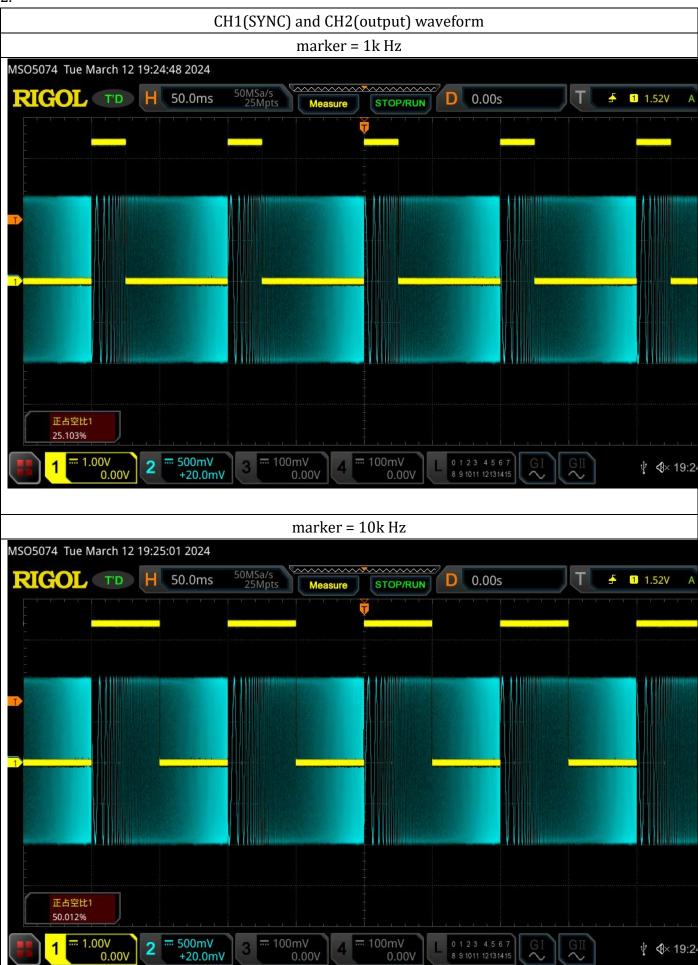
REPORT

Experiment 0: Basic AC Sweep Configuration

1.



2.



Lab03 **Audio Frontend** marker = 100k HzMSO5074 Tue March 12 19:25:11 2024 50MSa/s 25Mpts RIGOL TD 50.0ms D 0.00s 1 1.52V STOP/RUN 正占空比1 75.018% 500mV +20.0mV 100mV 0.00V 100mV 0.00V 0 1 2 3 4 5 6 7 8 9 1011 12131415 1.00V 0.00V 2 **∲ ∜**× 19:24 marker = 1M Hz MSO5074 Tue March 12 19:25:20 2024 50MSa/s 25Mpts RICOL T'D 50.0ms D 0.00s 1.52V STOP/RUN

> 100mV 0.00V

.... 100mV 0.00V 0 1 2 3 4 5 6 7 8 9 1011 12131415

∲ ∮× 19:24

正占空比1

--- 1.00V 0.00V == 500mV +20.0mV

2

Question:

What did you find?

I found that the duty cycle of the marker increased alongside the marker frequency.

What is the main purpose of the marker frequency?

The marker is used to locate the 3dB frequency. We must match it with the waveform at 0.707 of its max amplitude.

What is the meaning of the "fish" waveform?

The "fish" is composed of sine waves of different frequency from 1 Hz to 100k Hz. The lowest frequency is on the left and the highest frequency is on the right. The leftmost waves represent the maximum gain provided by the filter. The higher frequency waves to the right are attenuated by the filter. That's why the right side of the "fish" starts shrinking.

How do we find the 3dB frequency?

• Vo / Freq. Plot

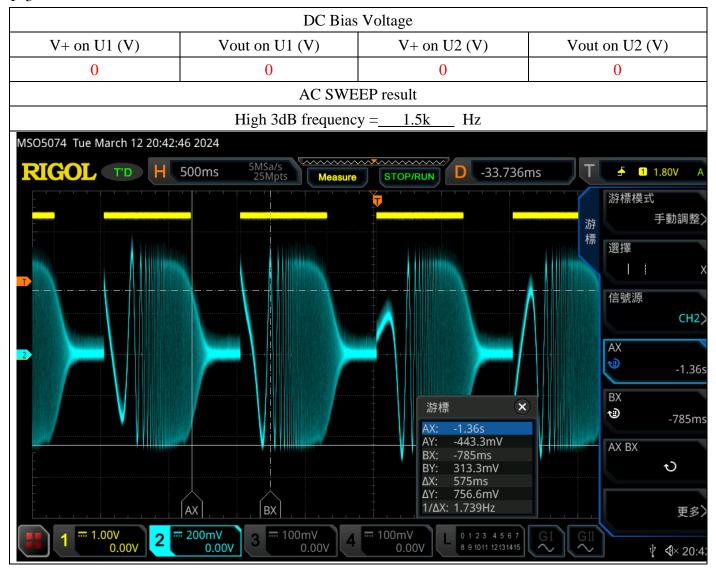
- 1. Find the maximum amplitude (not Vpp) of the waveform
- 2. Multiply this value by $\frac{1}{\sqrt{2}}$ (0.707)
- 3. Adjust the horizontal cursor to match this new value
- 4. Adjust the vertical cursor to the intersection between the horizontal cursor and the waveform
- 5. Change the marker frequency on the FG so that the marker's edge intersects with the vertical cursor
- 6. The marker frequency is approximately the 3dB frequency

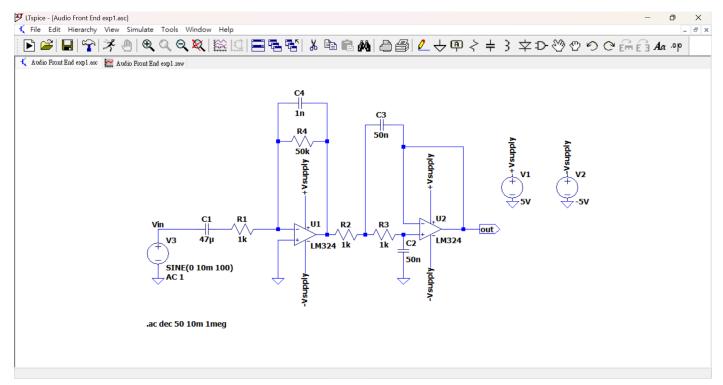
Bode Plot

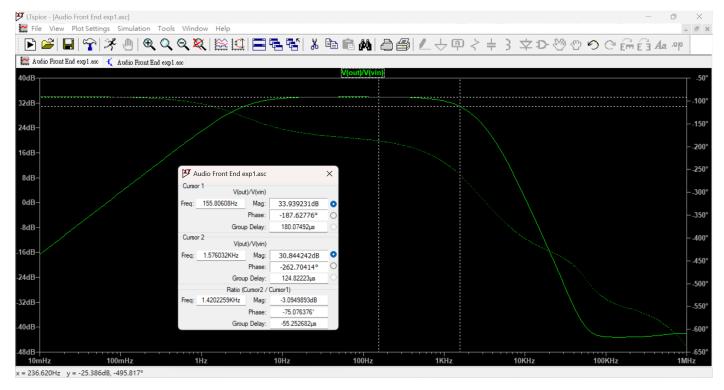
- 1. Place one cursor at the highest point of the plot, marking the maximum gain.
- 2. Slide the second cursor to the point where there's a -3dB difference between it and the 1st cursor.
- 3. The frequency marked by the 2^{nd} cursor is approximately the 3dB frequency

Experiment 1: Split Supply vs. Single Supply

1~3

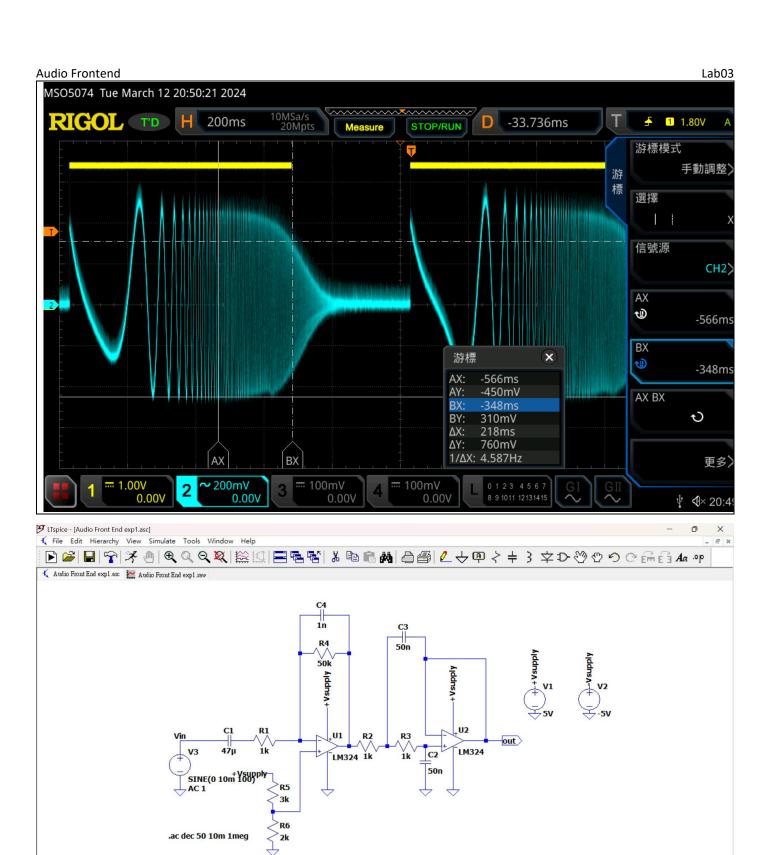


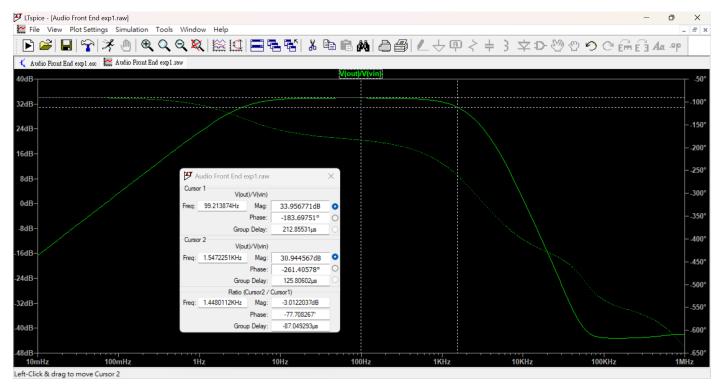




4~6

	DC Bias	Voltage	
V+ on U1 (V)	Vout on U1 (V)	V+ on U2 (V)	Vout on U2 (V)
1.98	1.98	1.98	1.98
	AC SWE	EP result	
	High 3dB frequency =	1.8k Hz	





Split Supply:

The split supply op-amps allows for both positive and negative signal swings (in this experiment, it accepts swings from -5V to 5V). This is required for applications such as audio amplifiers, instrumentation systems, and analog signal processing. They provide better symmetry and linearity for both positive and negative input/output signals.

Single Supply:

In a single supply op-amp, the input and output voltage ranges are limited to positive voltages only (0V to 5V in this experiment), usually referenced to ground (0V). In this case, any negative voltage swing in the input signal will be limited or clipped at ground level (0V), resulting in distortion. However, we applied an additional 2V at the non-inverting input. This provides an offset voltage to the signal and prevents it from being clipped.

Filter Terms:

Passband:

The passband is the range of frequencies in a filter's frequency response where the filter is designed to pass signals with minimal attenuation or signal loss.

> Stopband:

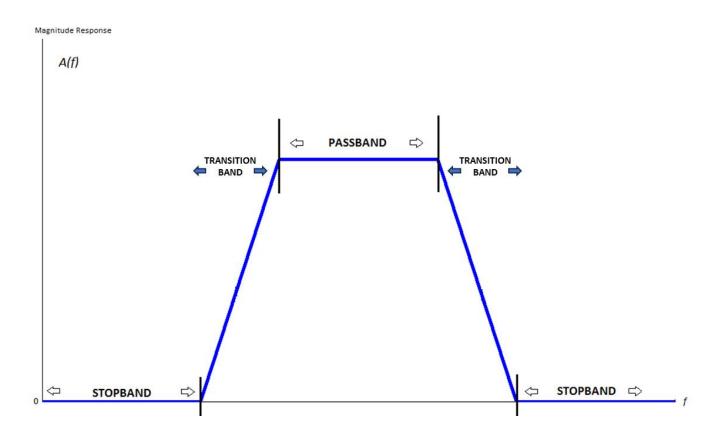
The stopband refers to the range of frequencies that the filter is intended to attenuate or reject. It is the frequency band where the filter's frequency response is expected to have a very low gain or magnitude.

> Transition Band (Roll-off Region):

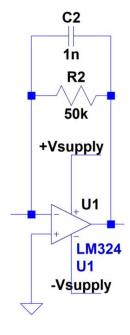
The frequency range in a filter's frequency response that lies between the passband and the stopband.

> Roll-off:

A filter roll-off refers to the rate at which the filter's frequency response transitions from the passband to the stopband. It describe how quickly the filter attenuates or rejects frequencies beyond the cutoff frequency. The roll-off is typically expressed in decibels per octave (dB/octave) or decibels per decade (dB/decade).



First-Order Low-Pass Filter:



1. Frequency Response: The first-order low-pass filter has a single pole in the transfer function, resulting in a slope of -20 dB/decade or -6 dB/octave in the stop-band region.

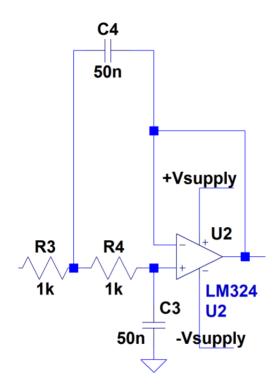
- 2. Roll-off Rate: The roll-off rate of a first-order filter is relatively gradual, with a slope of -20 dB/decade or -6 dB/octave.
- 3. Attenuation: The maximum attenuation in the stop-band is limited, approaching -40 dB at very high frequencies.
- 4. Phase Response: -45° at the cutoff frequency.
- 5. -3dB Cutoff Frequency:

$$f_0 = \frac{1}{2\pi RC}$$

Second-Order Low-Pass Filter:

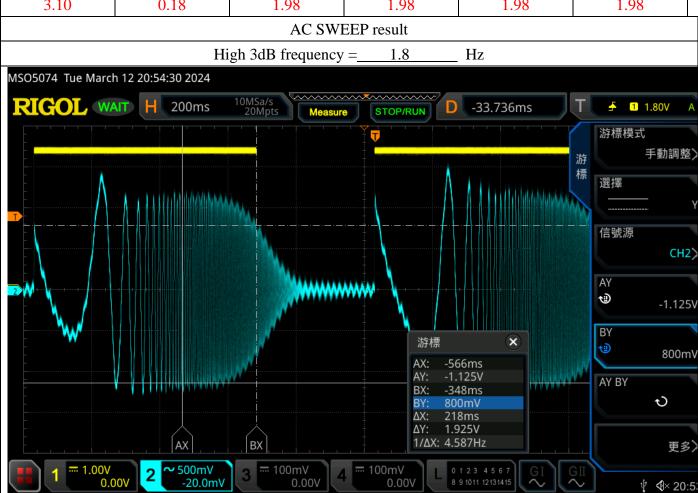
- 1. Frequency Response: The second-order low-pass filter has two poles in the transfer function, resulting in a steeper slope of -40 dB/decade or -12 dB/octave in the stop-band region.
- 2. Roll-off Rate: The roll-off rate of a second-order filter is steeper, with a slope of -40 dB/decade or -12 dB/octave, providing better rejection of unwanted high frequencies.
- 3. Attenuation: The maximum attenuation in the stop-band can be much higher than a first-order filter, depending on the filter design and specs.
- 4. Phase Response: -90° at the cutoff frequency.
- 5. -3dB Cutoff Frequency:

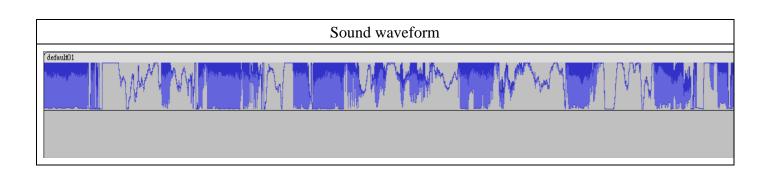
$$f_0 = \frac{1}{2\pi\sqrt{R_1R_2C_1C_2}}$$



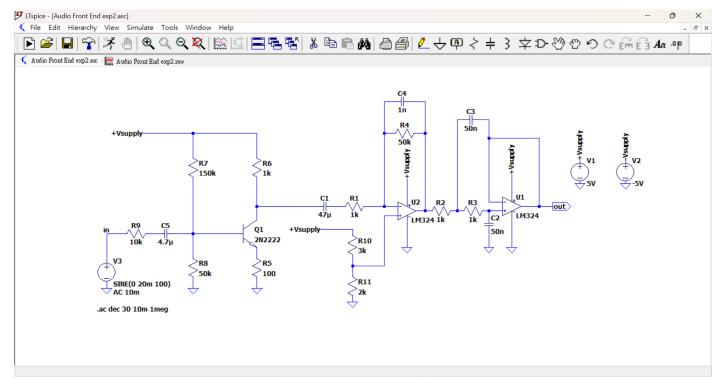
Experiment 2: Audio Front End

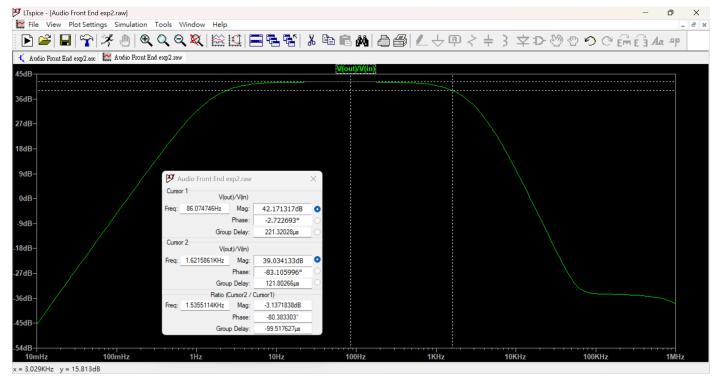
DC Bias Voltage							
Vc (V)	Ve (V)	V+ on U1 (V)	Vout on U1	V+ on U2 (V)	Vout on U2		
			(V)		(V)		
3.10	0.18	1.98	1.98	1.98	1.98		





Play your .wav file to TA





Observation:

The npn transistor serves as an additional amplifier that increases the maximum gain of the system to 42dB.

Usually, a normal recording using the microphone on my laptop has both positive and negative swings. However, the recording from this experiment uses the single supply circuit in Exp1, so the entire signal is offset by 2V. Without this 2V provided by the voltage divider at the non-inverting input, there would have been a loss of information due to clipping. In IC design, single supply amplifier designs like this are often preferred over split supply amplifiers because it's easier, cheaper, and more area-efficient to implement.

Reference:

- 1. Engineering.com: https://www.engineering.com/Ask@/qactid/1/qaqid/2083.aspx
- 2. ROHM Semiconductor: https://www.rohm.com/products/faq-search/faqId/1614
- 3. Learning About Electronics: https://www.learningaboutelectronics.com/Articles/Difference-between-a-single-and-dual-supply-op-amp.php