Electronics Operational Amplifier Dynamic Characteristics

Terry Sturtevant

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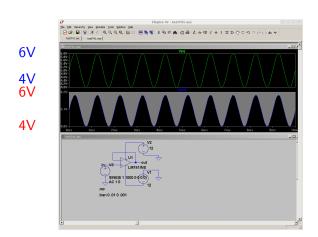
September 4, 2014

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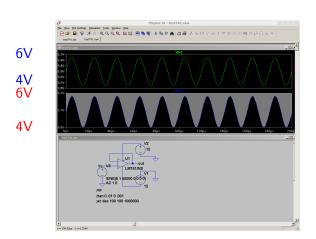
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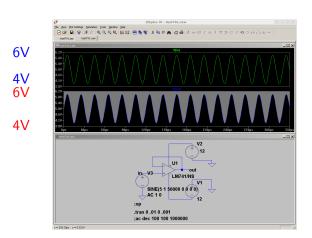
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- The open loop gain is the gain without feedback.
- Open loop gain is constant at low frequencies,
 but then starts to decrease at higher frequencies.



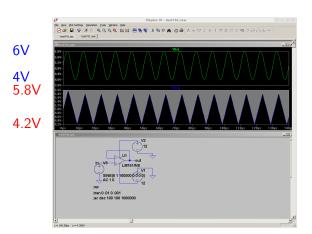
Here's a voltage follower with a 1 kHz input.



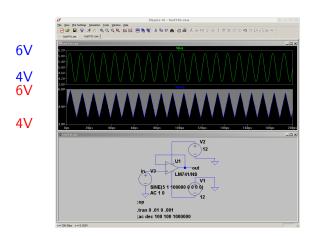
With a 50 kHz input, the output still matches pretty closely.



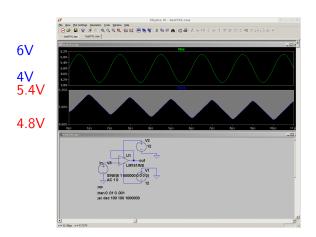
Here it is on the same scale.



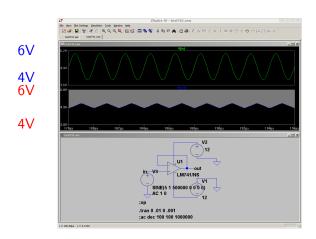
By 100kHz, the output is starting to diverge from the input.



It's more obvious if the scales are the same.

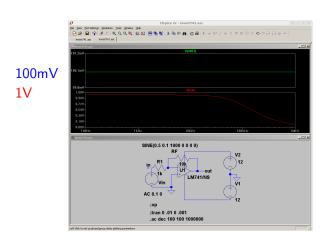


With a 500kHz input, the output change is very clear.

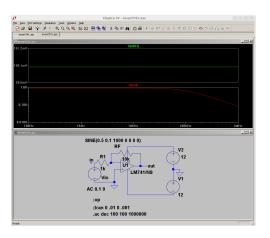


On the same scale, it's obvious.

The **rolloff** can be seen in the next figures.

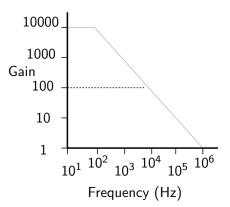


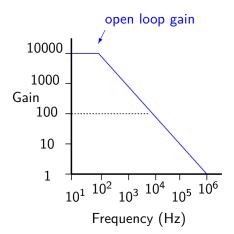
Here's what an AC analysis shows for a gain of 10.

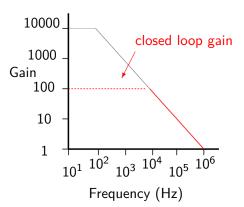


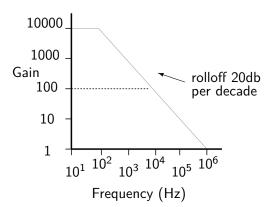
A logarithmic scale is helpful sometimes.

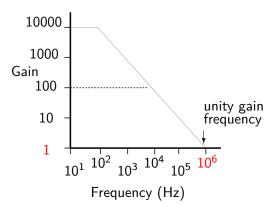
The rolloff affects both open-loop and closed-loop gain.

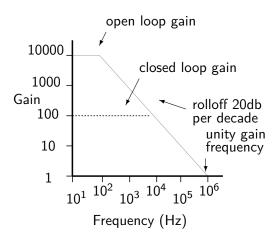












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- So

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- A **decade** is a factor of 10, so rolloff of 20 db/decade represents a *decrease* in $\left(\frac{V_o}{V_i}\right)$ of a factor of 10 as the frequency *increases* by a factor of 10.

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- The unity-gain frequency \rightarrow frequency at which $\left(\frac{V_o}{V_i}\right)=1$.
- (I will use the term A to refer to $\left(\frac{V_o}{V_i}\right)$ and G to denote the gain in decibels, as described above. This convention is common, but not universal, so when consulting information on the subject be sure to pay attention to how gain is expressed.)

• When negative feedback is employed in an op amp circuit, the **closed-loop gain** is decreased.

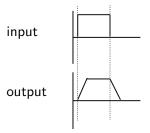
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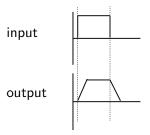
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- As frequency increases, and the open-loop gain decreases due to rolloff, the values of open-loop and closed-loop gain eventually coincide.
- From this point on, the closed loop gain also begins to experience rolloff at the same rate.
- Thus the unity-gain frequency for an op amp circuit is the same, regardless of the mid-band gain.

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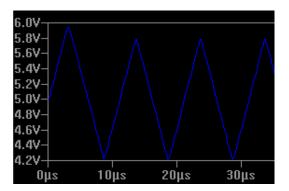
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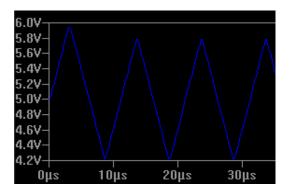
The *rate* of possible change of the output voltage is referred to as the **slew rate**.

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The rise (or fall) is about 1.6 volts in 5 μ S, or about 0.3 V/ μ S.

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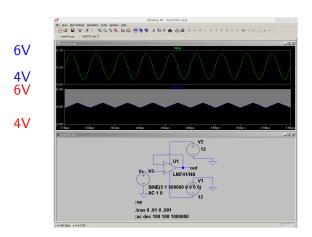
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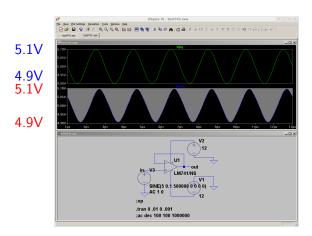
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Sine waves will be the *least* distorted of any waveform.



Here's a previous circuit; 1V, 500kHz into a voltage follower.



Reducing the input to 0.1V makes the output much closer.

Active filters

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- These are called active filters.

High Pass Filters

Band Pass Filters

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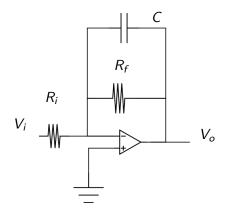
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- (Note: there are numerous designs for filters; the ones shown here are simple to analyze.)

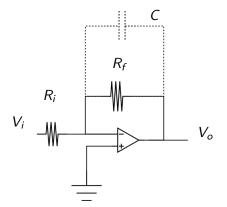
Low Pass Filters

High Pass Filters Band Pass Filters

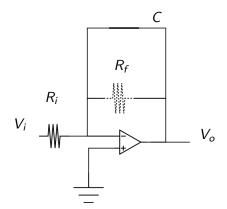
Low pass filter



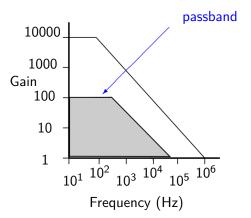
Low pass filter at low frequencies



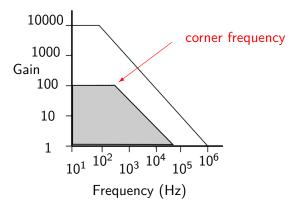
Low pass filter at high frequencies



Low pass filter Bode plot



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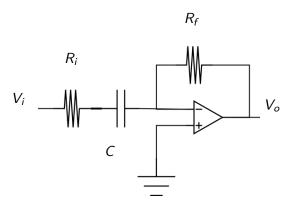
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- A **high pass filter** allows frequencies *above* a certain corner frequency, f_c to pass, and attenuates those *below*.
- As for the low pass filter, the corner frequency is not an absolute limit; it is the point at which amplification begins.
- The high pass filter has a limitation due to the rolloff of the op amp.
- As with any amplifier circuit, it is never possible to have gain above the open loop gain of the op amp, so any high pass filter is actually a **band pass filter**, which will be discussed below.

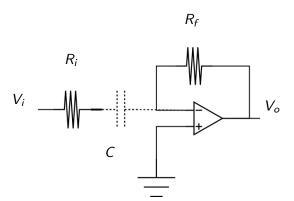
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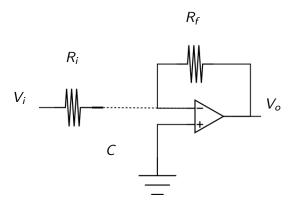
High pass filter



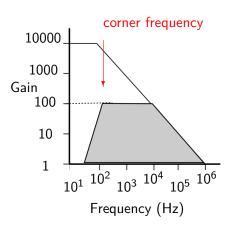
High pass filter at *low* frequencies



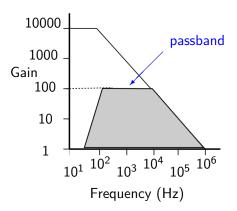
High pass filter at high frequencies



High pass filter Bode plot



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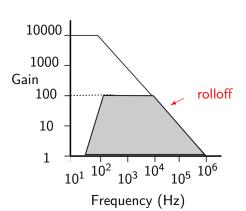


Low Pass Filters

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High pass filter Bode plot



In the input part of the circuit, at high frequencies $X_c \approx 0$ and so the gain is that of a normal inverting amplifier. As the frequency decreases, at a **corner frequency** $X_c = R_i$. Below this frequency,

$$Z \propto \frac{1}{f}$$

since $X_c > R_i$.

Low Pass Filters High Pass Filters Band Pass Filters

A band pass filter combines a low pass and a high pass filter to allows frequencies between two corner frequencies, f_{c1} and f_{c2} to pass, and attenuates those outside that range. The next figure shows the Bode plot for a band pass filter.

Bandpass filter Bode plot

