025. (visited on 12/29/2025).
- [2] STMicroelectronics, *Stm32h7a3ri Datasheet*.