

# Nonlinear Op-Amp Circuits

- Most typical applications require op amp and its components to act linearly
  - I-V characteristics of passive devices such as resistors, capacitors should be described by linear equation (Ohm's Law)
  - For op amp, linear operation means input and output voltages are related by a constant proportionality ( $A_v$  should be constant)
- Some application require op amps to behave in nonlinear manner (logarithmic and antilogarithmic amplifiers)

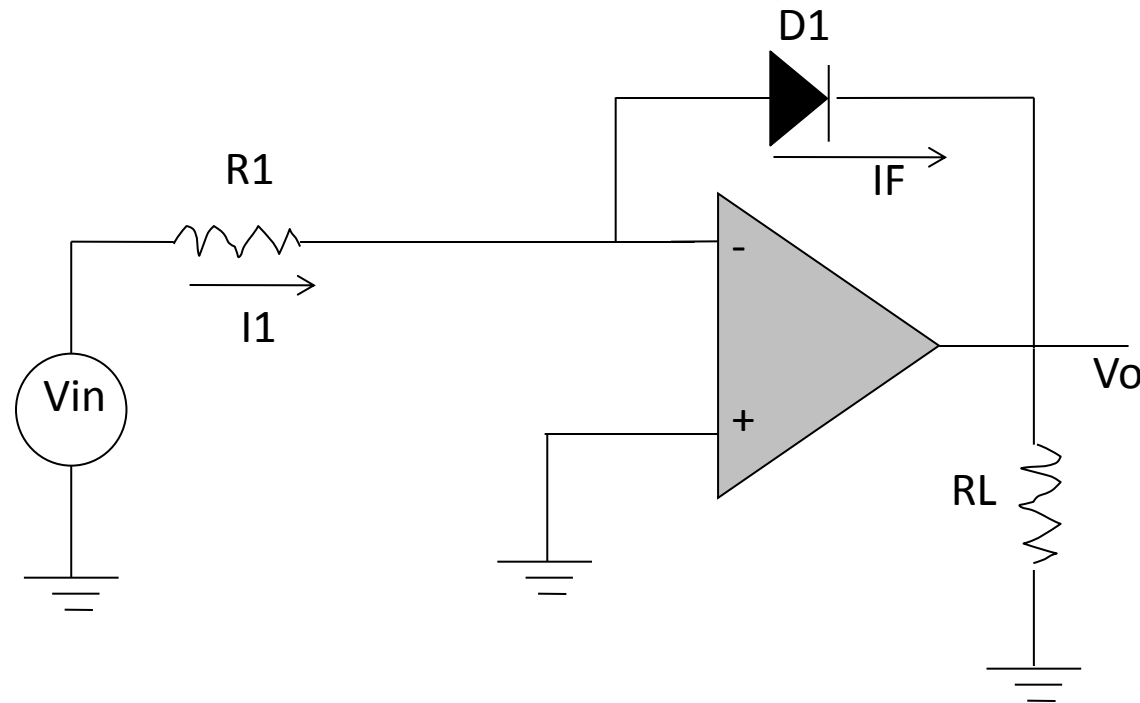
# Logarithmic Amplifier

- Output voltage is proportional to the logarithm of input voltage
- A device that behaves nonlinearly (logarithmically) should be used to control gain of op amp
  - Semiconductor diode
- Forward transfer characteristics of silicon diodes are closely described by Shockley's equation

$$I_F = I_S e^{(V_F/\eta V_T)}$$

- $I_S$  is diode saturation (leakage) current
- $e$  is base of natural logarithms ( $e = 2.71828$ )
- $V_F$  is forward voltage drop across diode
- $V_T$  is thermal equivalent voltage for diode (26 mV at 20°C)
- $\eta$  is emission coefficient or ideality factor (2 for currents of same magnitude as  $I_S$  to 1 for higher values of  $I_F$ )

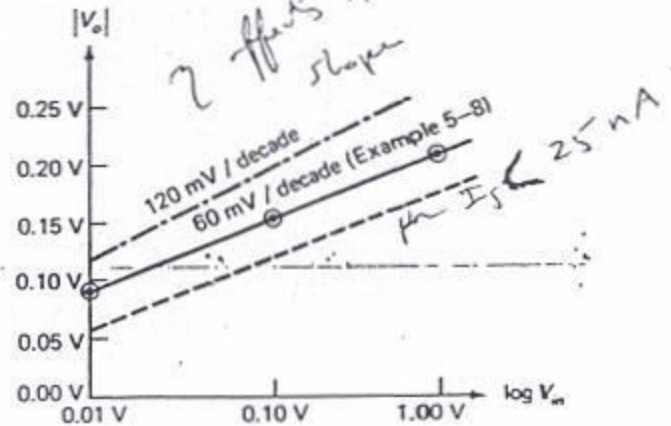
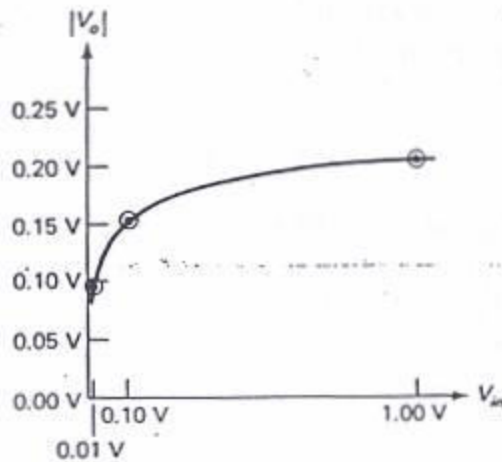
# Basic Log Amp operation



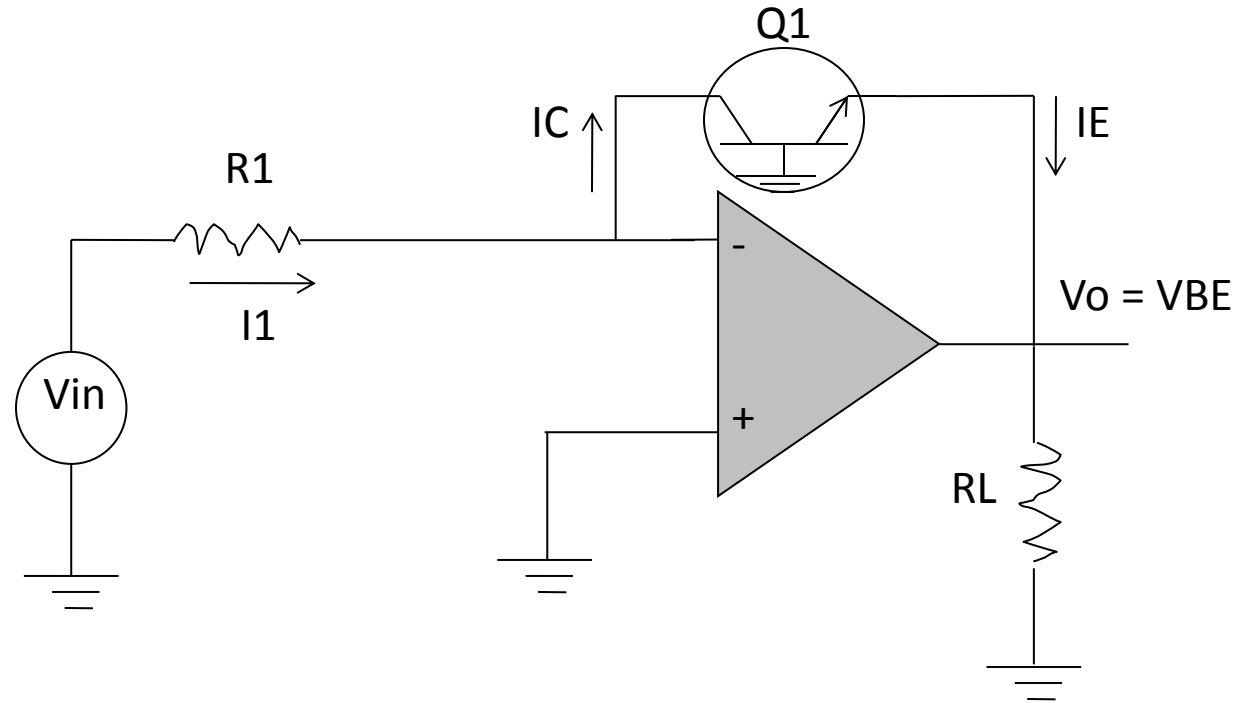
- $I1 = V_{in}/R1$
  - $I_F = - I1$
  - $I_F = - V_{in}/R1$
  - $V_o = -V_F = -\eta V_T \ln(I_F/I_S)$
  - $V_o = -\eta V_T \ln[V_{in}/(R1 I_S)]$
  - $r_D = 26 \text{ mV} / I_F$
  - $I_F < 1 \text{ mA}$  (log amps)
- At higher current levels ( $I_F > 1 \text{ mA}$ ) diodes begin to behave somewhat linearly

# Logarithmic Amplifier

- Linear graph: voltage gain is very high for low input voltages and very low for high input voltages
- Semilogarithmic graph: straight line proves logarithmic nature of amplifier's transfer characteristic
- Transfer characteristics of log amps are usually expressed in terms of slope of  $V_o$  versus  $V_{in}$  plot in millivolts per decade
- $\eta$  affects slope of transfer curve;  $I_s$  determines the y intercept



# Additional Log Amp Variations



$$I_C = I_{E_S} e^{(V_{BE}/V_T)}$$

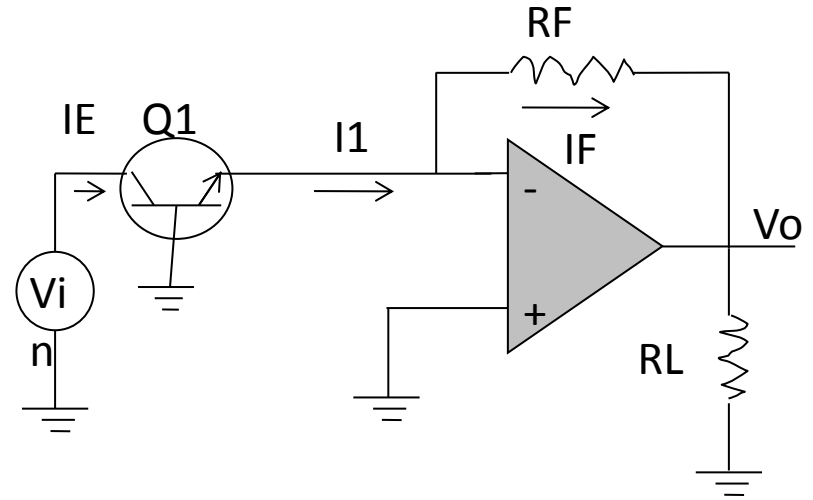
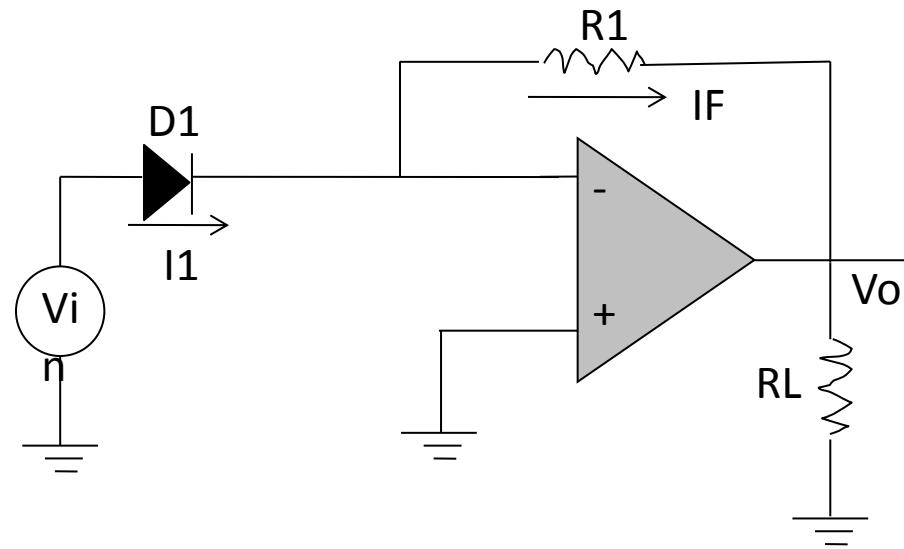
- $I_{E_S}$  is emitter saturation current
- $V_{BE}$  is drop across base-emitter junction

- Often a transistor is used as logging element in log amp (transdiode configuration)
- Transistor logging elements allow operation of log amp over wider current ranges (greater dynamic range)

# Antilogarithmic Amplifier

- Output of an antilog amp is proportional to the antilog of the input voltage
- with diode logging element
  - $V_0 = -R_F I_S e^{(V_{in}/V_T)}$
- With transdiode logging element
  - $V_0 = -R_F I_{ES} e^{(V_{in}/V_T)}$
- As with log amp, it is necessary to know saturation currents and to tightly control junction temperature

# Antilogarithmic Amplifier



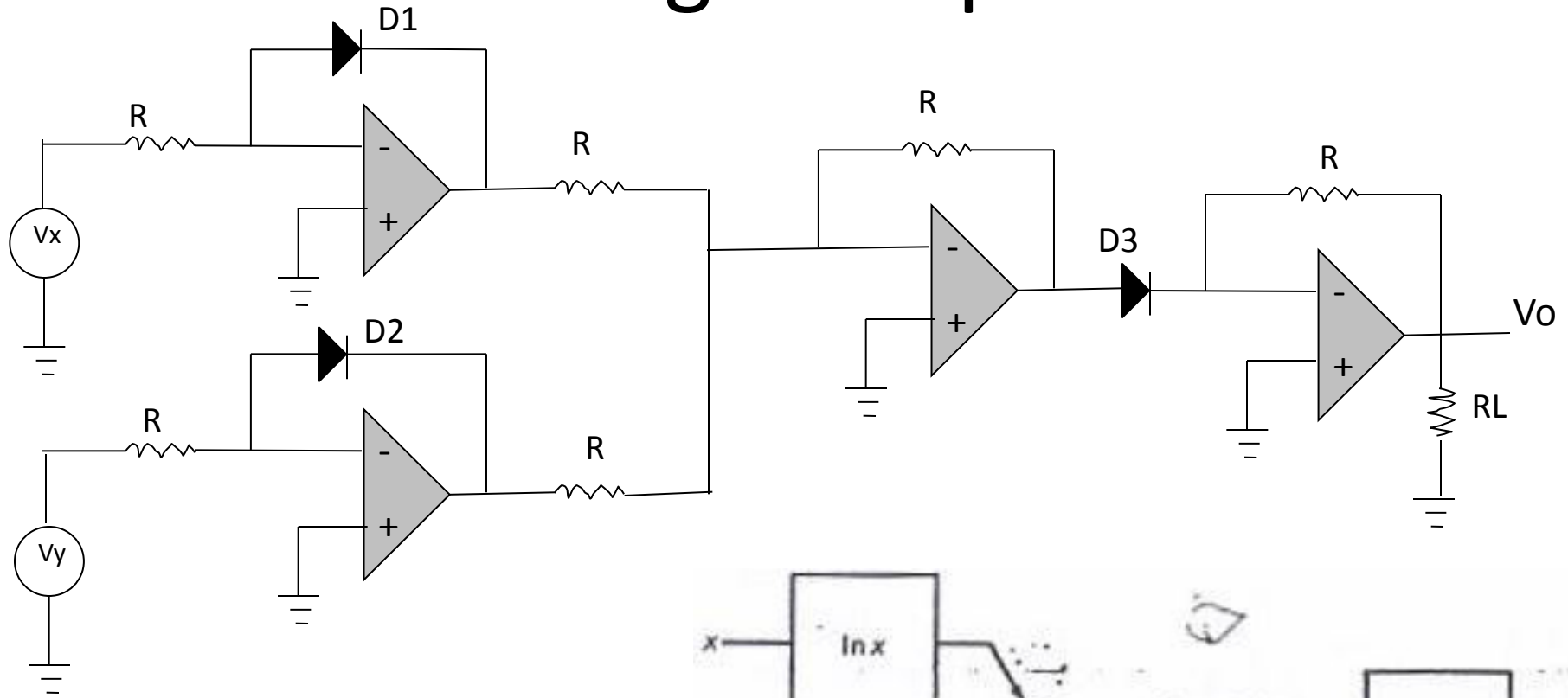
$$(\alpha = 1) I_1 = I_C = I_E$$

# Logarithmic Amplifier Applications

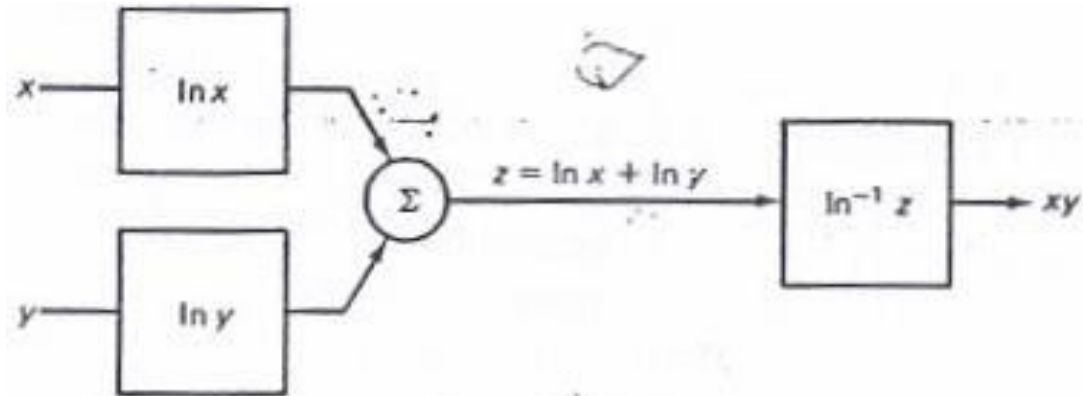
- Logarithmic amplifiers are used in several areas
  - Log and antilog amps to form analog multipliers
  - Analog signal processing
- Analog Multipliers
  - $\ln xy = \ln x + \ln y$
  - $\ln (x/y) = \ln x - \ln y$



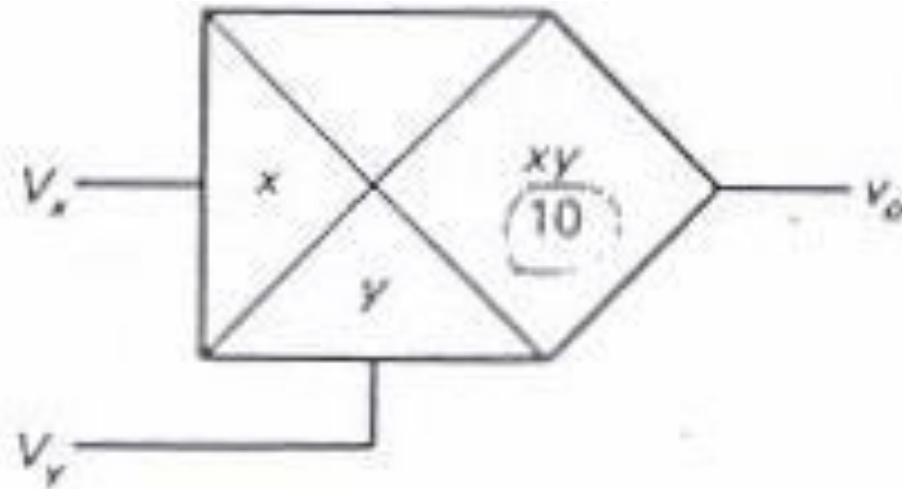
# Analog Multipliers



One-quadrant multiplier: inputs must both be of same polarity

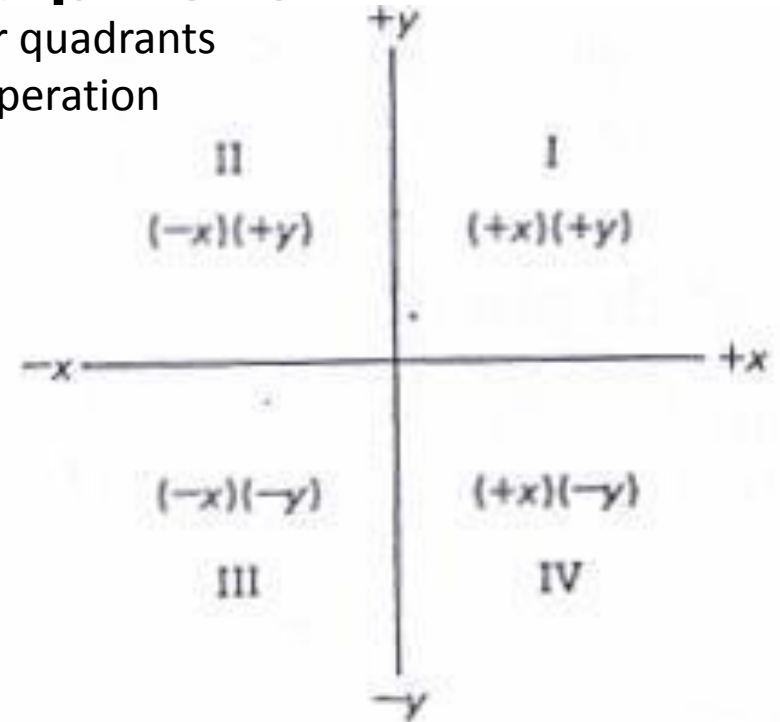


# Analog Multipliers



General symbol

Four quadrants  
of operation

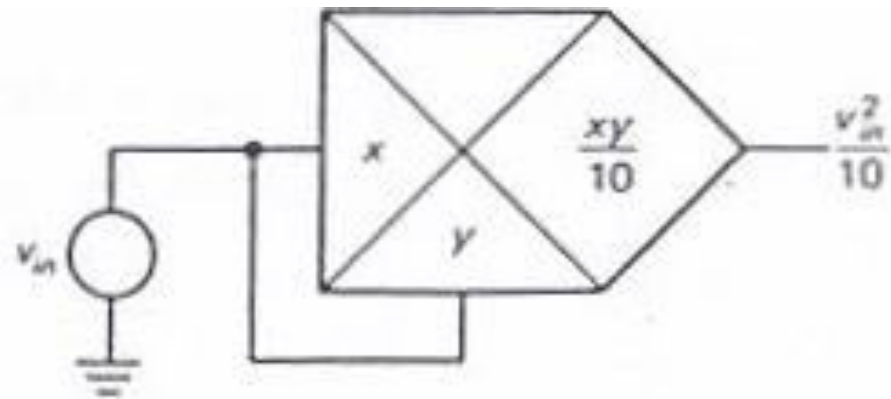


Two-quadrant multiplier: one input should have positive voltages, other input could have positive or negative voltages

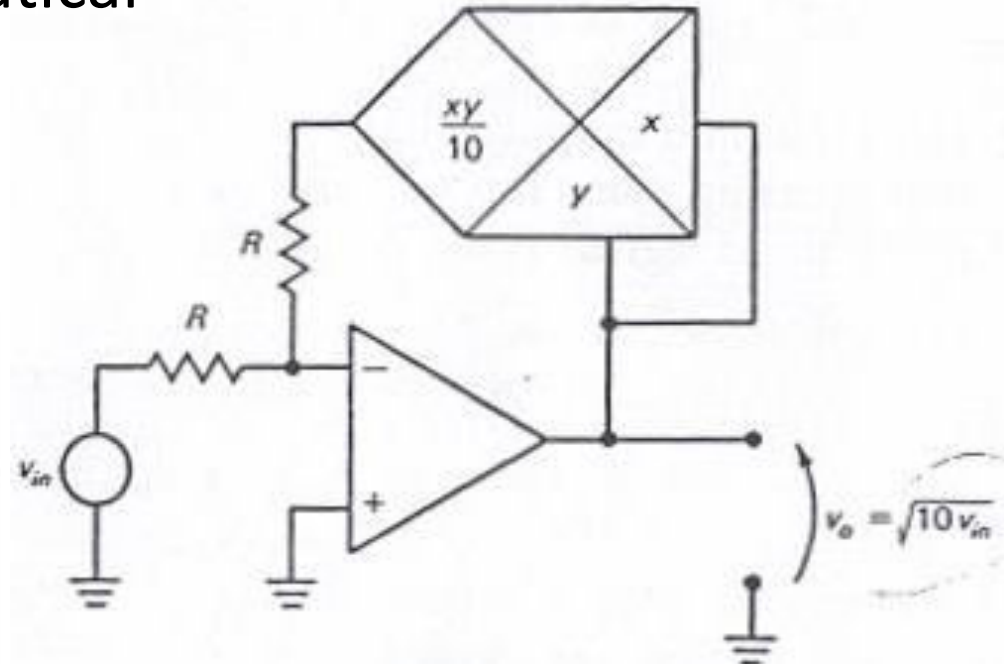
Four-quadrant multiplier: any combinations of polarities on their inputs

# Analog Multipliers

Implementation of mathematical operations



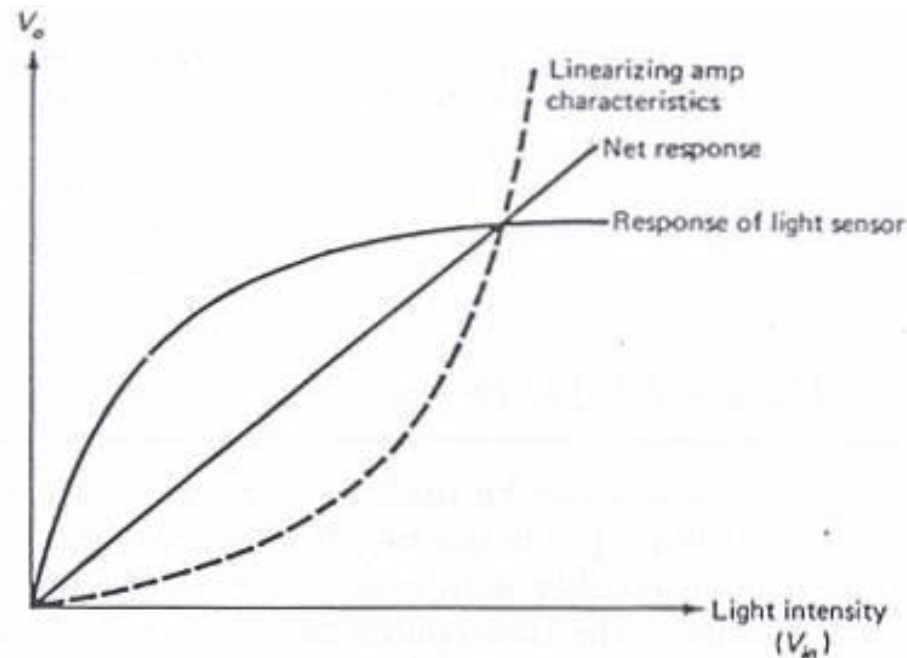
Squaring Circuit



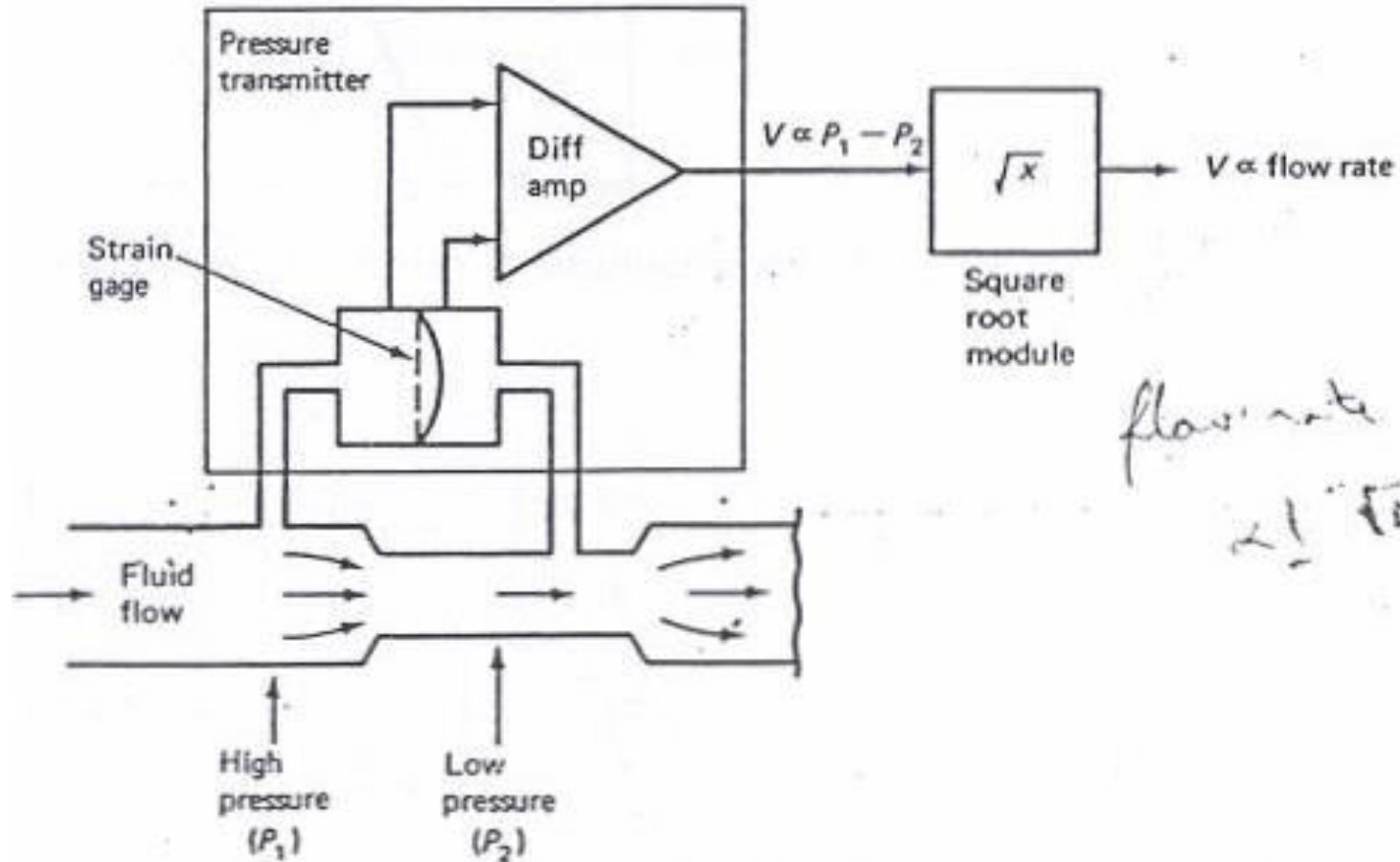
Square root Circuit

# Signal Processing

- Many transducers produce output voltages that vary nonlinearly with physical quantity being measured (thermistor)
- Often It is desirable to linearize outputs of such devices; logarithmic amps and analog multipliers can be used for such purposes
- Linearization of a signal using circuit with complementary transfer characteristics



# Pressure Transmitter



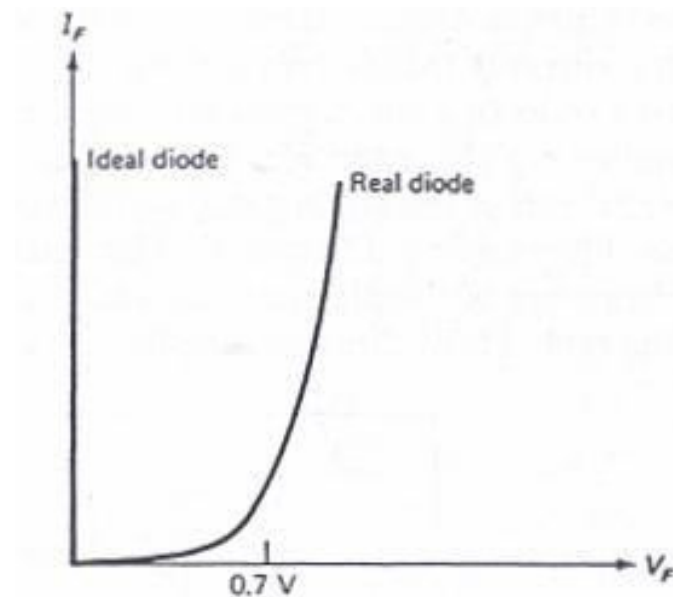
Pressure transmitter produces an output voltage proportional to difference in pressure between two sides of a strain gage sensor

# Pressure Transmitter

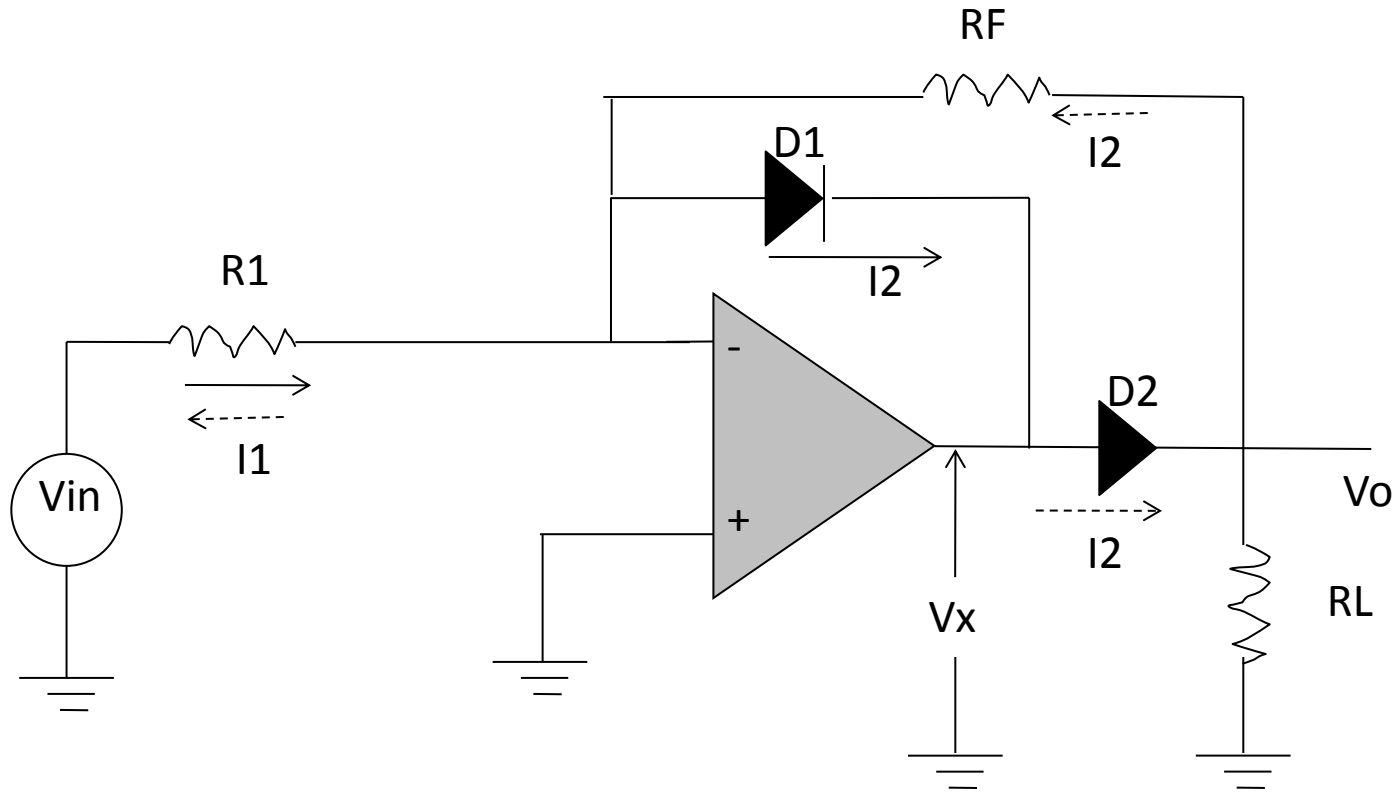
- A venturi is used to create pressure differential across strain gage
- Output of transmitter is proportional to pressure differential
- Fluid flow through pipe is proportional to square root of pressure differential detected by strain gage
- If output of transmitter is processed through a square root amplifier, an output directly proportional to flow rate is obtained

# Precision Rectifiers

- Op amps can be used to form nearly ideal rectifiers (convert ac to dc)
- Idea is to use negative feedback to make op amp behave like a rectifier with near-zero barrier potential and with linear I/O characteristic
- Transconductance curves for typical silicon diode and an ideal diode



# Precision Half-Wave Rectifier

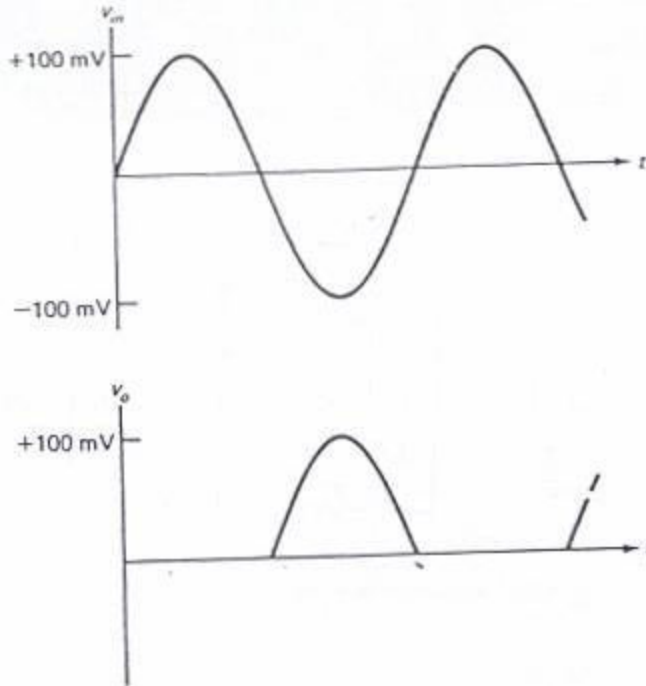


- Solid arrows represent current flow for positive half-cycles of  $V_{in}$  and dashed arrows represent current flow for negative half-cycles

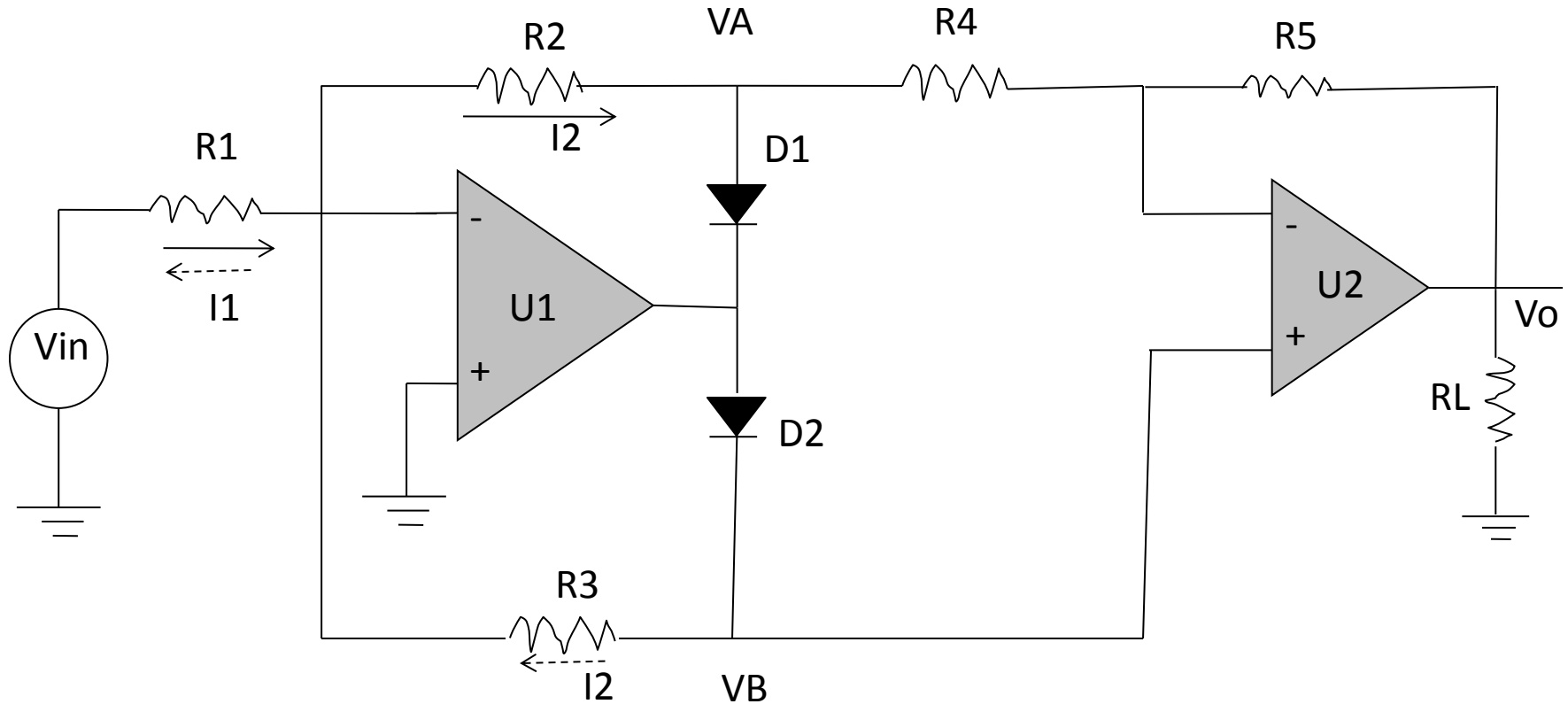


# Precision Half-Wave Rectifier

- If signal source is going positive, output of op amp begins to go negative, forward biasing  $D_1$ 
  - Since  $D_1$  is forward biased, output of op amp  $V_x$  will reach a maximum level of  $\sim -0.7V$  regardless of how far positive  $V_{in}$  goes
  - This is insufficient to appreciably forward bias  $D_2$ , and  $V_0$  remains at  $0V$
- On negative-going half-cycles,  $D_1$  is reverse-biased and  $D_2$  is forward biased
  - Negative feedback reduces barrier potential of  $D_2$  to  $0.7V/A_{OL}$  ( $\sim = 0$ )
  - Gain of circuit to negative-going portions of  $V_{in}$  is given by  $A_V = -R_F/R_1$



# Precision Full-Wave Rectifier



- Solid arrows represent current flow for positive half-cycles of  $V_{in}$  and dashed arrows represent current flow for negative half-cycles

# Precision Full-Wave Rectifier

- Positive half-cycle causes  $D_1$  to become forward-biased, while reverse-biasing  $D_2$ 
  - $V_B = 0 \text{ V}$
  - $V_A = -V_{in} R_2/R_1$
  - Output of  $U_2$  is  $V_0 = -V_A R_5/R_4 = V_{in} (R_2 R_5/R_1 R_4)$
- Negative half-cycle causes  $U_1$  output positive, forward-biasing  $D_2$  and reverse-biasing  $D_1$ 
  - $V_A = 0 \text{ V}$
  - $V_B = -V_{in} R_3/R_1$
  - Output of  $U_2$  (noninverting configuration) is
$$V_0 = V_B [1 + (R_5/R_4)] = -V_{in} [(R_3/R_1) + (R_3 R_5/R_1 R_4)]$$
  - if  $R_3 = R_1/2$ , both half-cycles will receive equal gain

# Precision Rectifiers

- Useful when signal to be rectified is very low in amplitude and where good linearity is needed
- Frequency and power handling limitations of op amps limit the use of precision rectifiers to low-power applications (few hundred kHz)
- Precision full-wave rectifier is often referred to as absolute magnitude circuit

# ACTIVE FILTERS

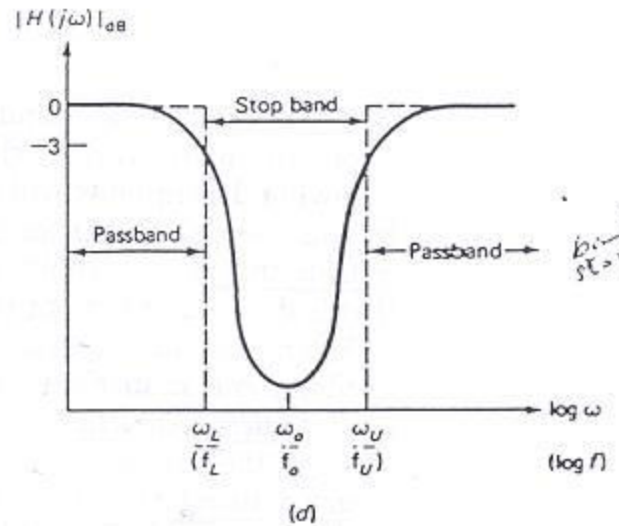
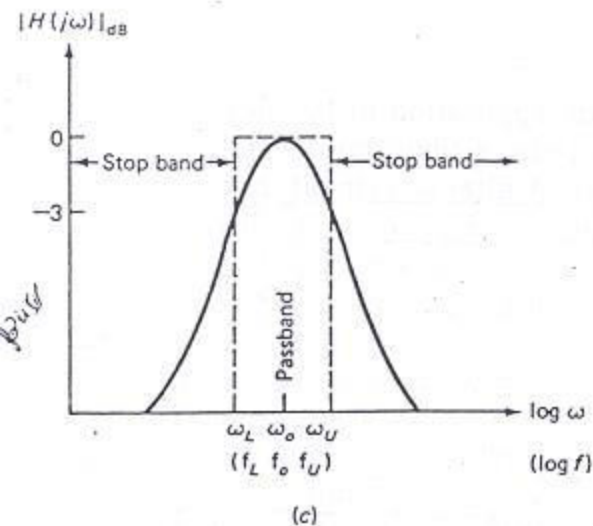
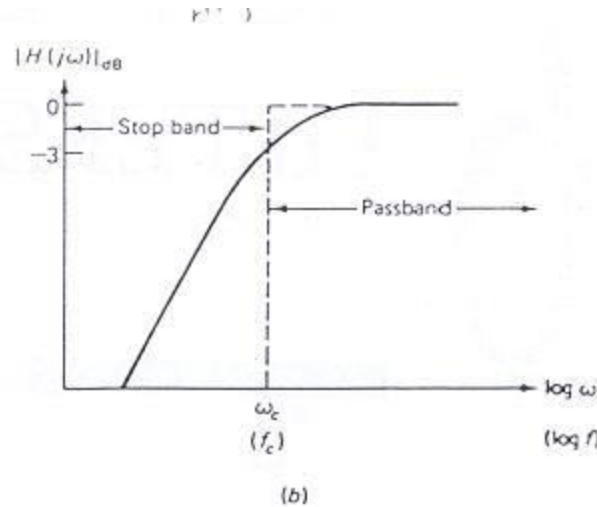
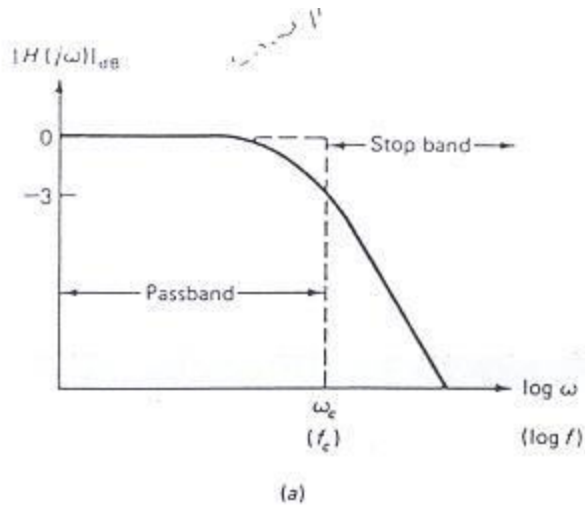
# Active Filters

- Op amps have wide applications in design of active filters
- Filter is a circuit designed to pass frequencies within a specific range, while rejecting all frequencies that fall outside this range
- Another class of filters are designed to produce an output that is delayed in time or shifted in phase with respect to filter's input
- Passive filters: constructed using only passive components (resistors, capacitors, inductors)
- Active filters: characteristics are augmented using one or more amplifiers; constructed using op amps, resistors, and capacitors only
  - Allow many filter parameters to be adjusted continuously and at will

# Filter Fundamentals

- Five basic types of filters
  - Low-pass (LP)
  - High-pass (HP)
  - Bandpass (BP)
  - Bandstop (notch or band-reject)
  - All-pass (or time-delay)

# Response Curves



- $\omega$  is in rad/s
- $|H(j\omega)|$  denotes frequency-dependent voltage gain of filter
- Complex filter response is given by

$$H(j\omega) = |H(j\omega)| \angle \theta(j\omega)$$

- If signal frequencies are expressed in Hz, filter response is expressed as  $|H(jf)|$



# Filter Terminology

- Filter passband: range of frequencies a filter will allow to pass, either amplified or relatively unattenuated
- All other frequencies are considered to fall into filter's stop band(s)
- Frequency at which gain of filter drops by 3.01 dB from that of passband determines where stop band begins; this frequency is called corner frequency ( $f_c$ )
- Response of filter is down by 3 dB at corner frequency (3 dB decrease in voltage gain translates to a reduction of 50% in power delivered to load driven by filter)
- $f_c$  is often called half-power point

# Filter Terminology

- Decibel voltage gain is actually intended to be logarithmic representation of power gain
- Power gain is related to decibel voltage gain as
  - $A_p = 10 \log (P_o/P_{in})$
  - $P_o = (V_o^2/Z_L)$  and  $P_{in} = (V_{in}^2/Z_{in})$
  - $A_p = 10 \log [(V_o^2/Z_L) / (V_{in}^2/Z_{in})]$
  - $A_p = 10 \log (V_o^2 Z_{in} / V_{in}^2 Z_L)$
  - If  $Z_L = Z_{in}$ ,  $A_p = 10 \log (V_o^2/V_{in}^2) = 10 \log (V_o/V_{in})^2$
  - $A_p = 20 \log (V_o/V_{in}) = 20 \log A_v$
- When input impedance of filter equals impedance of load being driven by filter, power gain is dependent on voltage gain of circuit only

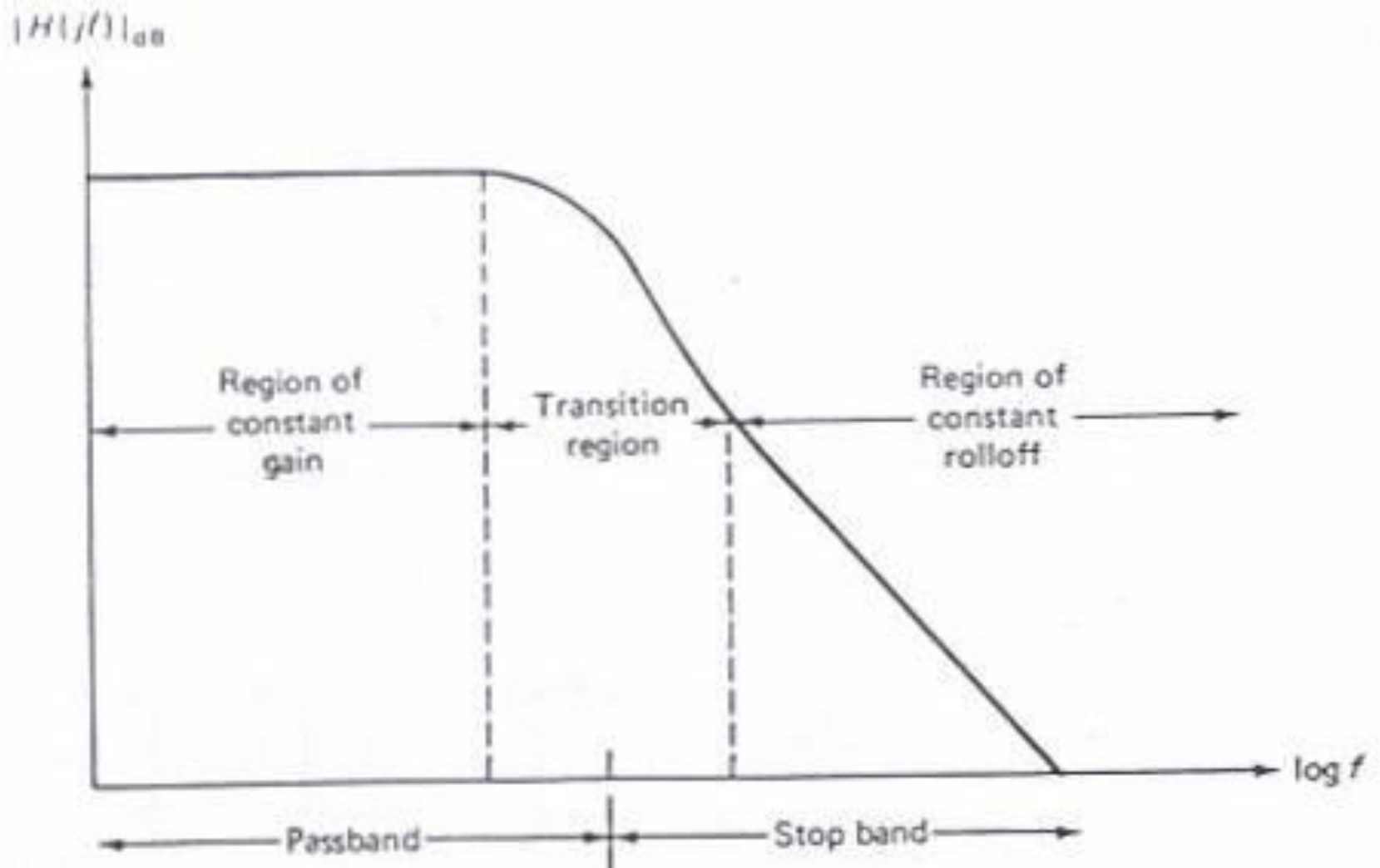
# Filter Terminology

- Since we are working with voltage ratios, gain is expressed as voltage gain in dB
  - $|H(j\omega)|_{dB} = 20 \log (V_o/V_{in}) = 20 \log A_V$
- Once frequency is well into stop band, rate of increase of attenuation is constant (dB/decade rolloff)
- Ultimate rolloff rate of a filter is determined by order of that filter
- 1<sup>st</sup> order filter: rolloff of -20 dB/decade
- 2<sup>nd</sup> order filter: rolloff of -40 dB/decade
- General formula for rolloff = -20n dB/decade (n is the order of filter)
- Octave is a twofold increase or decrease in frequency
- Rolloff = -6n dB/octave (n is order of filter)

# Filter Terminology

- Transition region: region between relatively flat portion of passband and region of constant rolloff in stop band
- Give two filter of same order, if one has a greater initial increase in attenuation in transition region, that filter will have a greater attenuation at any given frequency in stop band
- Damping coefficient ( $\alpha$ ): parameter that has great effect on shape of LP or HP filter response in passband, stop band, and transition region (0 to 2)
- Filters with lower  $\alpha$  tend to exhibit peaking in passband (and stopband) and more rapid and radically varying transition-region response attenuation
- Filters with higher  $\alpha$  tend to pass through transition region more smoothly and do not exhibit peaking in passband and stopband

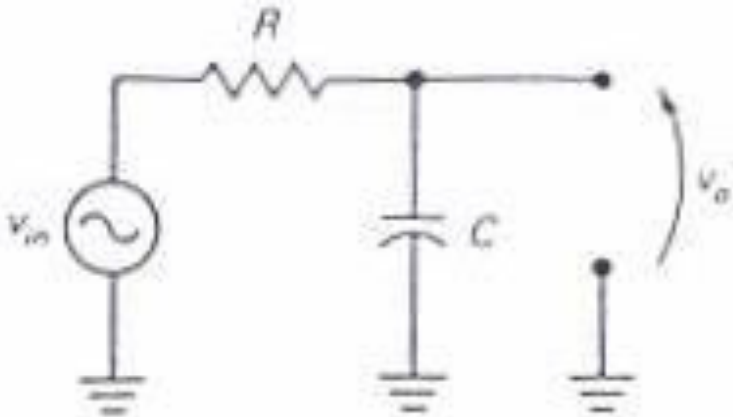
# LP Filter Response



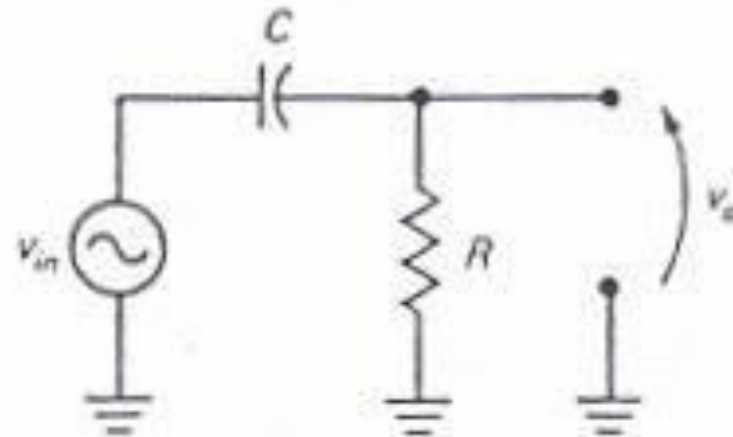
# Filter Terminology

- HP and LP filters have single corner frequency
- BP and bandstop filters have two corner frequencies ( $f_L$  and  $f_U$ ) and a third frequency labeled as  $f_0$  (center frequency)
- Center frequency is geometric mean of  $f_L$  and  $f_U$
- Due to log f scale,  $f_0$  appears centered between  $f_L$  and  $f_U$ 
$$f_0 = \text{sqrt}(f_L f_U)$$
- Bandwidth of BP or bandstop filter is
$$\text{BW} = f_U - f_L$$
- Also,  $Q = f_0 / \text{BW}$  (BP or bandstop filters)
- BP filter with high Q will pass a relatively narrow range of frequencies, while a BP filter with lower Q will pass a wider range of frequencies
- BP filters will exhibit constant ultimate rolloff rate determined by order of the filter

# Basic Filter Theory Review



(a)



(b)

- Simplest filters are 1<sup>st</sup> order LP and HP RC sections
  - Passband gain slightly less than unity
- Assuming negligible loading, amplitude response (voltage gain) of LP section is
 
$$H(j\omega) = (jX_C) / (R + jX_C)$$

$$|H(j\omega)| = X_C / \sqrt{R^2 + X_C^2} < \tan^{-1} (R/X_C)$$
- Corner frequency  $f_c$  for 1<sup>st</sup> order LP or HP RC section is found by making  $R = X_C$  and solving for frequency
 
$$R = X_C = 1/(2\pi fC)$$

$$1/f_c = 2\pi RC$$

$$f_c = 1/(2\pi RC)$$
- Gain (in dB) and phase response of 1<sup>st</sup> order LP
 
$$|H(jf)|_{dB} = 20 \log [1/\{\sqrt{1+(f/f_c)^2}\}] < \tan^{-1} (f/f_c)$$
- Gain (in dB) and phase response of 1<sup>st</sup> order HP
 
$$|H(jf)|_{dB} = 20 \log [1/\{\sqrt{1+(f_c/f)^2}\}] < \tan^{-1} (f_c/f)$$