

# Analog/Digital Guitar Synthesizer

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# Project Definition

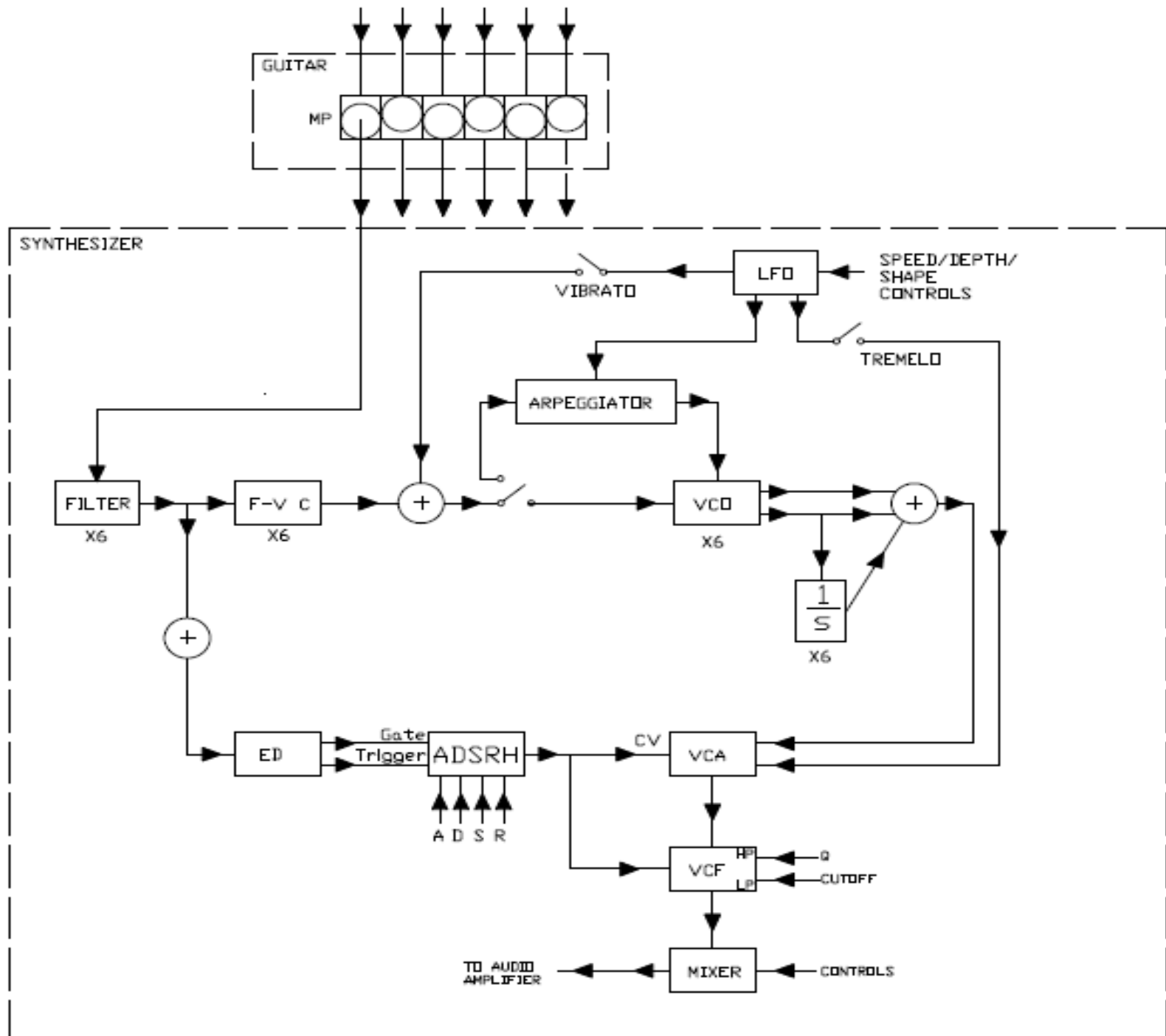
To use a guitar as a functional controller for an analog/digital synthesizer by taking information from a guitar as it's being played and create separate, fully controllable signals for each string.

# Goals/Requirements

- Musicality
- Accuracy – must sound in tune
- Playability – real time, to be used as a live instrument
- Portability – a musician must be able to take it on the road and set up quickly
- Controllable – must give the user options to modulate/control the synthesized signals by on-board controls

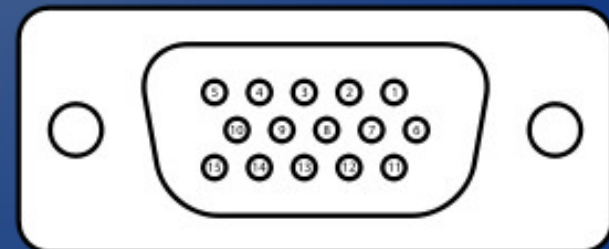
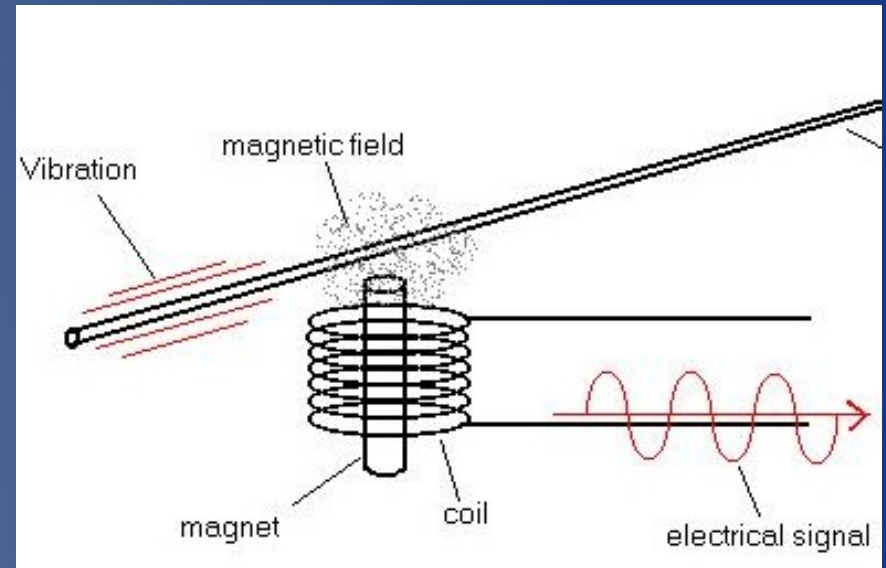
# Specifications

- Intonation – Within **3%** of the note value's frequency
- Synthesize full frequency range of guitar – **82.4Hz to 1046.5Hz**
- **+/- 12V** power supply
- Low Frequency oscillator with range from **3 to 33Hz**
- **<50ms** delay from initial input to final synthesized note
- Microcontroller CPU speed **1MHz** minimum



# Multichannel Magnetic Pickup

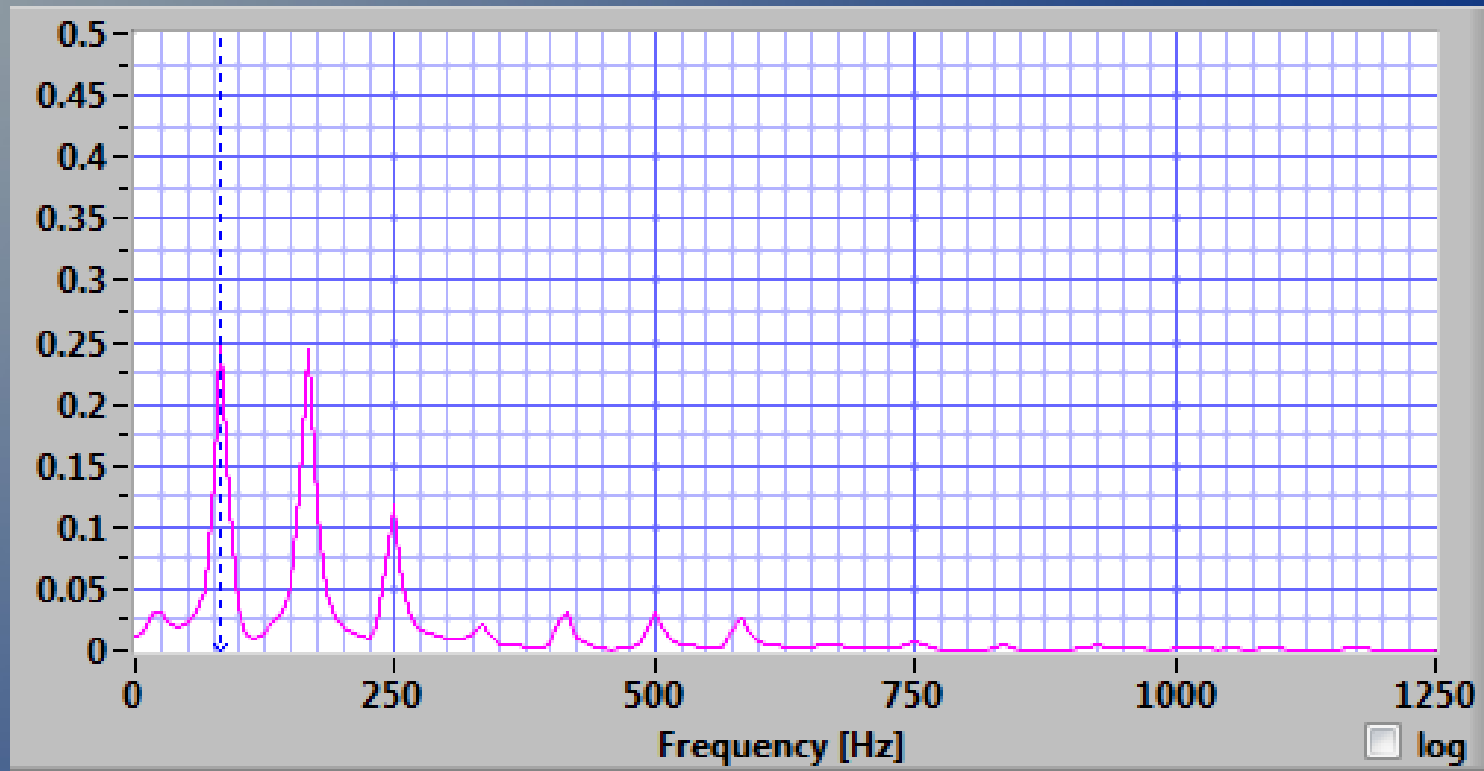
- Acquire the original envelope of the signal.
- Acquire the original frequency of the signal
- AlNiCo magnets are an industry standard.
- #30 g magnetic wire
- 6 individual signals must be produced.
- VGA cable has 15 pins.
- Problems can occur with a weak signal



# First Stage Filters

- Filter out naturally occurring harmonics from resonating guitar strings to allow for more accurate frequency tracking
- Must not completely filter out the 2nd harmonic, since this represents a valid note on the 12th fret (the octave)
- Must boost higher register on each string (shorter string, less energy = less volume and sustain for higher notes)
- Active notch filter centered at the 2<sup>nd</sup> harmonic of the open note on each string

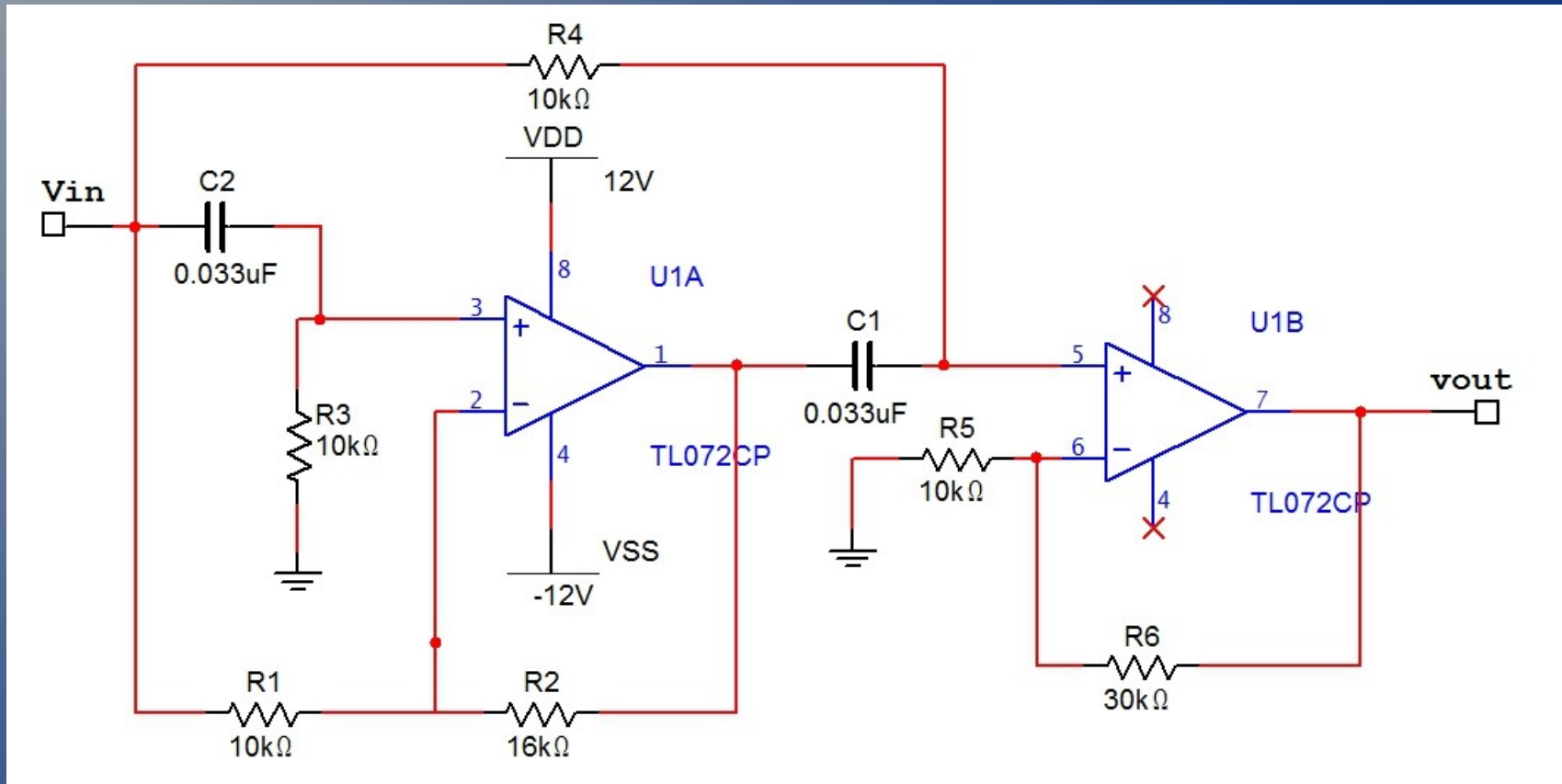
# Spectrum of lowest note on the E string (82.4Hz)



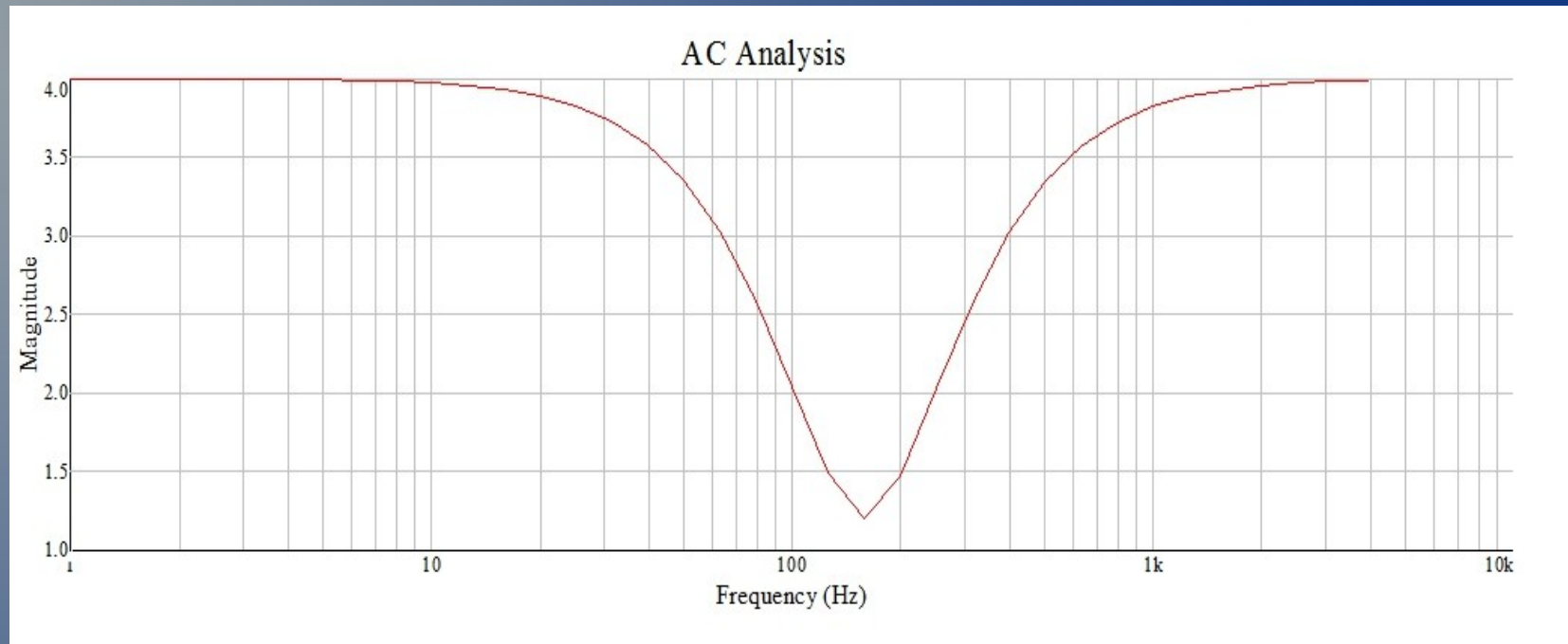
- Second harmonic is about as high as the fundamental
- Difficult to track frequency



# Design



- First op amp is the notch filter, second stage is a gain stage of about 4
- Opamp are TL074 quad op amps to save space on PCB



- Here: Low E string
- Gain of 2.5 for the lowest and highest notes on the string
- Gain of about 1.2 for the 12th fret. High enough gain for that particular note, but low enough in proportion to the lower fret's frequencies so that their second harmonics are overtaken by the fundamental.

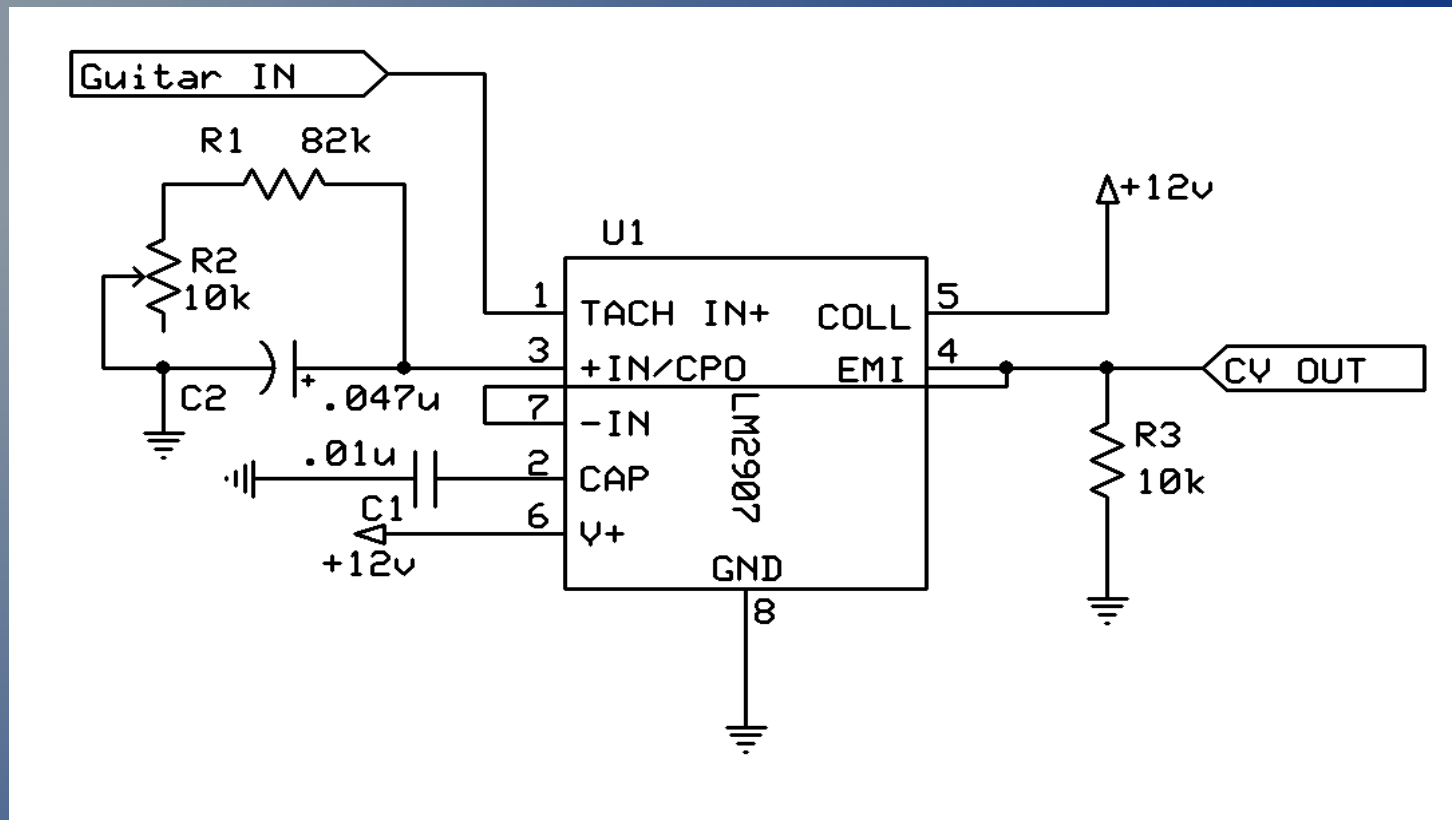
# Frequency to Voltage Converters

## Requirements

- Linearly convert frequency of the individual guitar inputs to a voltage to be “read” by the VCO
- Must be accurate to within 3% of input (it must sound relatively "in tune")
- Each string will have a full range of 0 to 5V, designed up to the 20th fret

# Design

- LM2907N frequency to voltage tachometer - 14pin dip
- Up to 28V supply range
- Designed specifically for variable reluctance magnetic pickups
- 0.3% linearity
- Easy design equation:  $V_{out} = V_{cc} * R1 * C1 * F_{in}$ ,
- Useable range is about 1.6 to 5V from open to 20<sup>th</sup> fret



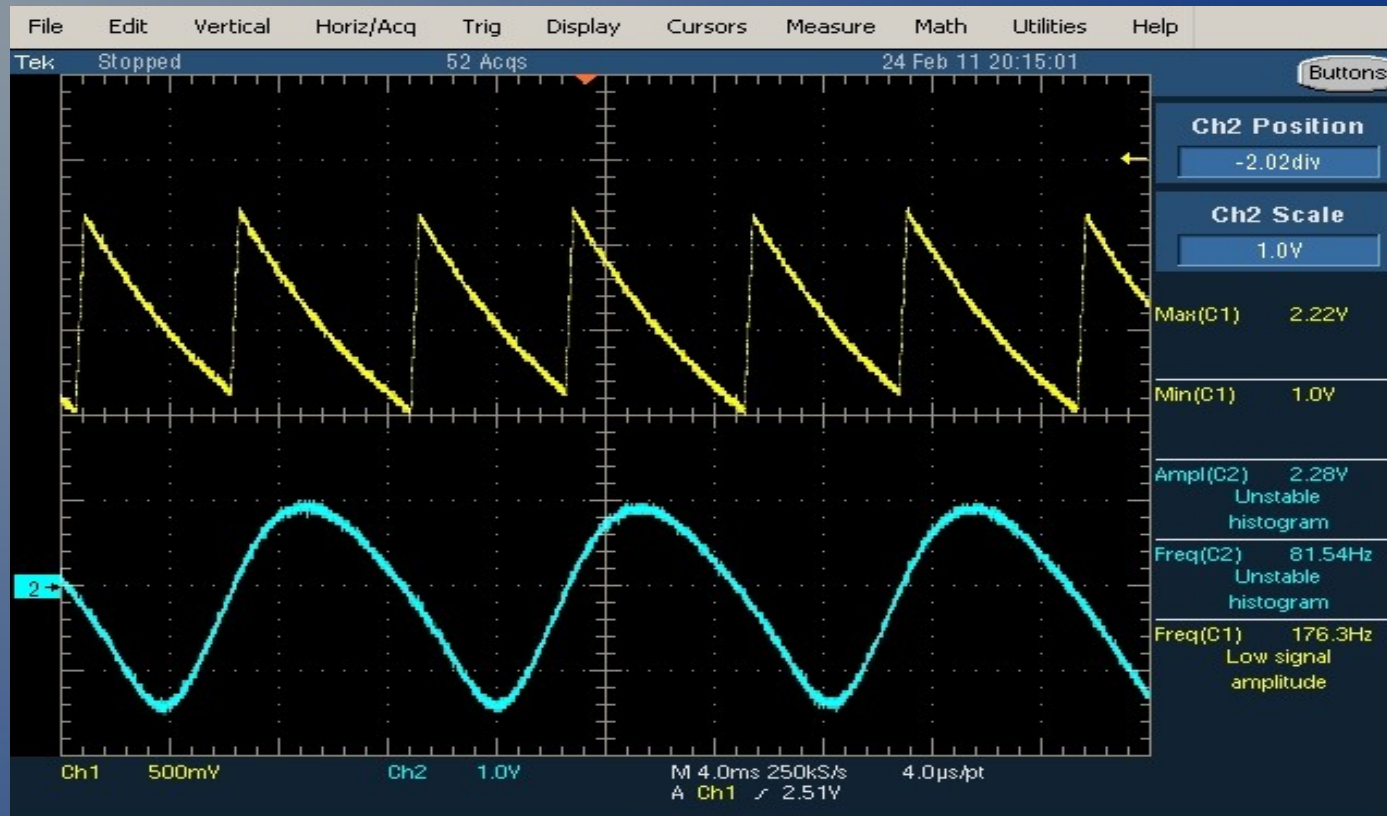
- C2 was selected to balance between quick response time and low ripple
- -the settling time is around 40ms. this might need to be shortened if the delay is too noticeable
- -even after "settling" there is still a ripple with twice the frequency of the input

## Table of values for each string

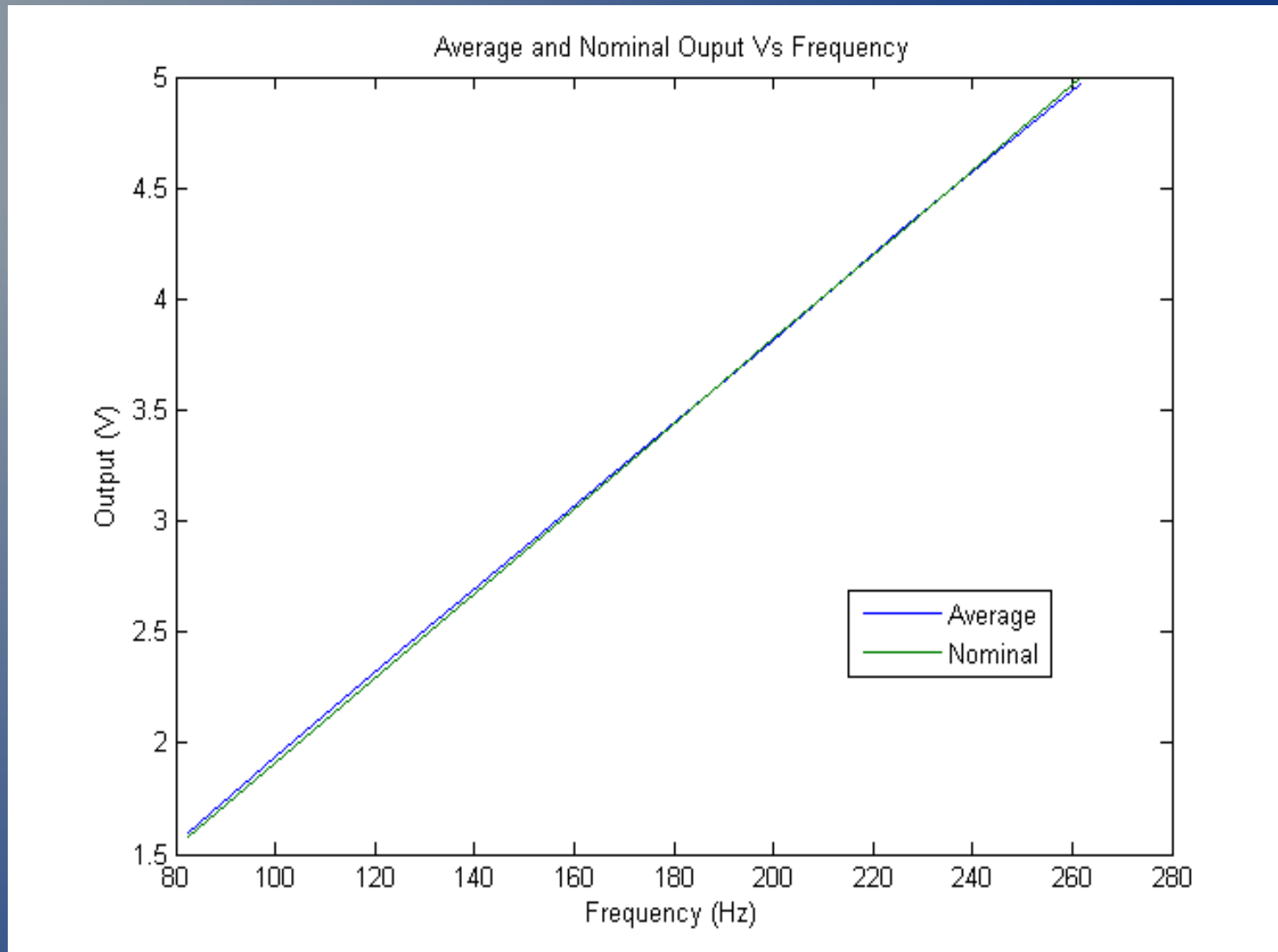
Open	20th fret	Slope	R desired	R + 100k trimpot	C
82.41	261.63	0.0191	160k	100k	0.01uF
110	349.23	0.0144	120k	82k	0.01uF
146.83	466.16	0.0107	160k	100k	0.0056uF
196	622.25	0.0080	120k	82k	0.0056uF
246.94	783.99	0.0064	240k	180k	0.0022uF
329.63	1046.5	0.0048	180k	120k	0.0022uF

- Real-world component values selected
- Trim pot will be used to "tune" the circuit during testing

# FVC Signal From Pickup



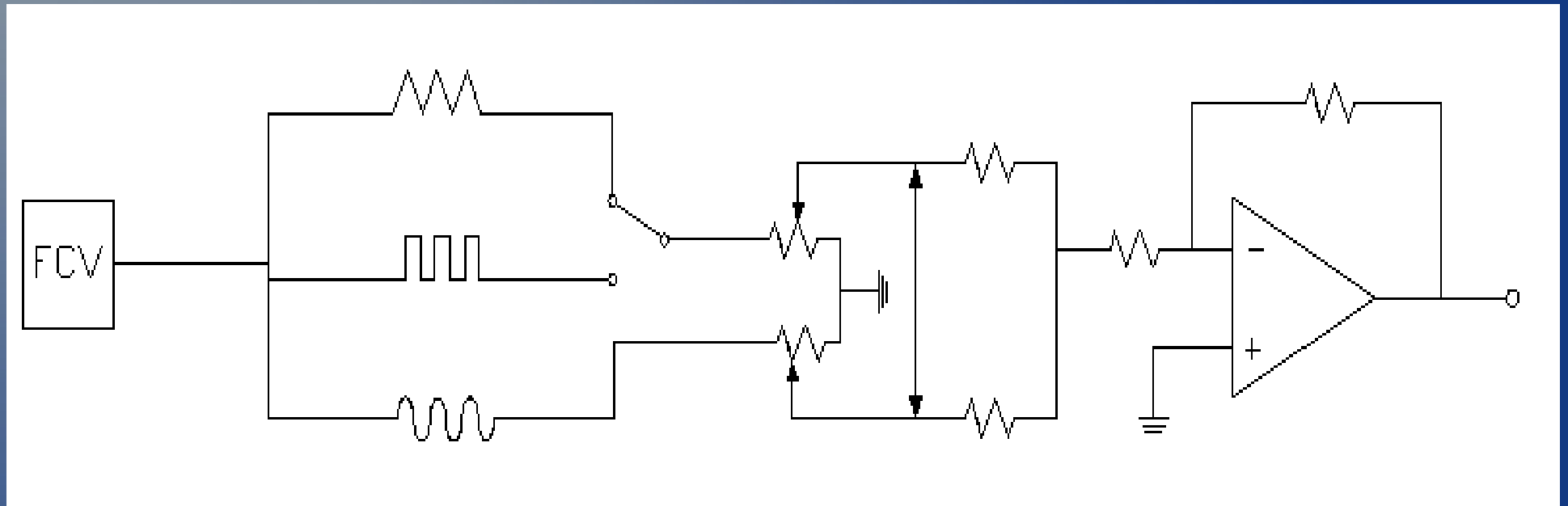
- Output ripple – would cause frequency variance problems of the VCO
- To be solved by the sample and hold block



- 1.65% error on the low end, 0.05% error on high end
- Sensitivity to error decreases as frequency increases
- Error on low end is more critical

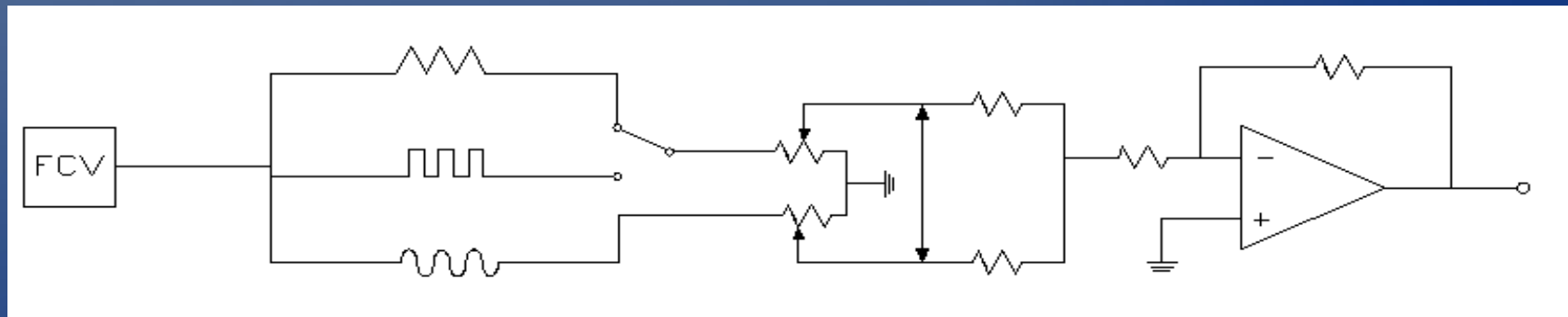


# Overview

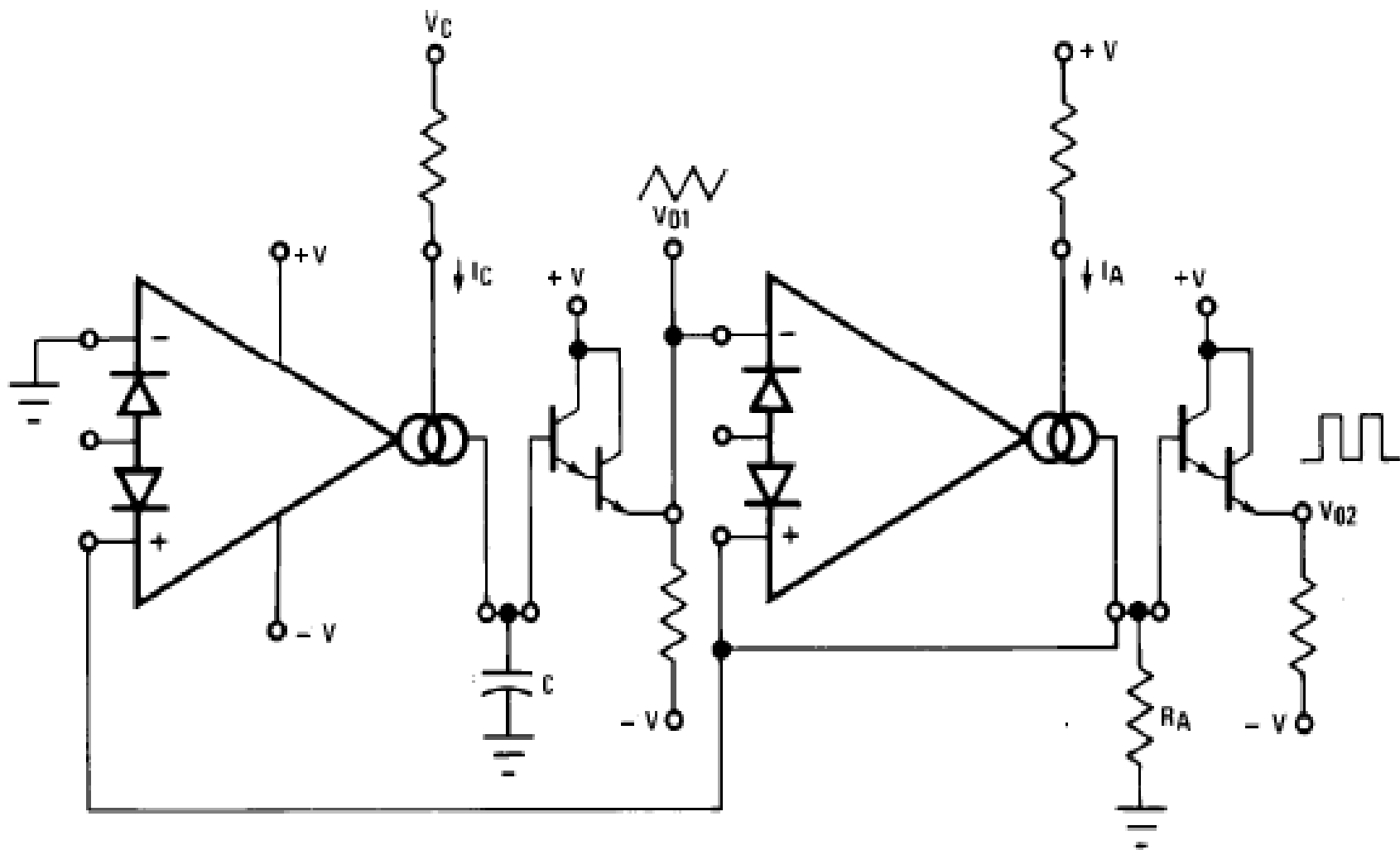


# Wave Shaping Requirements

- Generate three waveforms of equal frequency, amplitude and peak-amplitude phase (Sine, Square, Triangle)
- Exclusively use analog components



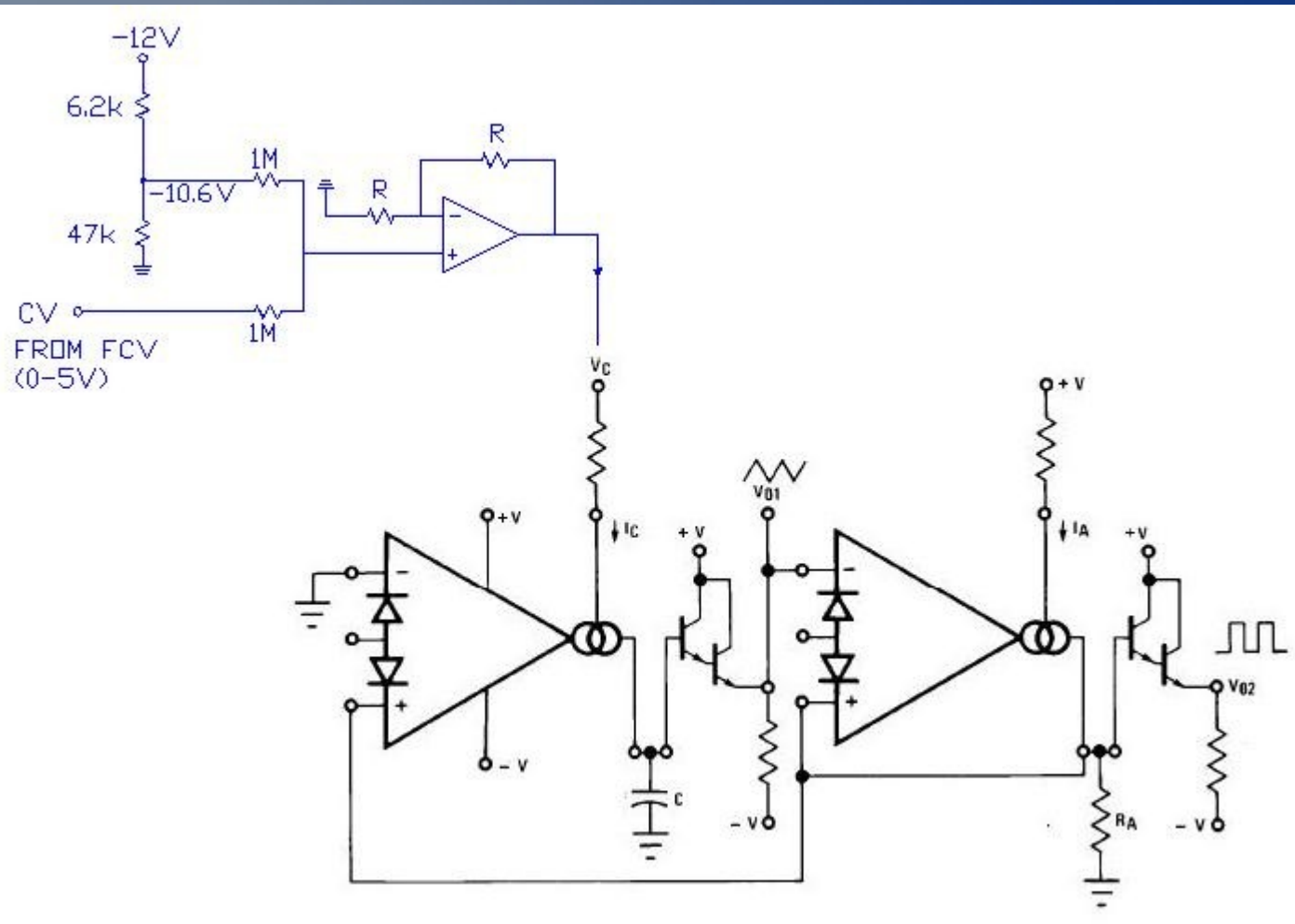
# LM13700



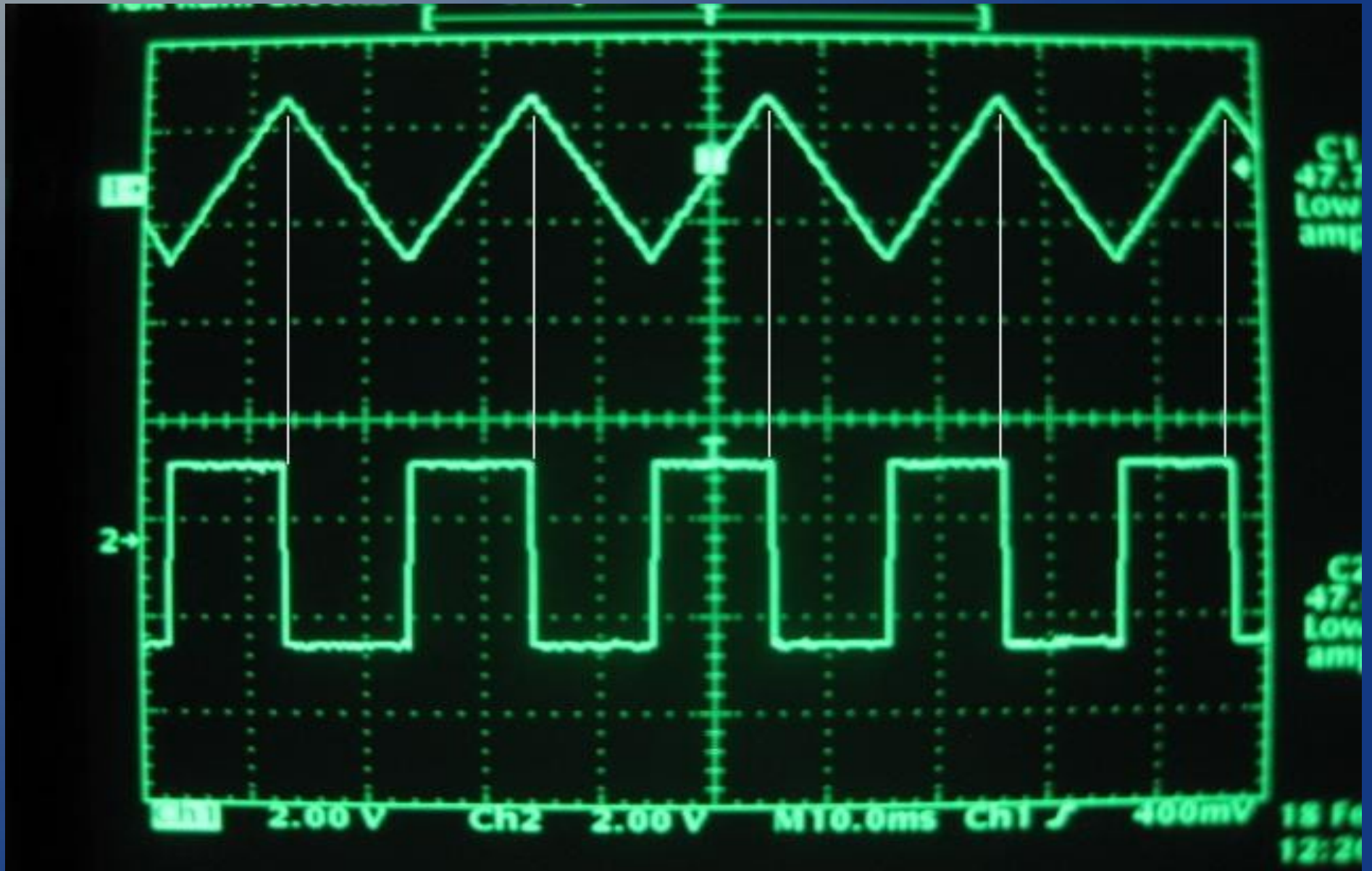
# VCO Design Approach

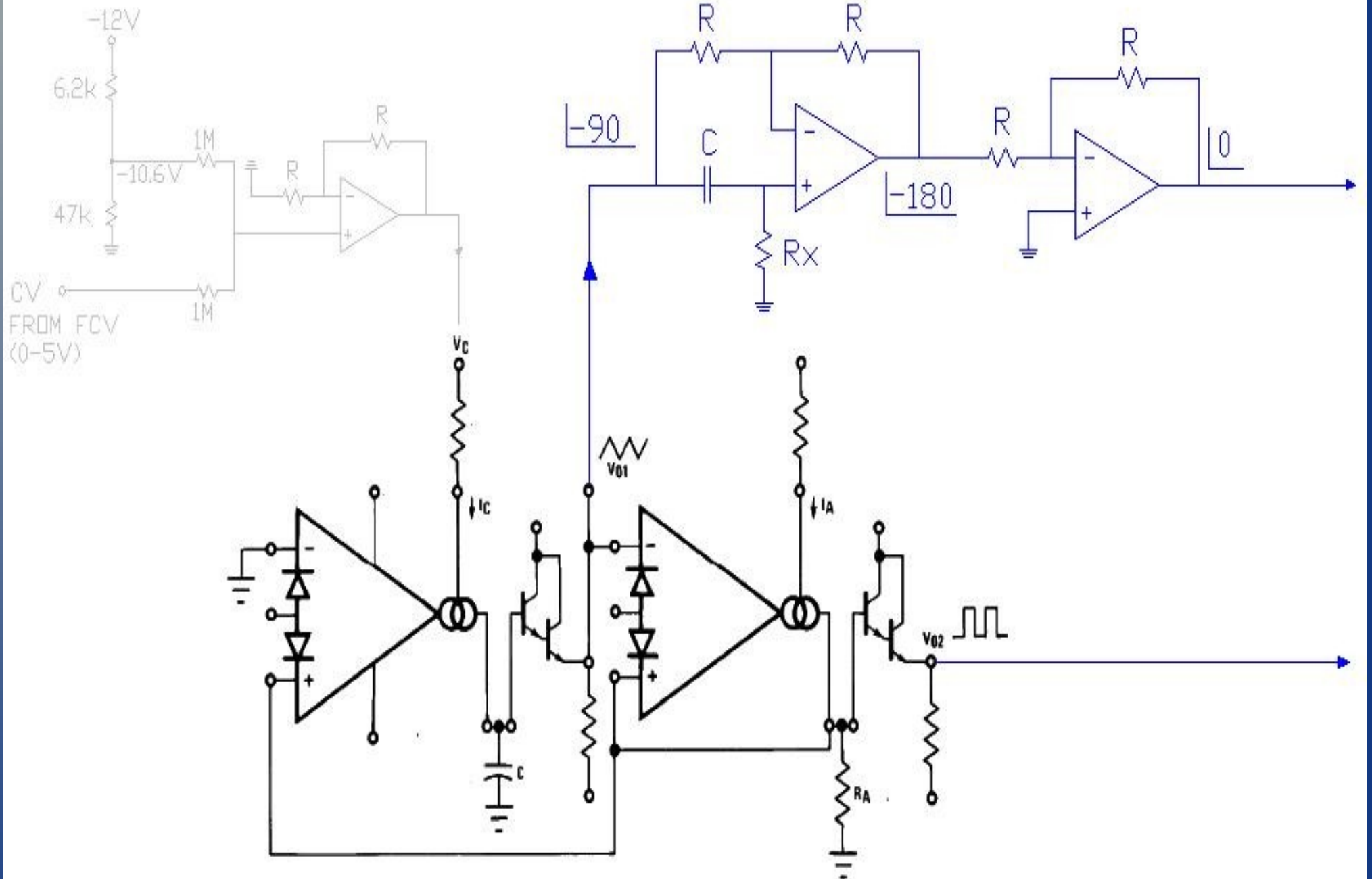
$$f_{\text{OSC}} = \frac{I_C}{4C I_A R_A}$$

$$I_C = \frac{V_C - 1.4 - V_-}{R} \rightarrow I_C = \frac{V_C + 10.6}{R}$$



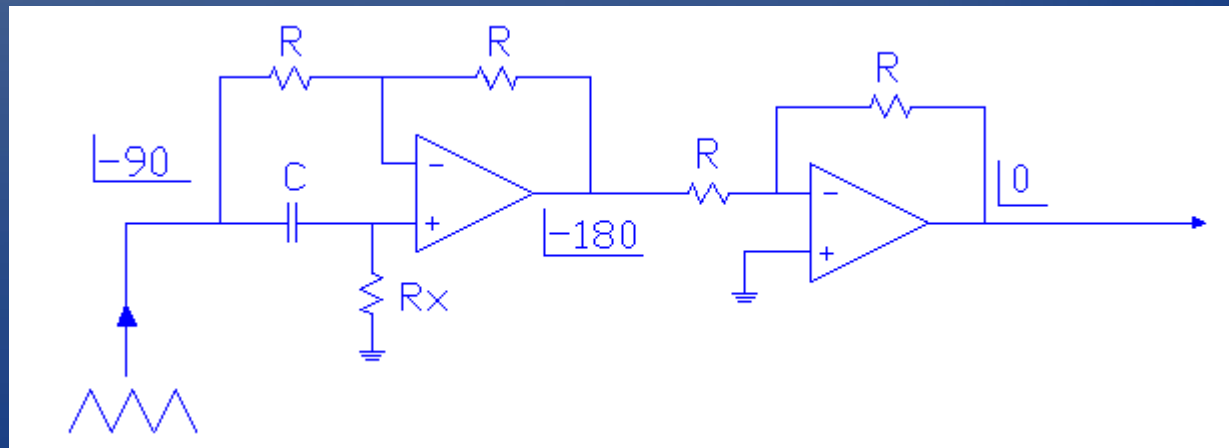
# Testing & Problems



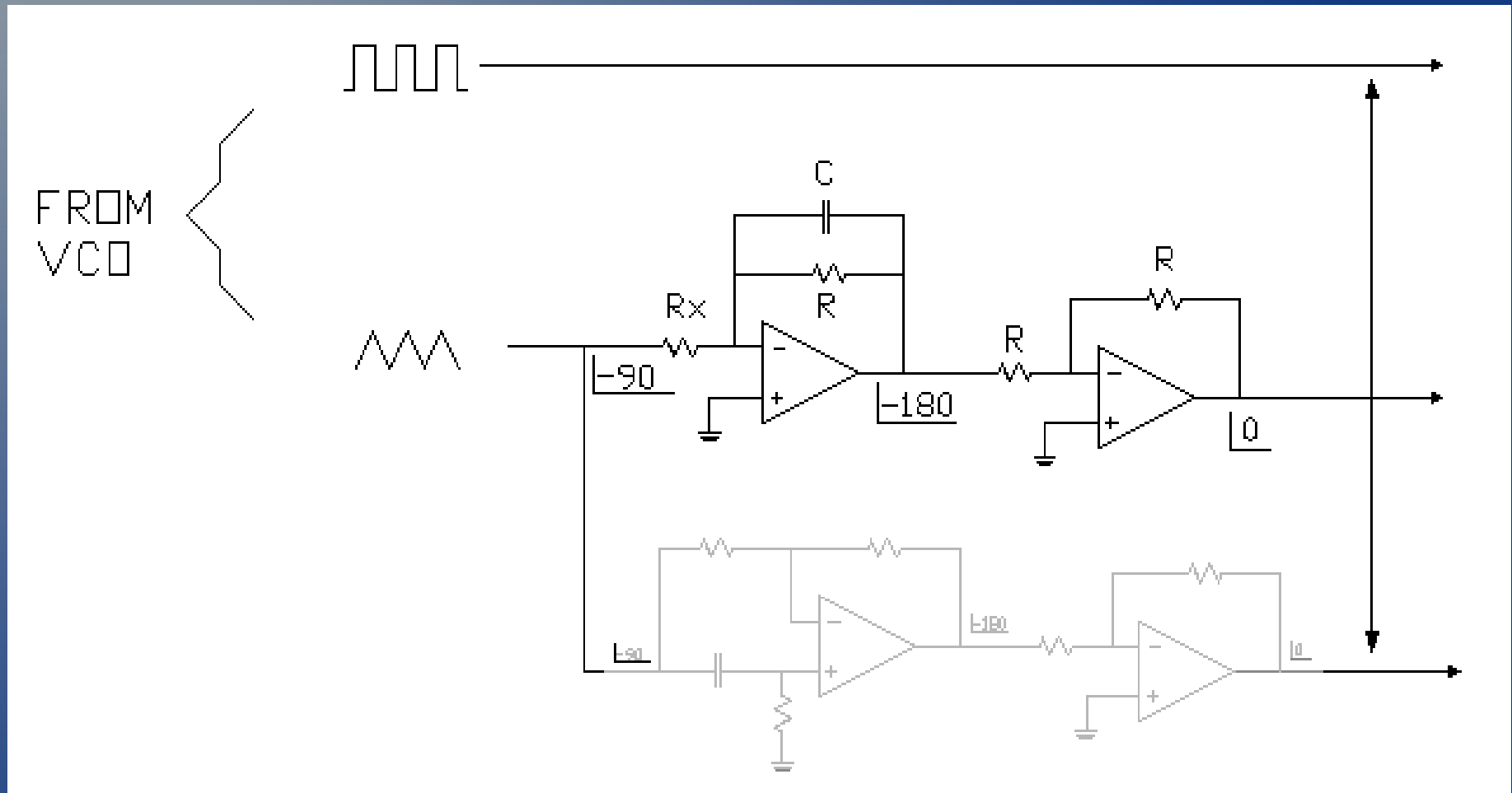


# All Pass Filter Design

- What to consider?
  - Phase variation at different frequencies
- $\omega = 1/RC$
- Use average frequency of each guitar string when choosing  $R_x$  and  $C$
- Use equal values of  $R$  for both filter and inverter to maintain unity gain (and simplicity)



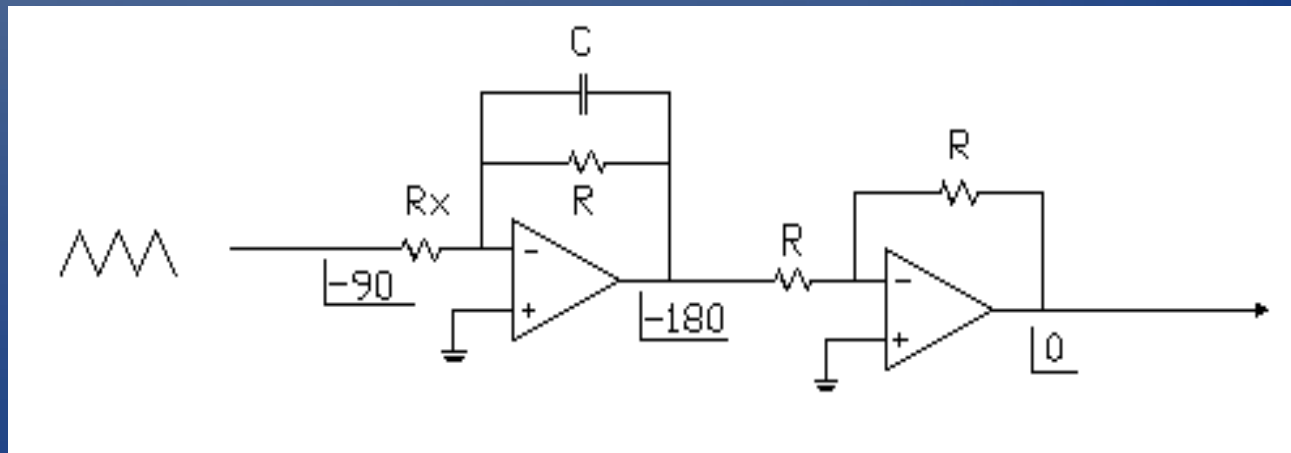
# Sine Wave Generator





# Integrator Design Approach

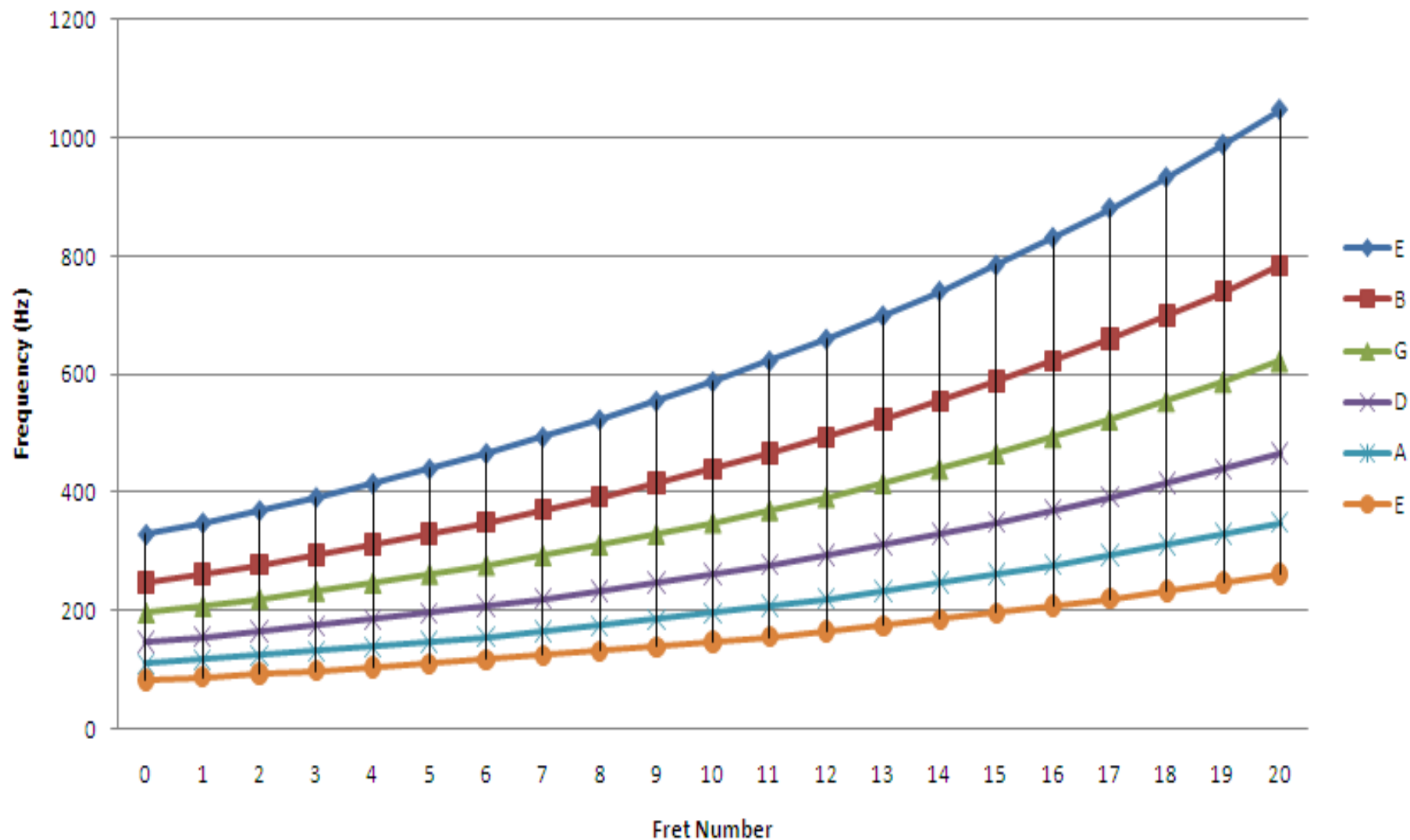
- Amplitude variation at different frequencies
- Design  $R_x$  and  $C$  based on average frequency
- $\omega = 1/R_x C$
- Match resistors  $R$  for unity gain



## Parts Required

- Parts required
  - 1 LM13700 x6
  - 1 LF351N x 6
  - 2 Dual Op-amp x 6
  - 22 Resistors x6
  - 1 Capacitor x6

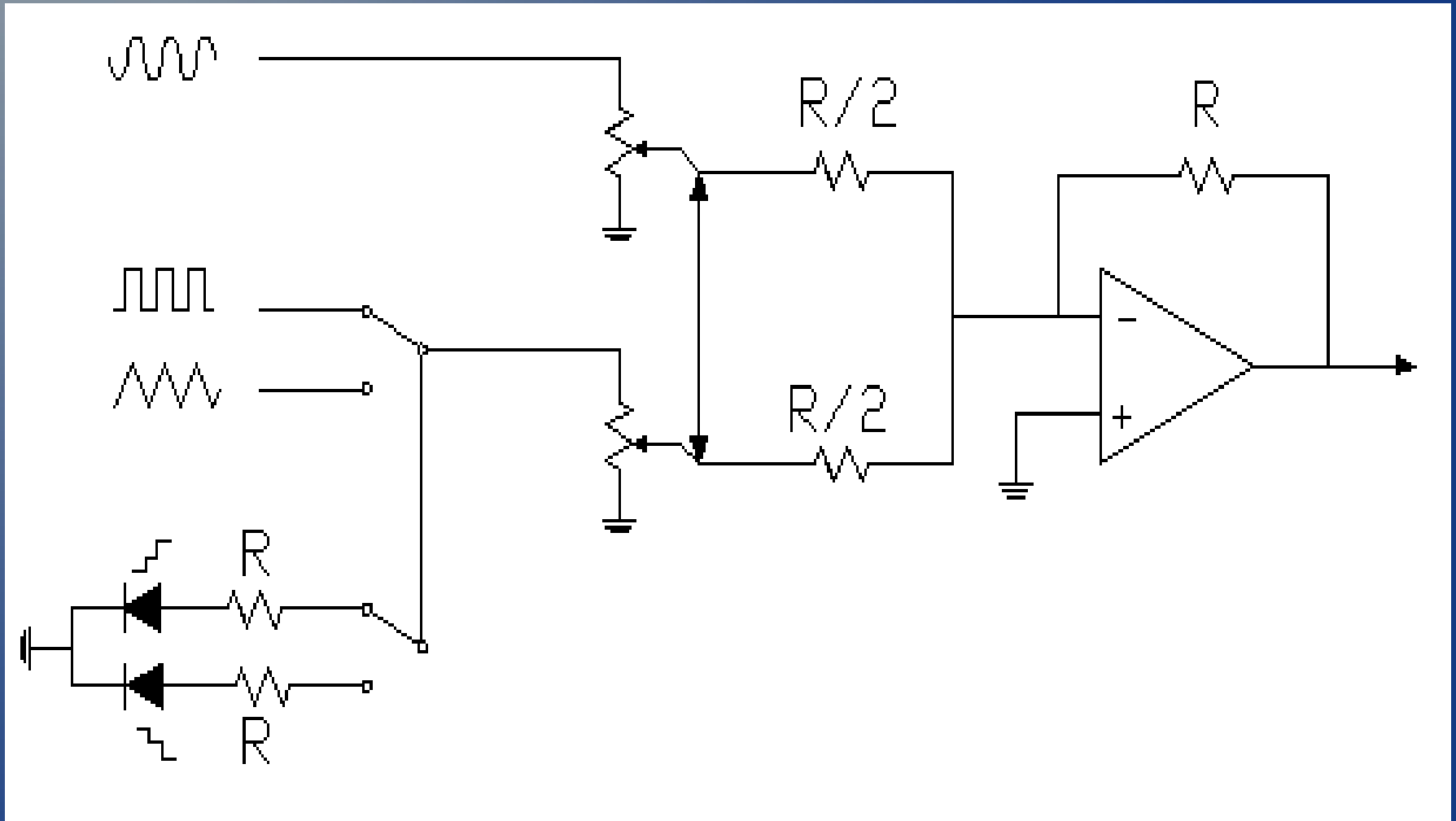
# Improvements



# Mixer Requirements

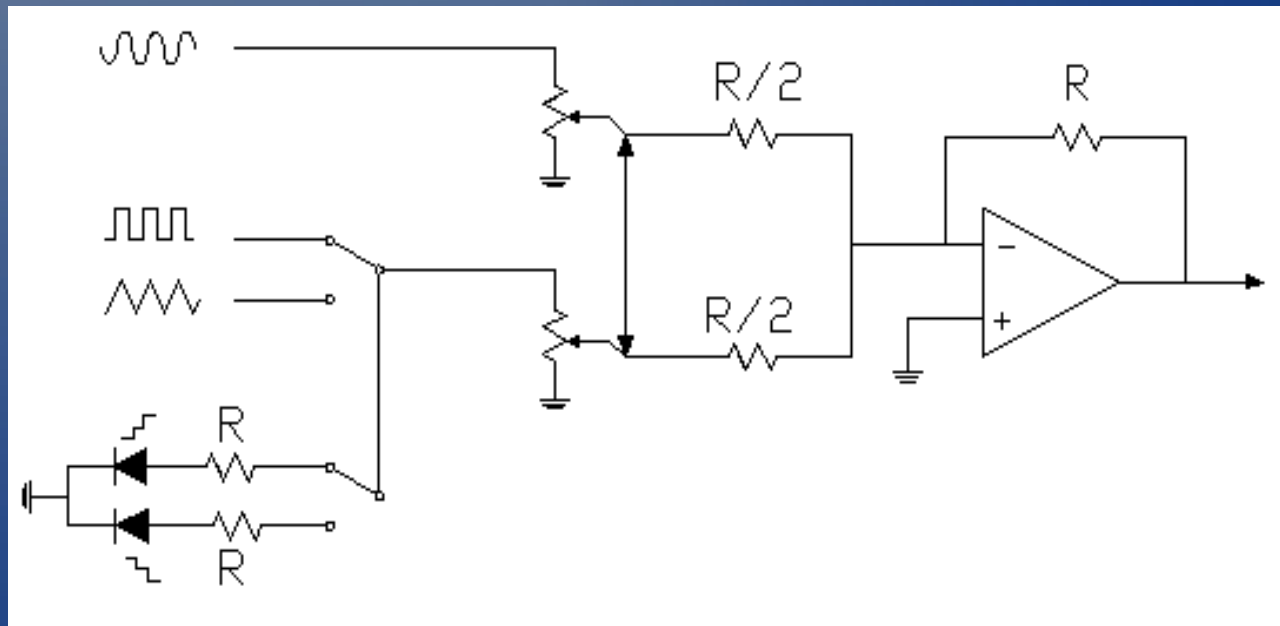
- Add two waveforms while maintaining unity gain
- Allow user to select between two waveforms to mix
  - Mix Triangle and Sine or
  - Mix Square and Sine
- LED's to indicate which waveform has been selected
- Allow proportion control between the two signals
- Mix all six resulting signals to send forward to filtering & modulation stages

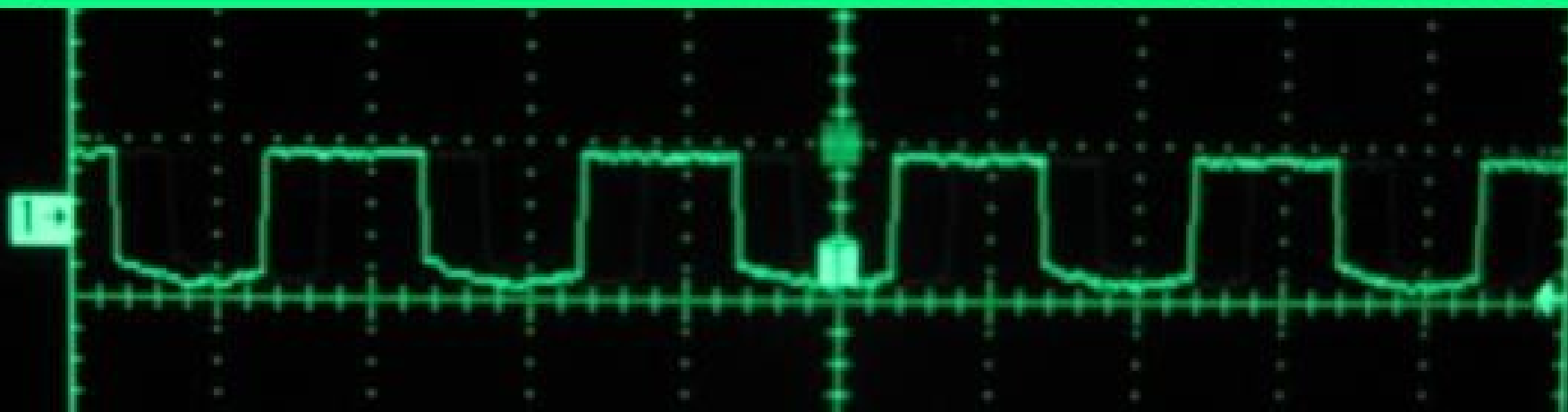
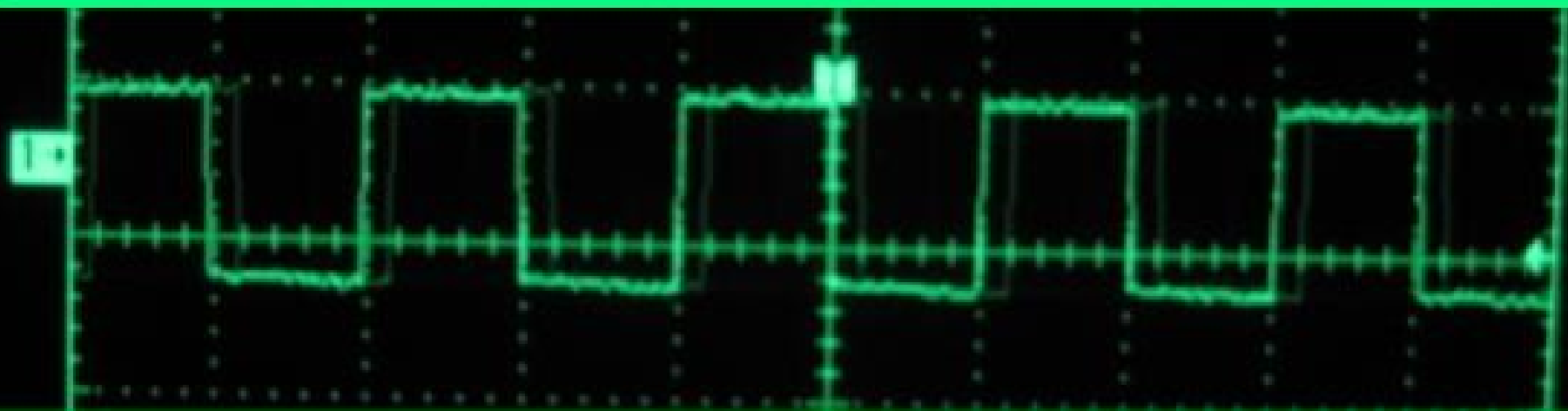
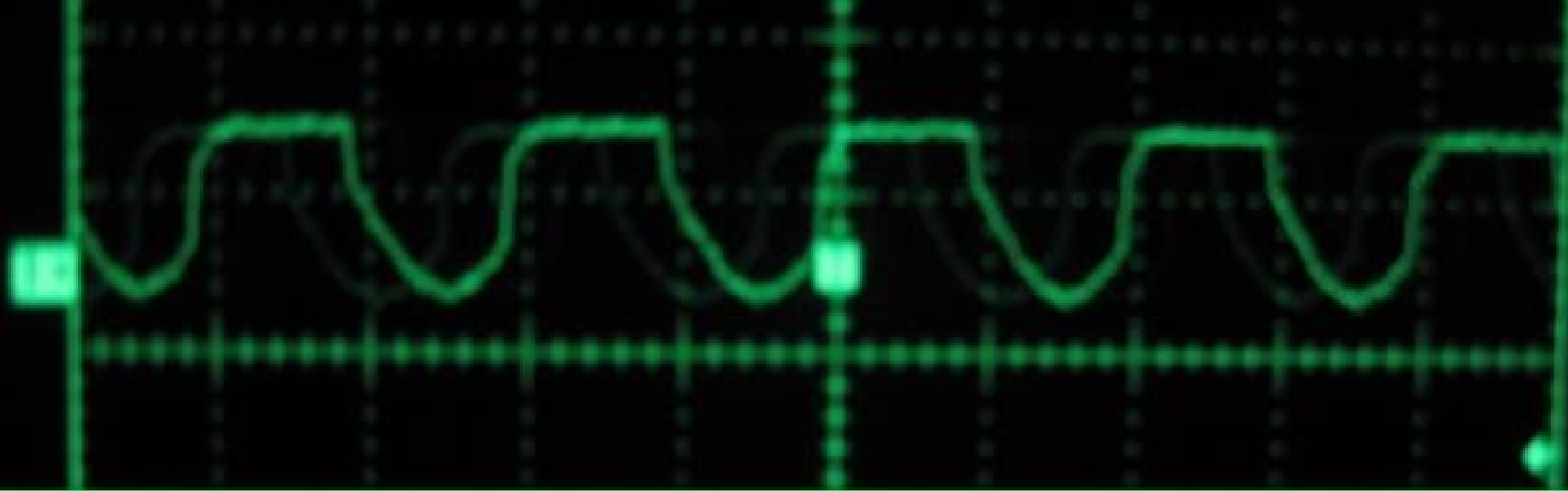
# Independent Mixing Stage

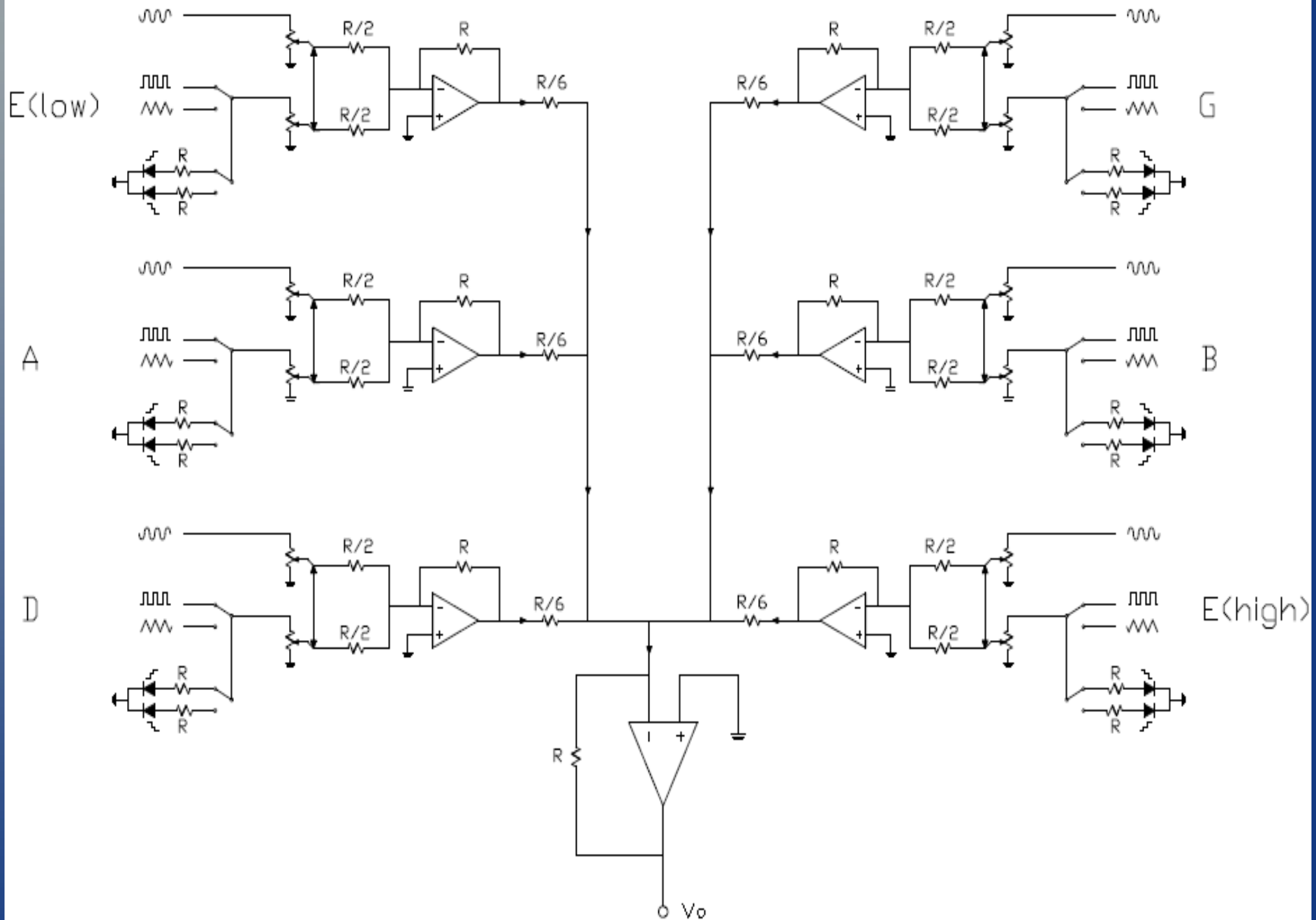


# Design Considerations

- R values to achieve unity gain
- Dual-gang potentiometer for proportion control
  - ~20% error is expected for sliding potentiometer









# Parts Required

- 1 op-amp (LF351) x7
- 5 Resistors x6
- 2 LED's x6
- 1 DPDT Toggle Switch x6
- 2 Sliding Potentiometers x6

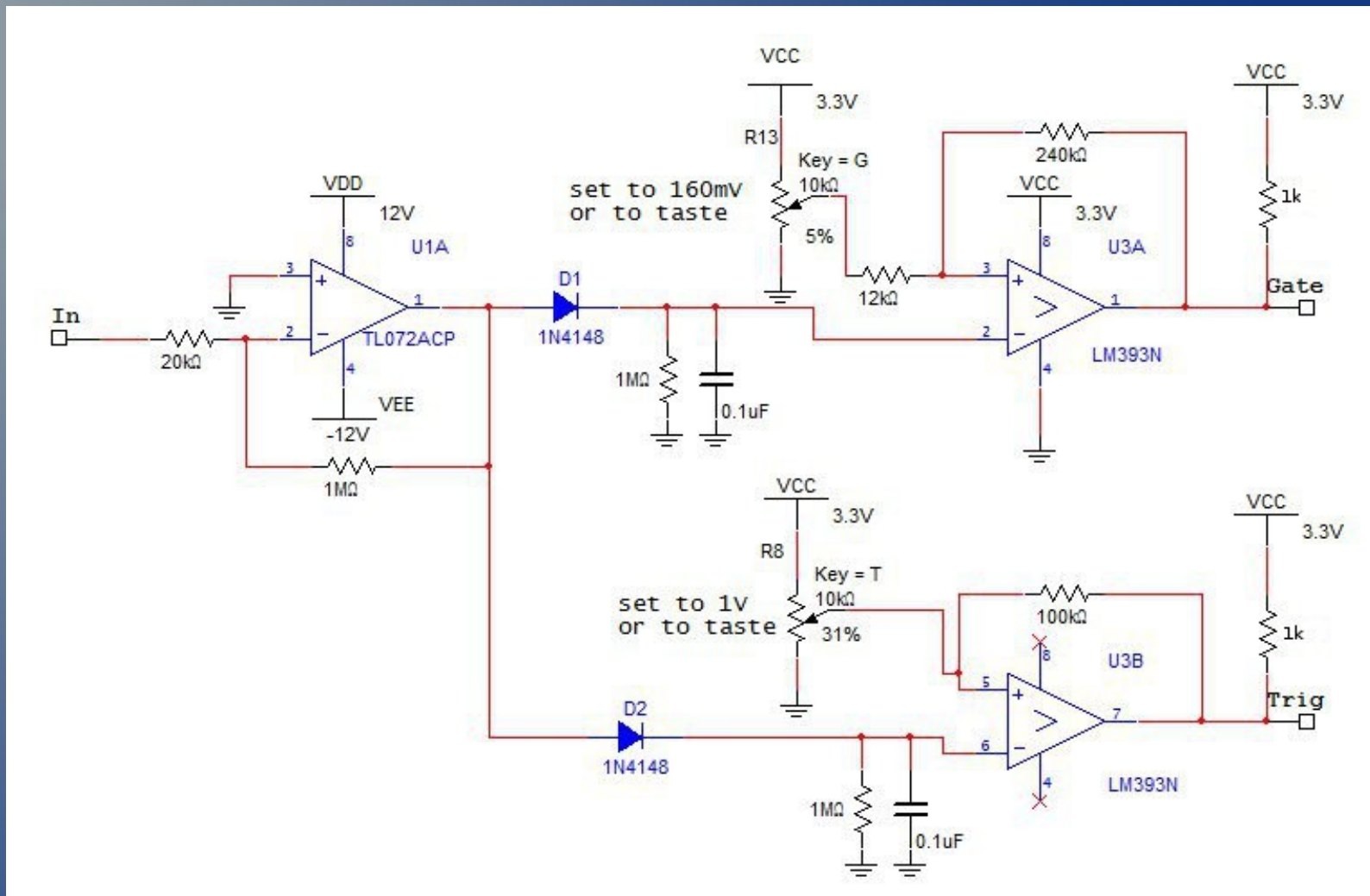
# Triggers and Gates

## Requirements

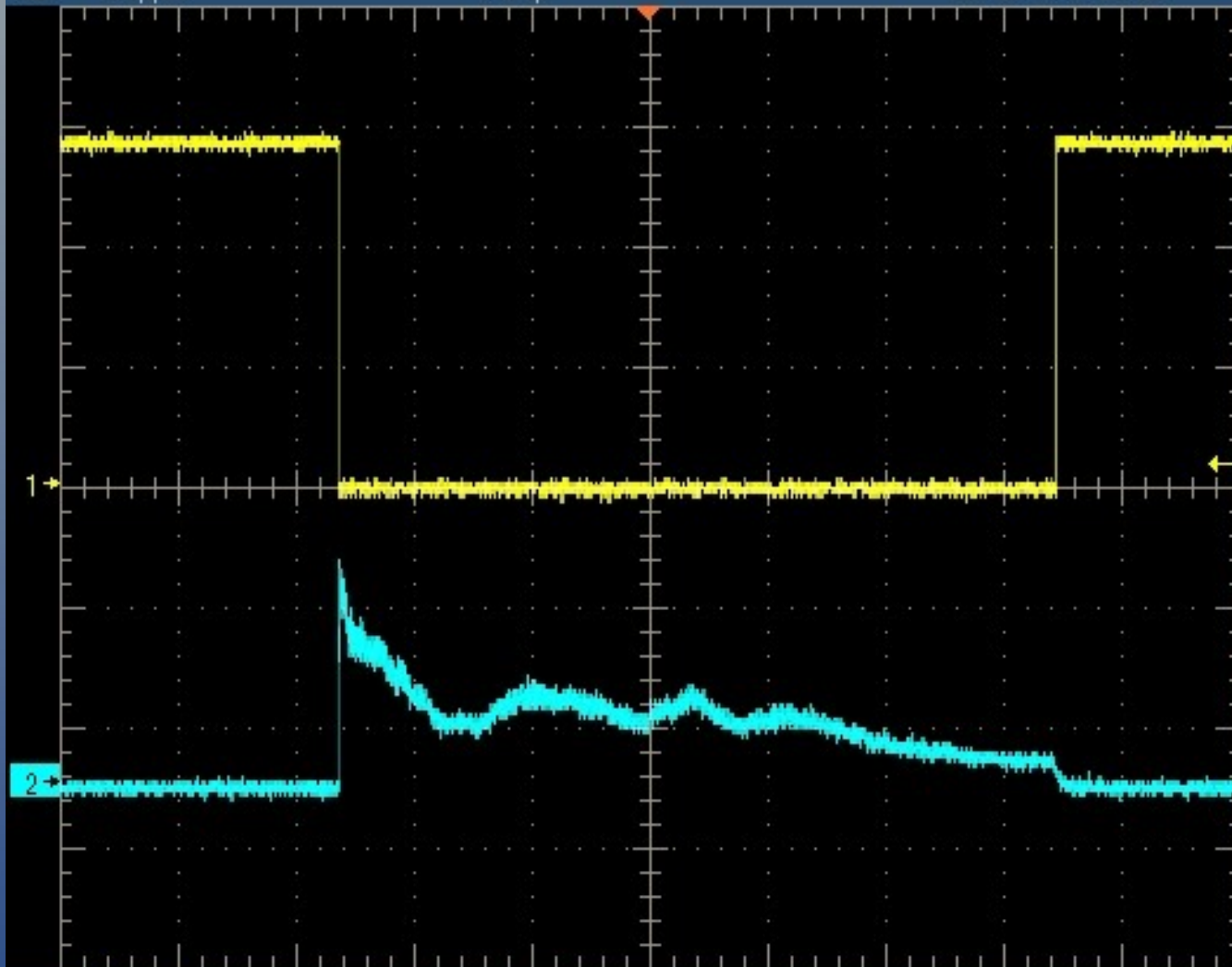
- The trigger is a binary pulse that is activated whenever a new note is hit on the guitar string
- Tells the sample/hold and ADSR blocks that a new note happened
- The gate is also a binary signal that is active whenever the pickups are sending signals
- Tells the sample/hold and ADSR blocks that a note is still being played

# Design

- Simple envelope detector followed by two comparators - one for gate and one for trigger
- Inverting comparator using LM339 quad comparator
- Gate will have a voltage reference of about 150mV with hysteresis of about 100mV
- Trigger will be less sensitive, so that it is only high when a guitar string is plucked – about 1V vref
- The natural envelope of a plucked string falls sharply after the initial attack, causing the trigger to be a short pulse



- The trigger is similar to the gate, but with a higher voltage reference of about 1V
- The output will switch from 3.3V to 0V



Max(C1) 2.92V

Min(C1) -160.0mV

Ampl(C2) 1.2V  
Unstable  
histogram

Freq(C2) 1.145Hz  
Low signal  
amplitude

Freq(C1) -Hz  
Zero  
period

Ch1 1.0V

Ch2 1.0V

M 1.0s 500S/s 2.0ms/pt

A Ch1 / 160mV

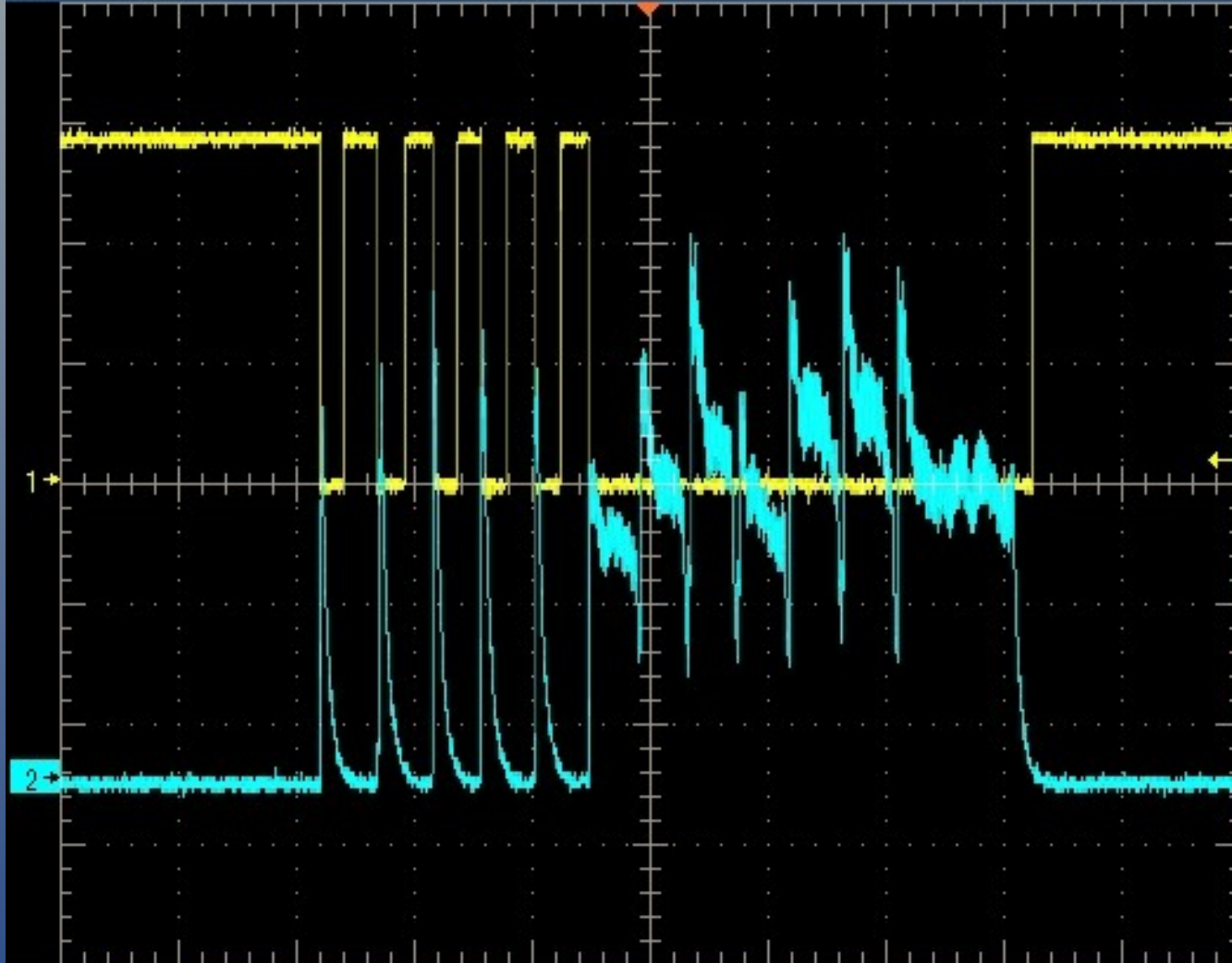
File Edit Vertical Horiz/Acq Trig Display Cursors Measure Math Utilities Help

Tek Stopped

1 Acqs

24 Feb 11 17:46:09

Buttons



Max(C1) 2.92V

Min(C1) -200.0mV

Ampl(C2) 2.44V

Unstable  
histogram

Freq(C2) 2.039Hz

Unstable  
histogram

Freq(C1) 2.049Hz

Ch1 1.0V

Ch2 1.0V

M 1.0s 500S/s

2.0ms/pt

A Ch1 / 160mV

# Sample and Hold Requirements

- Smooth out ripple from the frequency to voltage converter
- Samples the FVC signal, averages the values, outputs a smooth signal using a filtered PWM
- Allows the user to select between two modes - continuous and hold
- Continuous - continuously samples the FVC output while the gate is active
- Hold - activated via a foot switch, holds the last note(s) sampled. Allows some strings to be held while the rest are in continuous mode

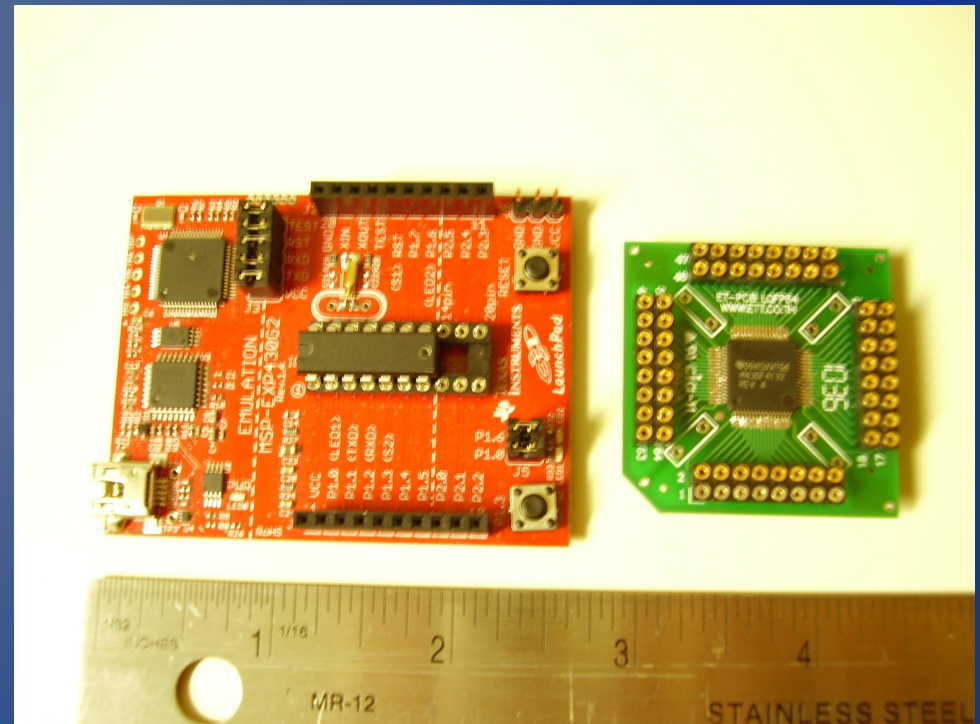
# Microcontroller

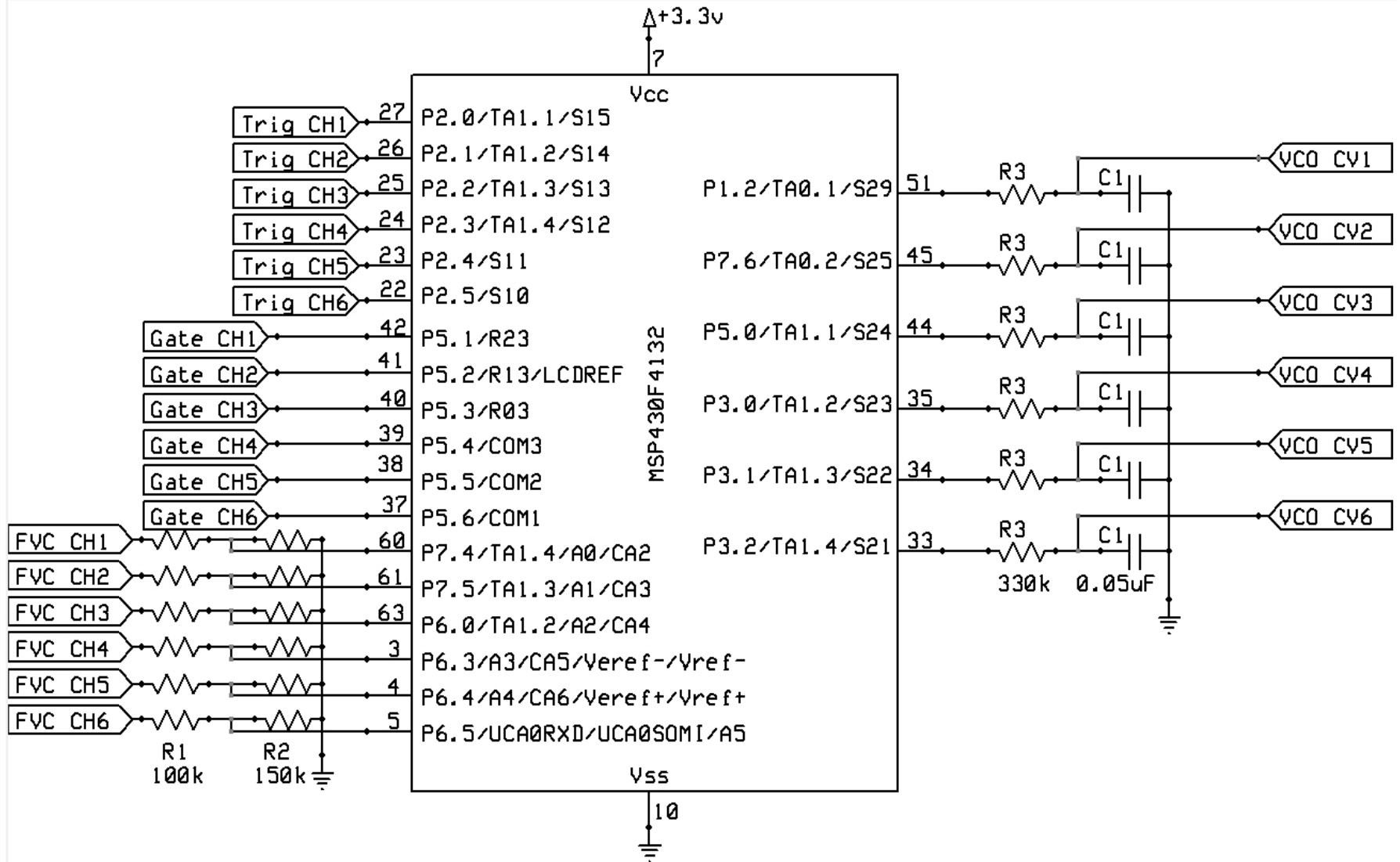
- MSP430F4132 microcontroller
- 8KB Flash, 512B SRAM, 8MHz
- 56 GPIO, 64 pin LQFP package
- Has multiple port pin interrupts
- 8 10-bit ADC channels
- Outputs up to 6 PWM signals from capture/compare peripheral
- Will accept the trigger outputs as port pin interrupts
- Output will be filtered PWM signals sent to the VCO



# MSP430 Development Board

- MSP430 Launchpad
- Cheap (\$4.30), available, programmable in C using CCS compiler
- 64 pin breakout board for the MSP430F4132 (for testing only)
- Programmed by the Launchpad via Spy-bi-wire interface



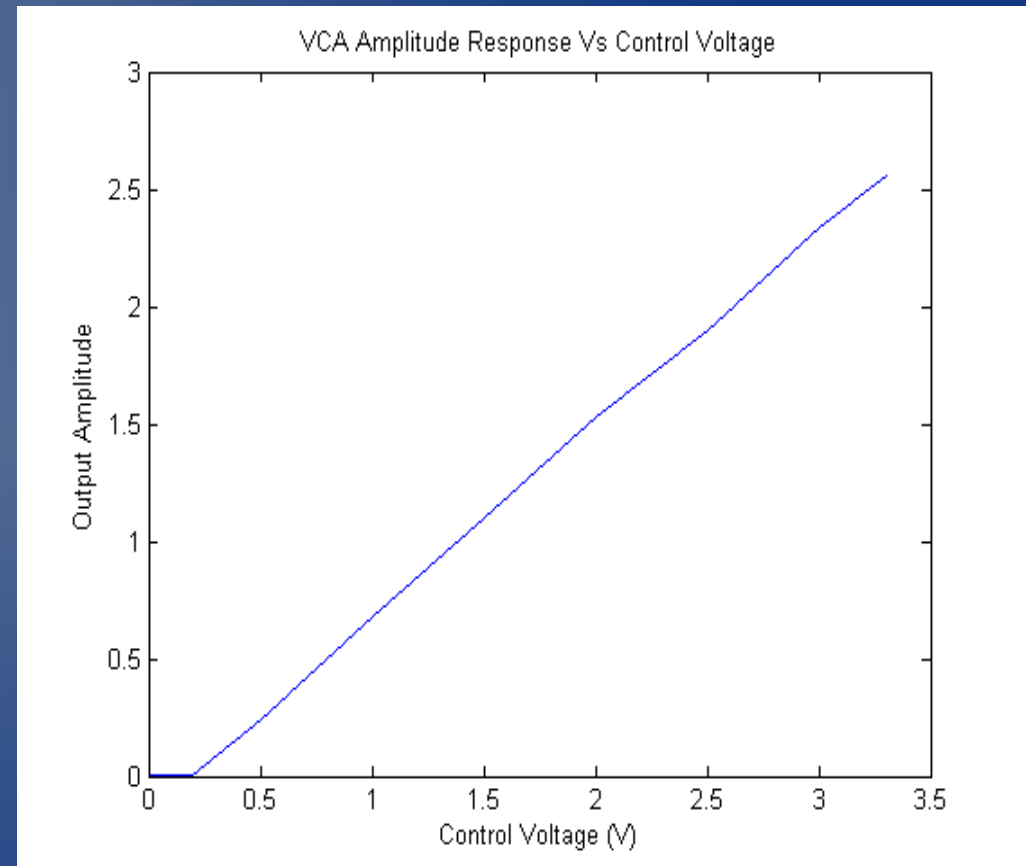


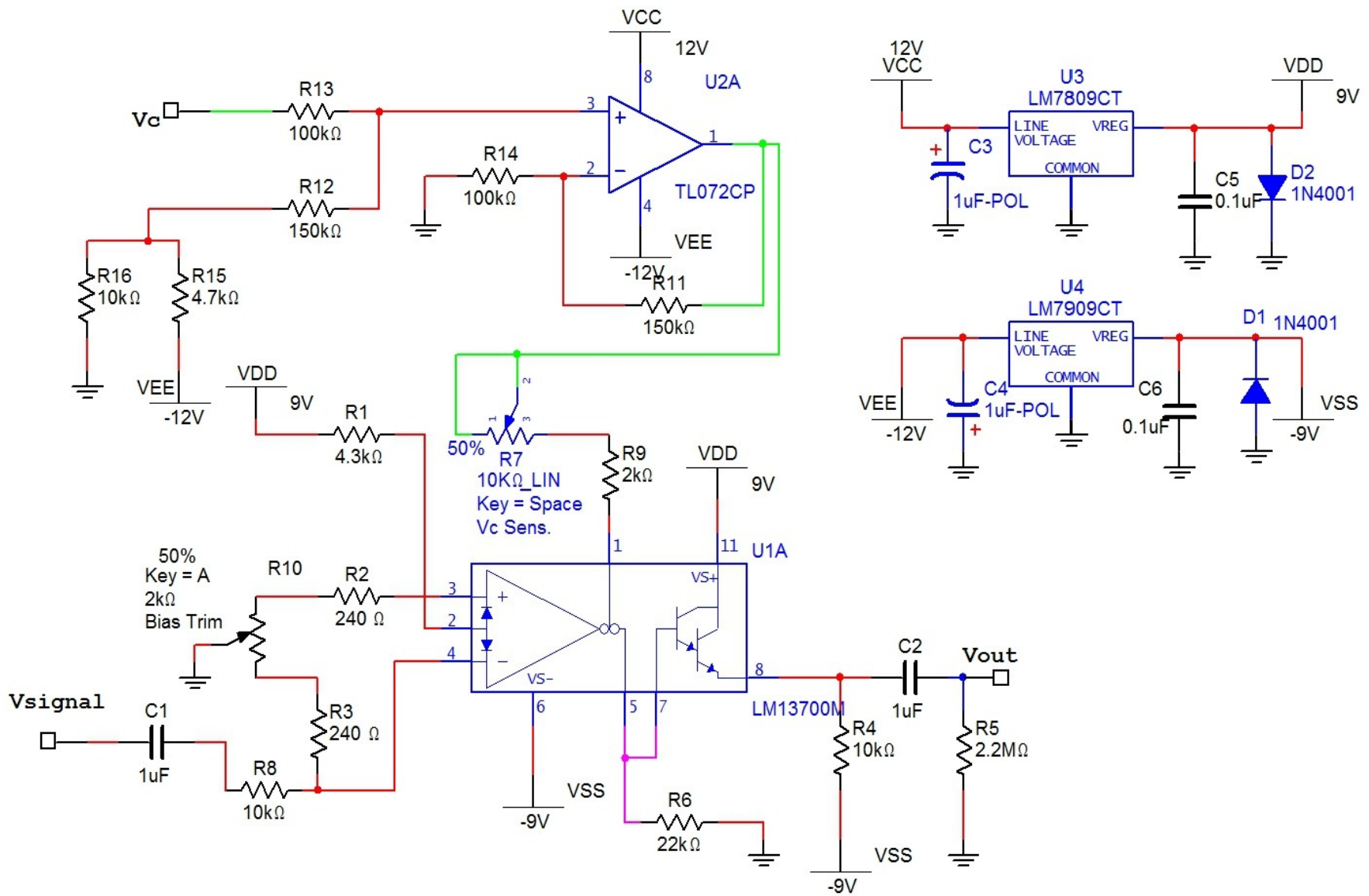
# Voltage Controlled Amplifier (VCA) Requirements

- To provide amplification that varies with a control voltage created by a contour generator
- Must be linear with respect to control voltage
- Have a flat frequency response of +/-1dB up to 20kHz
- Must produce undistorted output (no rail clipping)
- Minimum of unity gain
- Input offset voltage less than 1mV

# Design

- VCA using the LM13700
- Input offset voltage of 0.4mV
- Control voltage is from the ADSR contour generator
- Control voltage will be amplified from 3.3V to about 5V and shifted down by about 8V from by an opamp



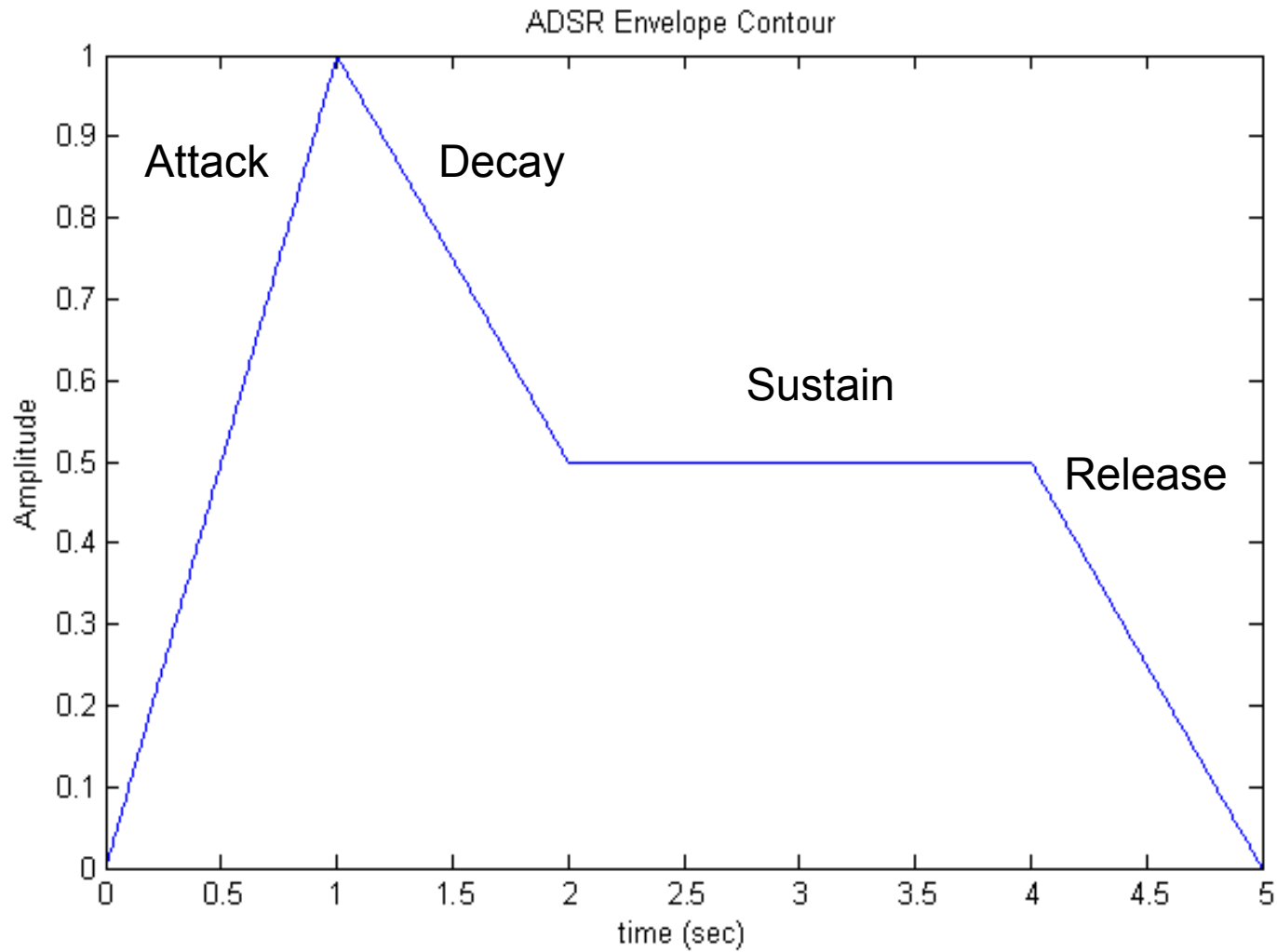


# Envelope Generator

ADSR – Attack, Decay, Sustain, Release

- Provide a user controlled contour generator
- Wide range of attack/decay/release times - 50ms to 5s
- Will provide user with numerous contour options from short, staccato to slow volume swells
- External controls are
  - Attack time – initial volume rise time
  - Decay time – time to fall to sustain level
  - Sustain level
  - Release time – time to fall to zero volume after note is release (string muted)

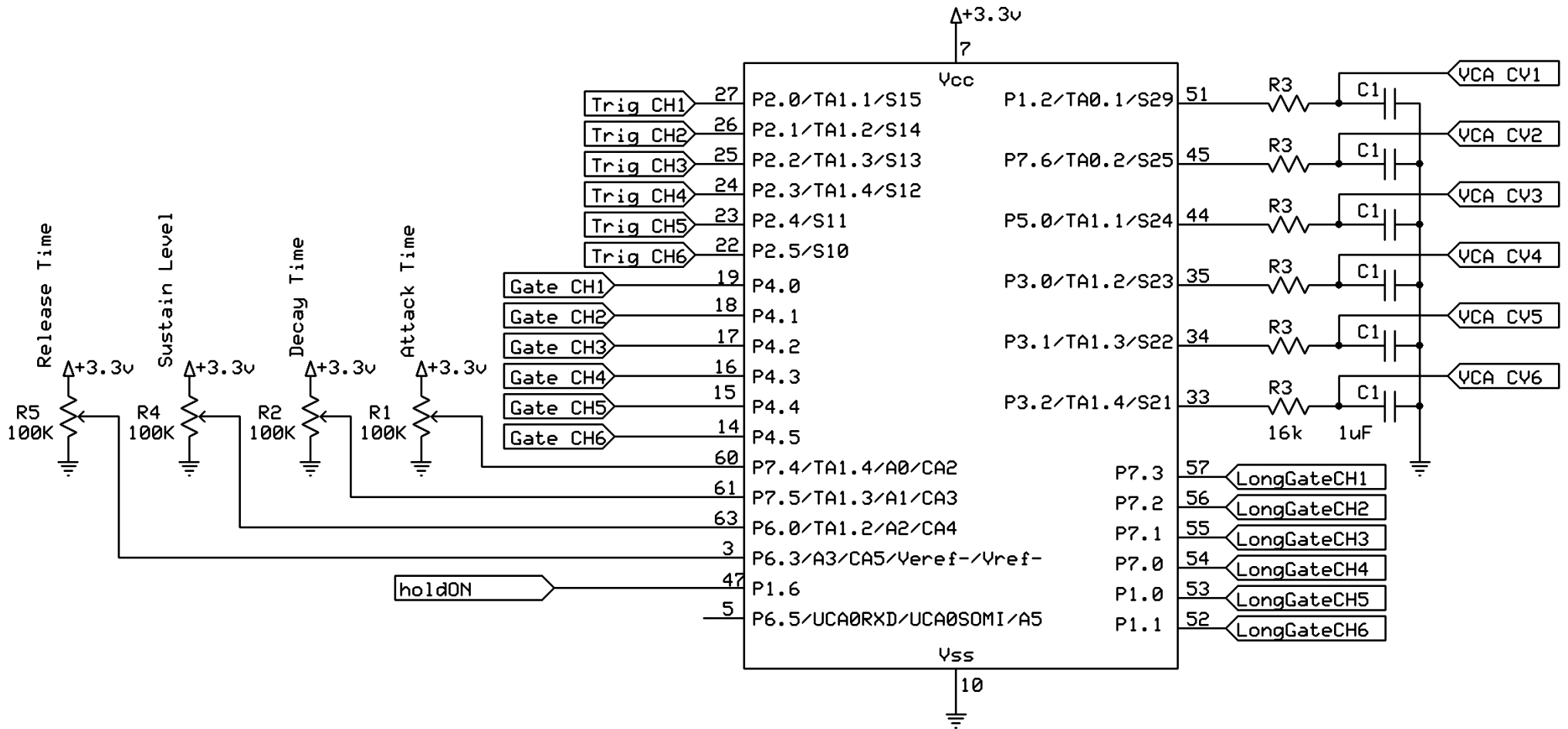
# General ADSR Shape



# Design

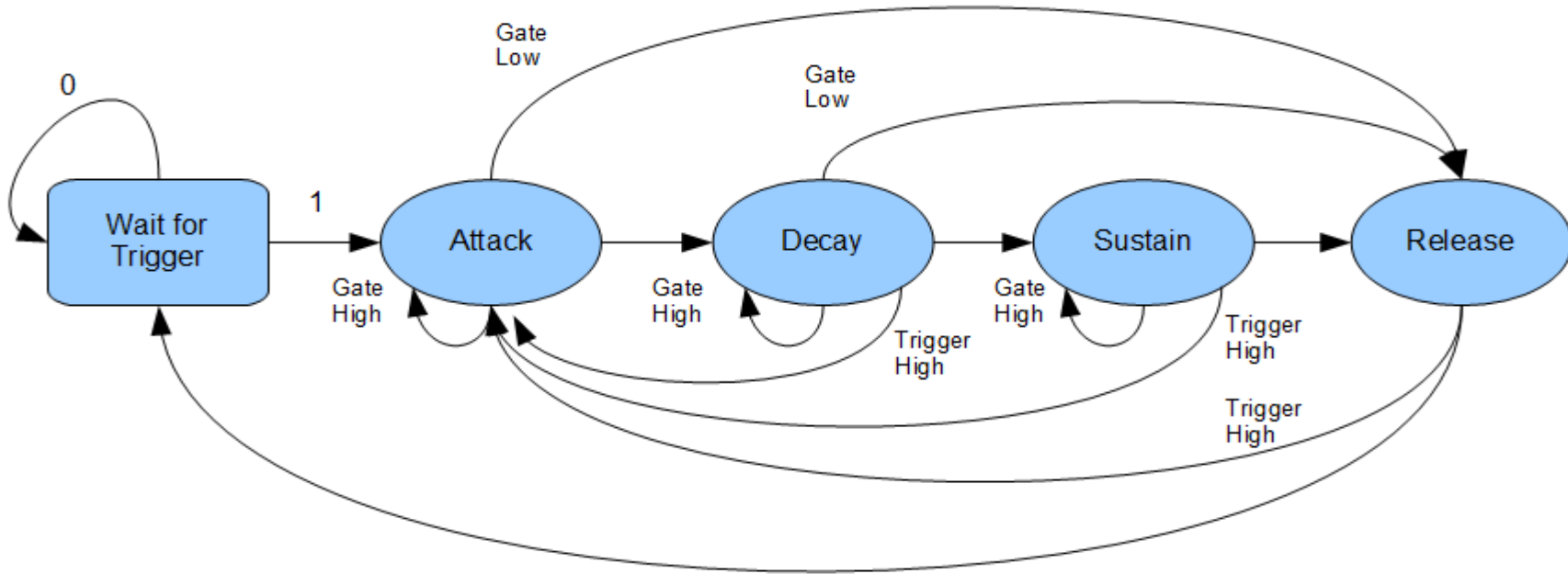
- MSP430F4132
- 6 PWM outputs, one for each channel's contour, 15.6kHz
- Will use 4 ADC pins for reading potentiometer positions
- ADC pins sampled every 32ms
- Will accept triggers and gates for each channel
- Utilizes wave tables for rising and falling exponential curves, simulating analog synthesizer ADSRs
- Also, the ear hears logarithmically, so an exponential shape in volume sounds more linear
- Time is sped up by skipping samples (principle of direct digital synthesis)

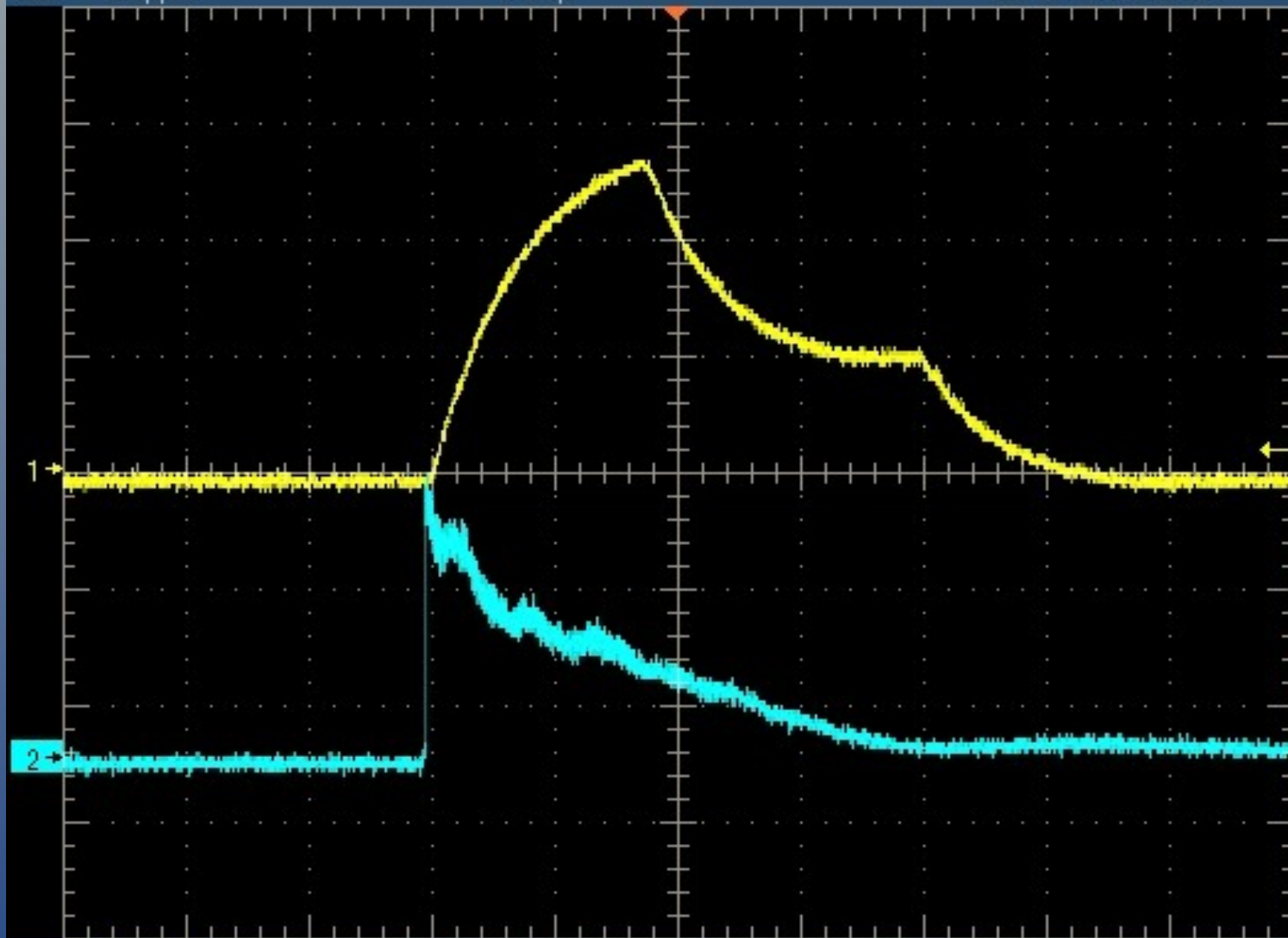




- Triggers as port pin edge interrupts
- Gates as GPIO
- Potentiometers into ADC ports
- PWM outputs filtered at about 10Hz

# ADSR Software Flow





Max(C1) 2.64V

Min(C1) -200.0mV

Ampl(C2) 1.16V  
Unstable histogram

Freq(C2) 447.0mHz  
Low signal amplitude

Freq(C1) -Hz  
Zero period

Ch1 1.0V

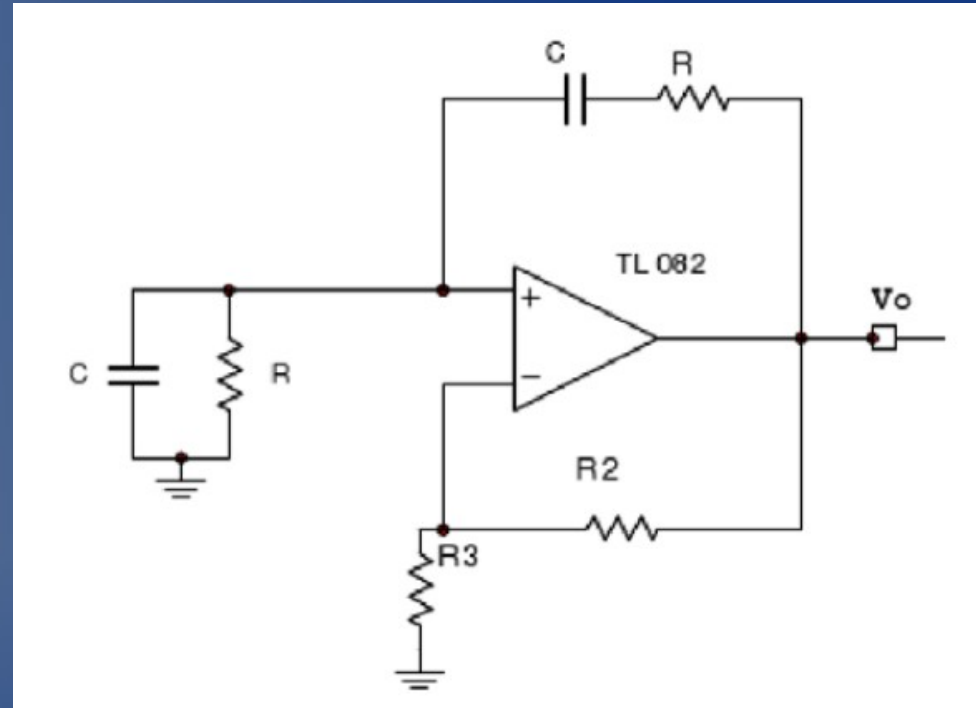
Ch2 1.0V

M 1.0s 500S/s 2.0ms/pt

A Ch1 / 160mV

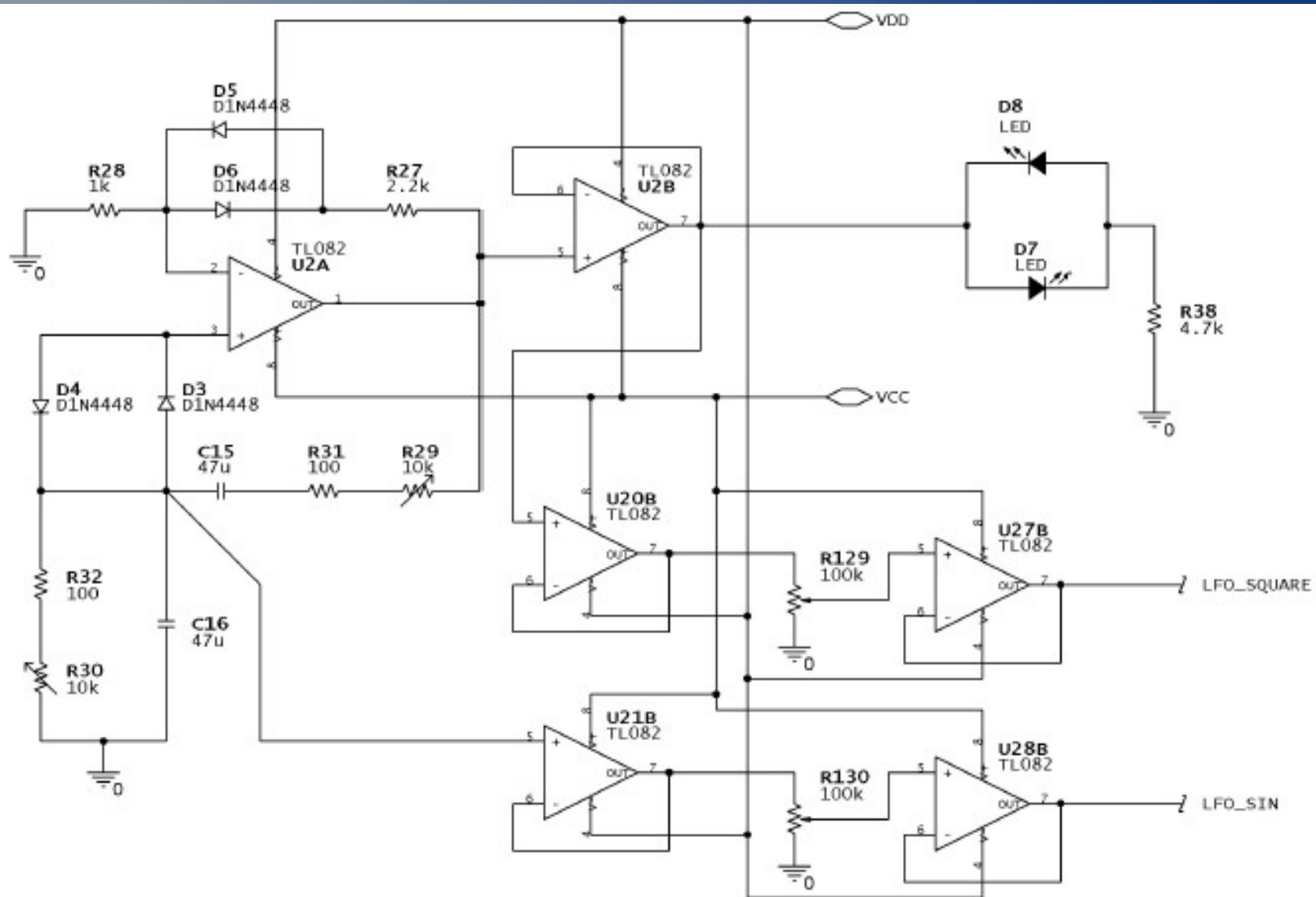
# Low Frequency Oscillator

- Based on a Wein-Bridge Oscillator.
- Produces frequencies from 3Hz to 33Hz.
- Produces both a sin wave and square wave oscillation, depending on where the signal is taken from.
- Will be used by vibrato, tremolo and the arpeggiator
- Independent of other designs, tracking is not an important issue.



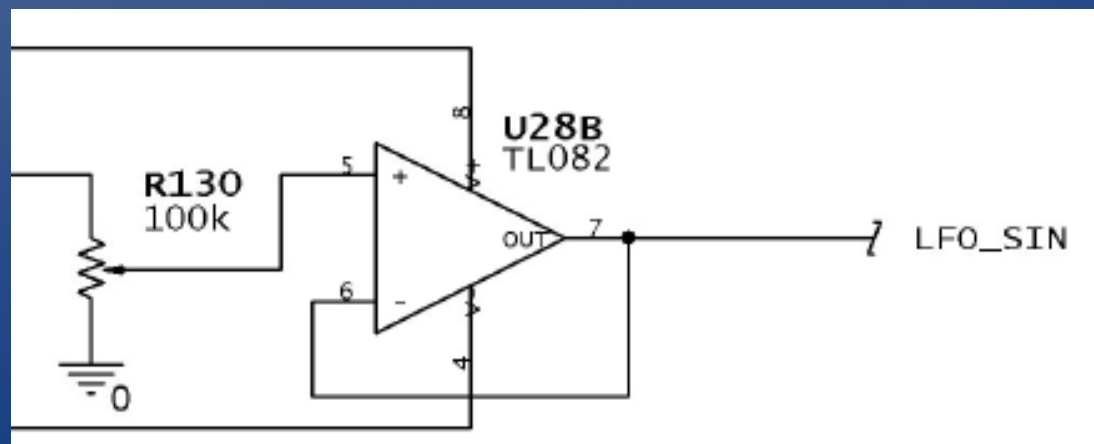
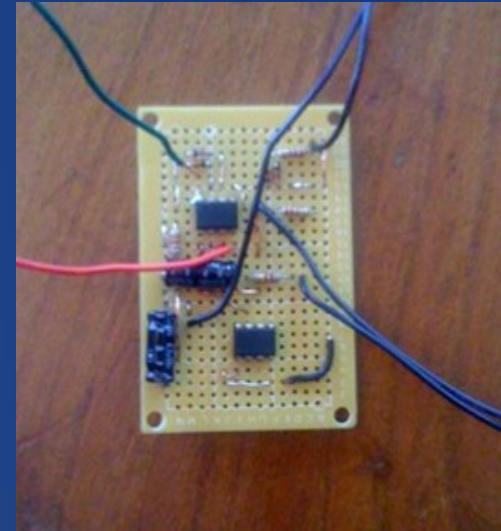
$$F_o = \frac{1}{2\pi RC}$$

# LFO Schematic



# LFO Rate and Depth

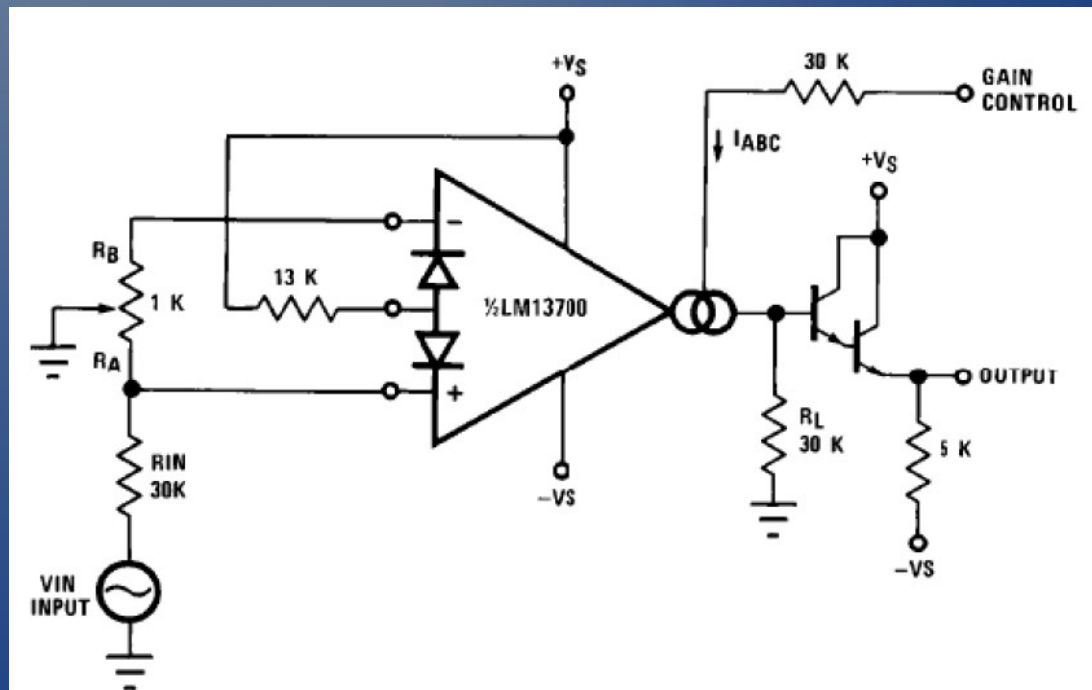
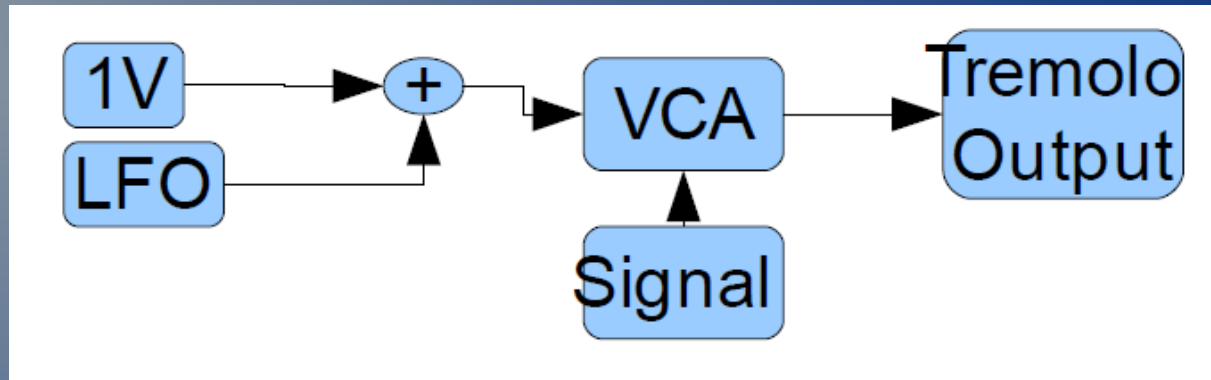
- Two potentiometers control the rate of oscillation. If they are Low the new resistance becomes 100 Ohms and the frequency is 33.3Hz. If they are high high the new resistance becomes 1100 Ohms and the frequency is 3.3 Hz.
- The amplitude of the LFO will be controlled independently for each effect.



# Tremolo

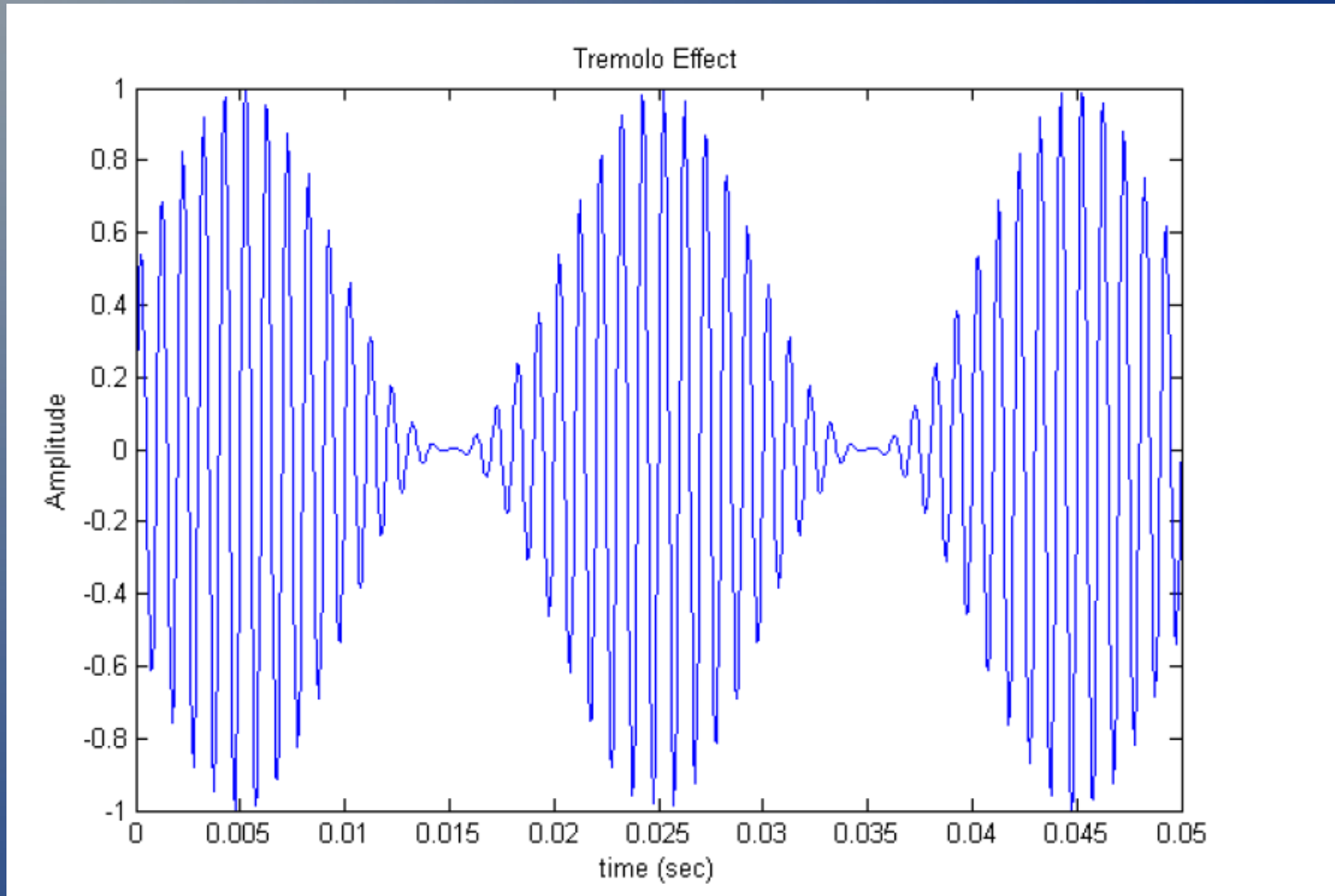
- Rapid increase and decrease in the amplitude (volume) of a signal.
- A signal is buffered from the LFO an adjustable amplitude (depth) and a maximum of 1V pk-pk and is used as the gain control.
- Added with a DC voltage of 1v. If the depth is set at its maximum. The signal will completely be attenuated when the LFO is negative
- The potentiometer is adjusted to minimize the effects of the control signal at the output

# Tremolo





# Tremolo

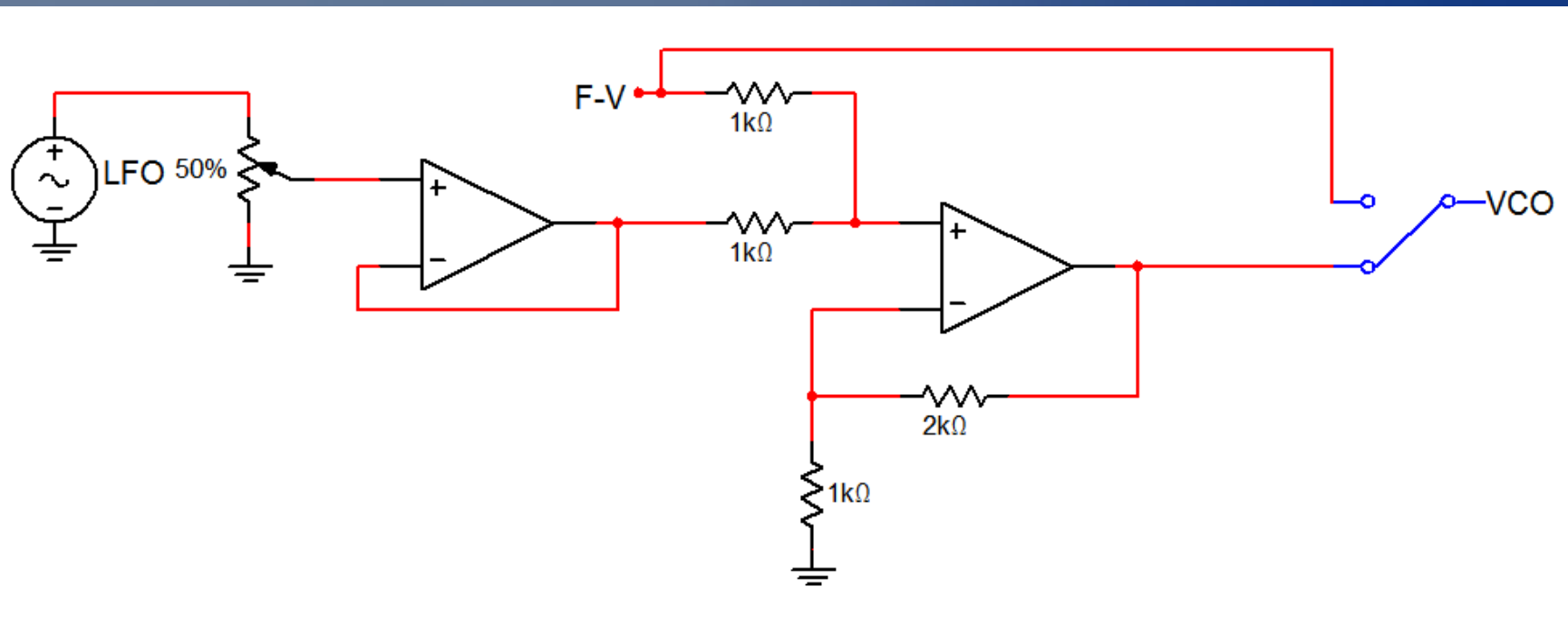
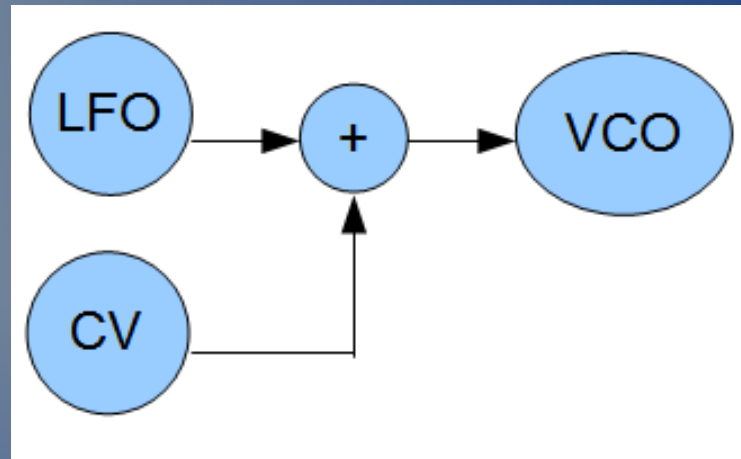


Sine wave LFO applied to control the amplitude of the signal. As the rate is increased on the LFO the rate of modulation changes as well.

# Vibrato

- Increase and decrease in frequency (pitch) of a signal.
- This is done by attenuating the signal from the LFO and adding it to the output of the F-V using the summing op-amp configuration.
- Every string has its own independent summer. Resulting in 6 SPDT or 3 DPDT (each controlling two strings) so the effect can be turn on for some strings and off for others

# Vibrato

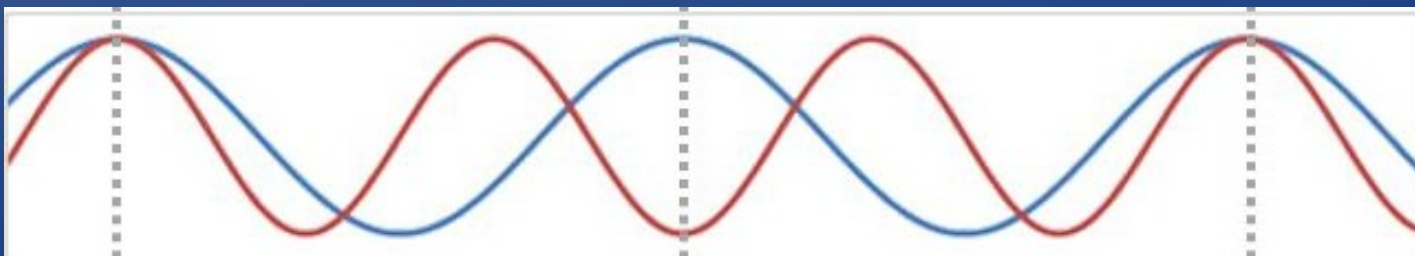


# Arpeggiator

- Automatically steps through a sequence of notes based on the input note, thus creating an arpeggio.
- This feature takes advantage of the linearity between the F-V converter and the VCO.
- ex. If an input frequency of 41.20 Hz (E) outputs a dc voltage of 1.6V, in order to hit the next E octave (82.41 Hz) on the VCO the output voltage of 1.6V would need to be doubled to 3.2V.
- Major chord – root, major 3rd, perfect 5th.
- Minor chord – root, minor 3rd, perfect 5th.

# Arpeggiator

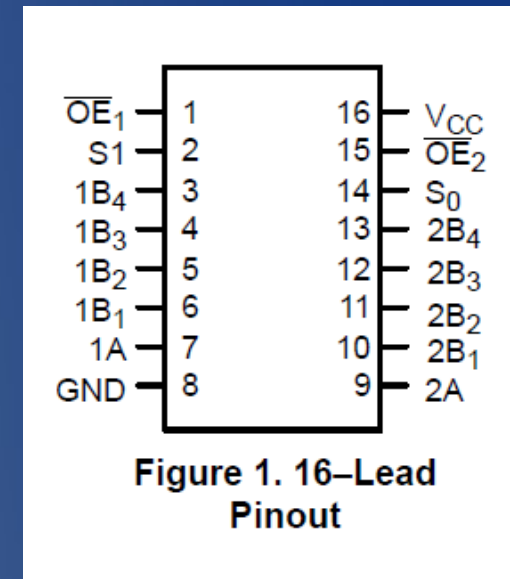
Note	Ratio	Interval
0	1:1	<u>Unison</u>
1	16:15	Major semitone
2	9:8	Major 2 <sup>nd</sup>
3	6:5	Minor 3 <sup>rd</sup>
4	5:4	Major 3 <sup>rd</sup>
5	4:3	Perfect 4 <sup>th</sup>
6	45:32	<u>Diatonic tritone</u>
7	3:2	Perfect 5 <sup>th</sup>
8	8:5	Minor 6 <sup>th</sup>
9	5:3	Major 6 <sup>th</sup>
10	9:5	Minor 7 <sup>th</sup>
11	15:8	Major 7 <sup>th</sup>
12	2:1	Octave



# Wavelength Ratios



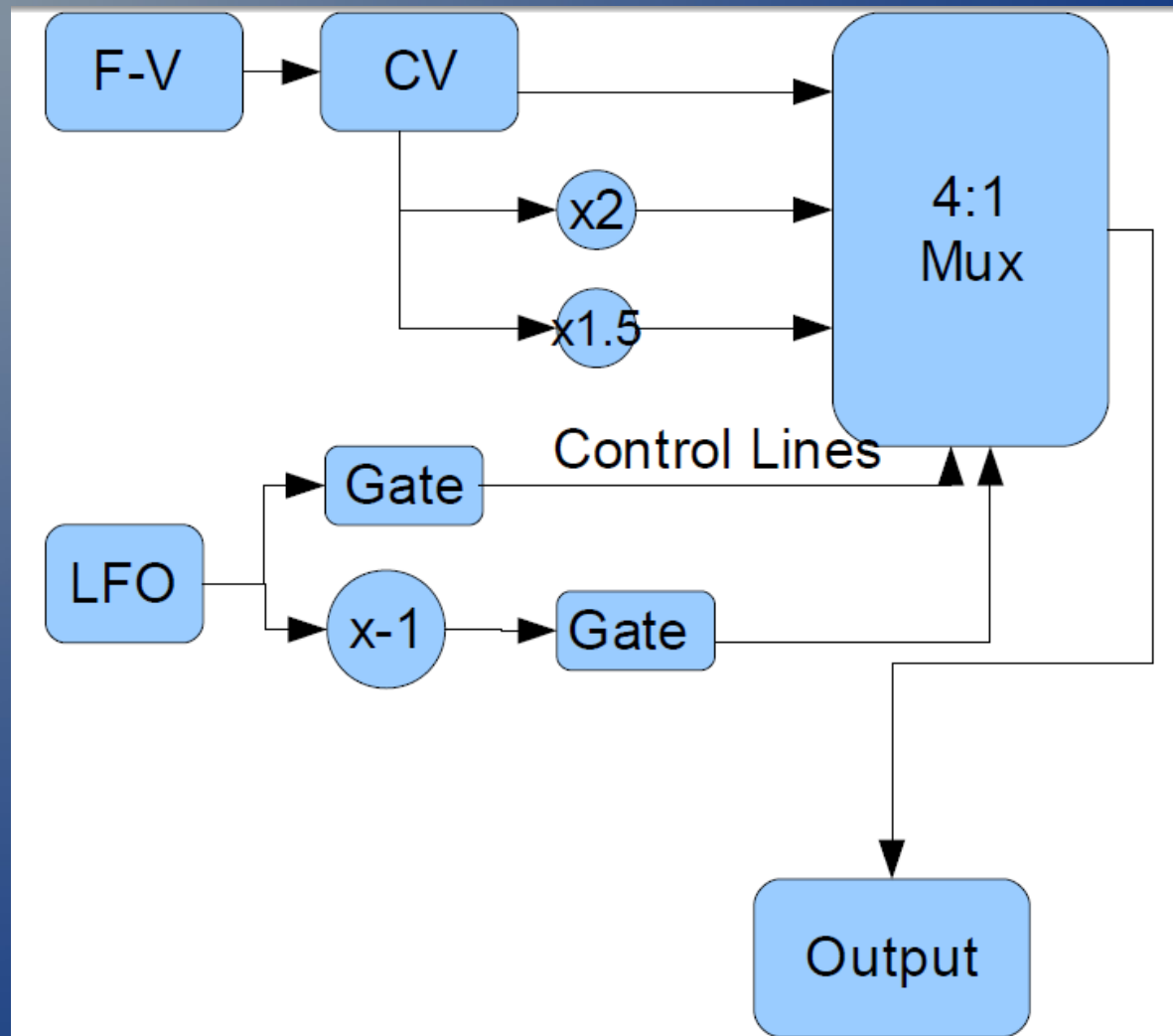
- Single op-amp non inverting gain configuration takes the F-V signal and applies the appropriate gain to each line
- Root – 1:1 no multiplication required
- Major 3rd – 5:4 requires x 1.25
- Perfect 5th – 3:2 requires x 1.5
- Octave – 2:1 requires x 2
- Analog 4:1 Mux 74FST3253 offers high frequency response with 7.5 nS switching times.
- Operates from -.5v to 7v



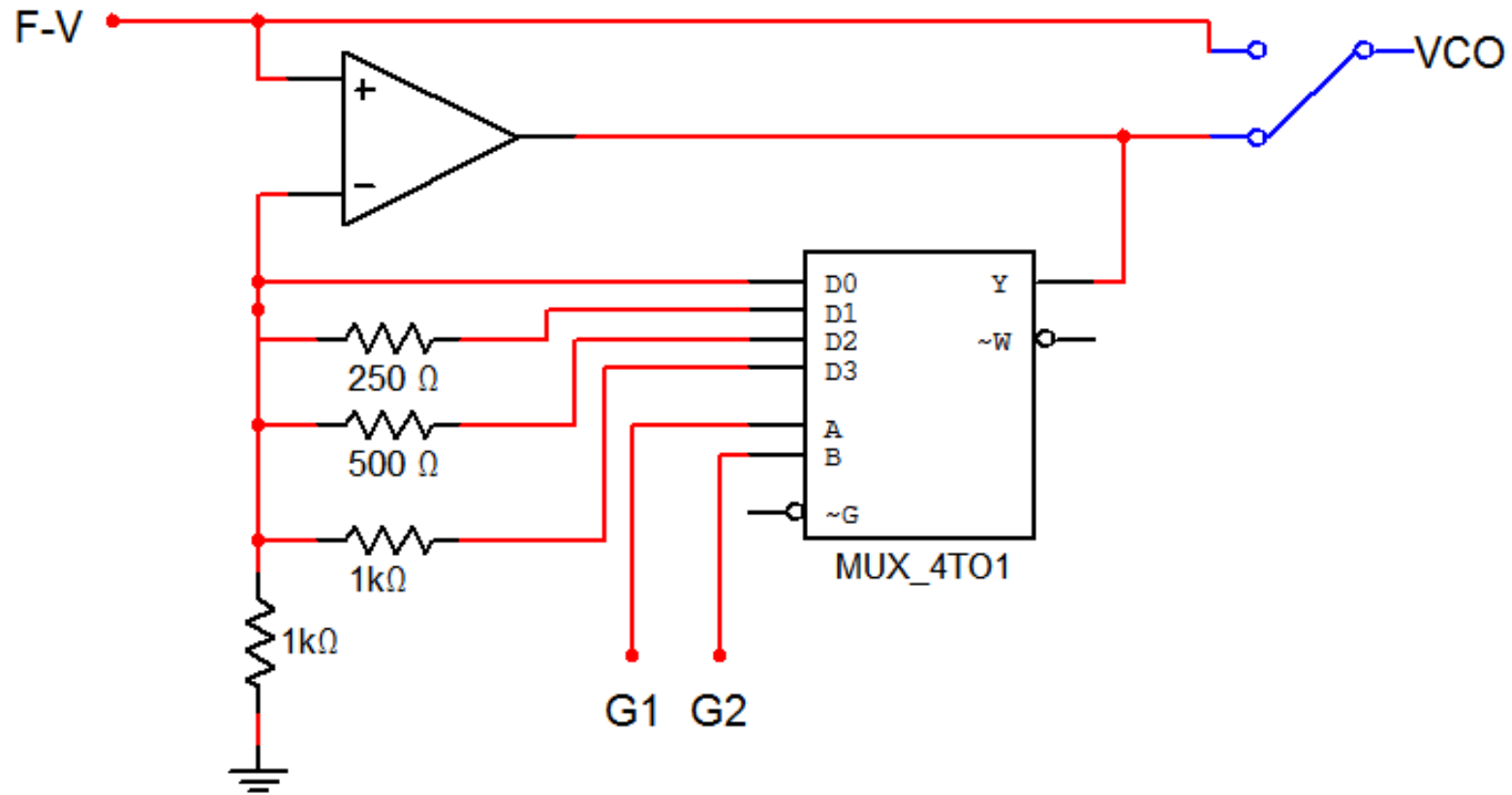
S <sub>1</sub>	S <sub>0</sub>	$\overline{OE}_1$	$\overline{OE}_2$	Function
X	X	H	X	Disconnect 1A
X	X	X	H	Disconnect 2A
L	L	L	L	A = B <sub>1</sub>
L	H	L	L	A = B <sub>2</sub>
H	L	L	L	A = B <sub>3</sub>
H	H	L	L	A = B <sub>4</sub>

**Figure 2. Truth Table**

# Arpeggiator Block Diagram



# Arpeggiator



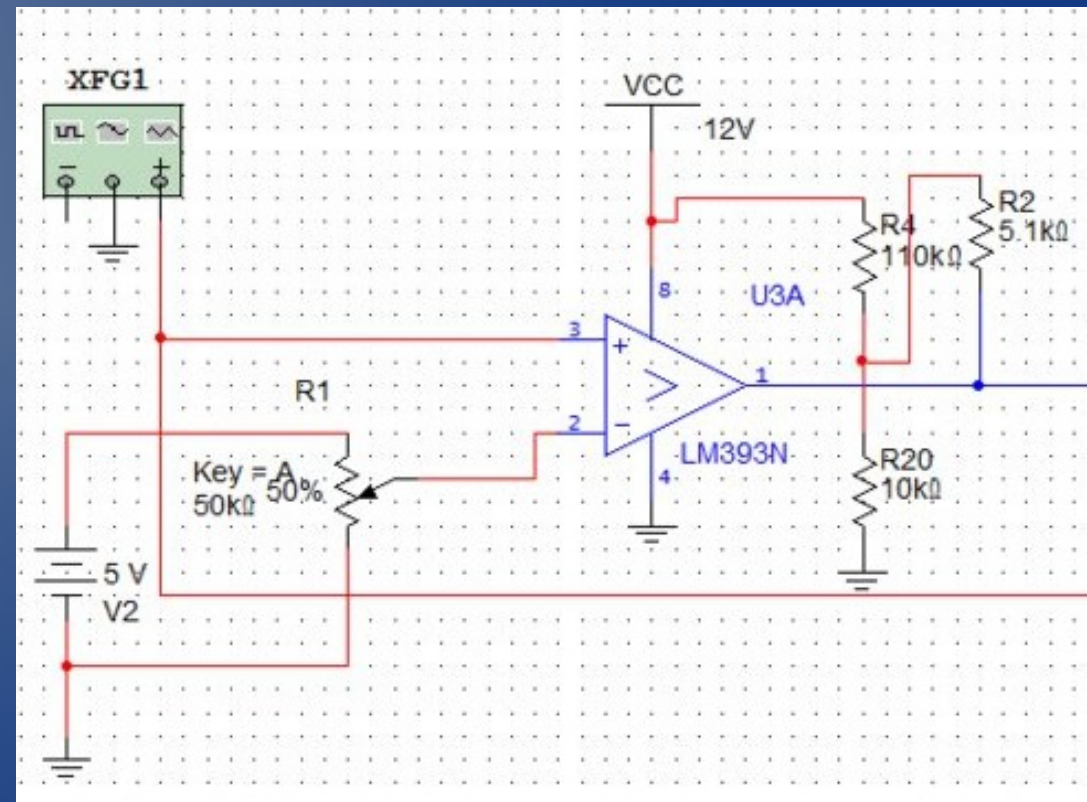


# 4:1 Mux Control Lines

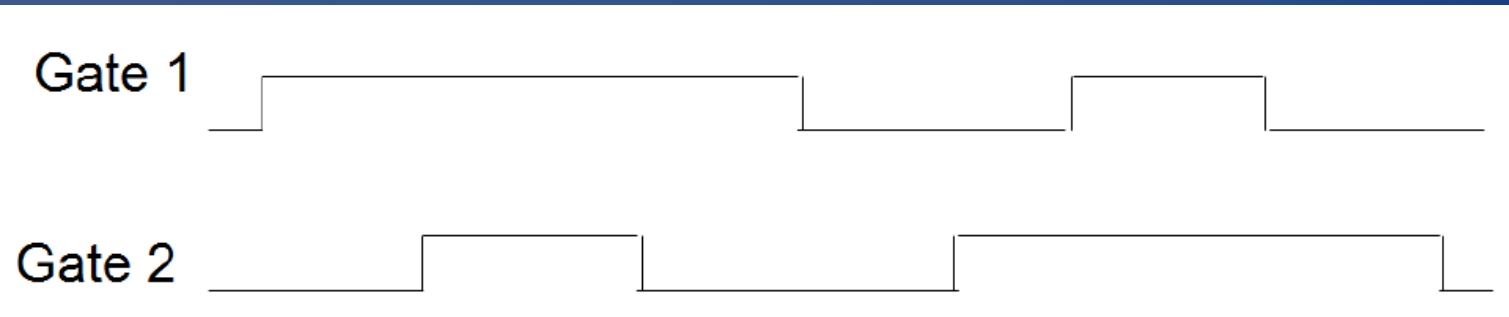
- Uses a rectified signal from the LFO with a normalized amplitude of 2V through two gates with different adjustable threshold voltages.
- When the LFO voltage is below the threshold voltage G1 the output is  $V_{low}$ .
- When the LFO voltage is higher the threshold voltage on G1 the gate outputs  $V_{high}$ .

# Schmitt Trigger Comparator

- There is only 2 gates that control the arpeggiation of all 6 strings.
- Potentiometer controls the threshold voltage and passes it through a comparator.



# Control Lines

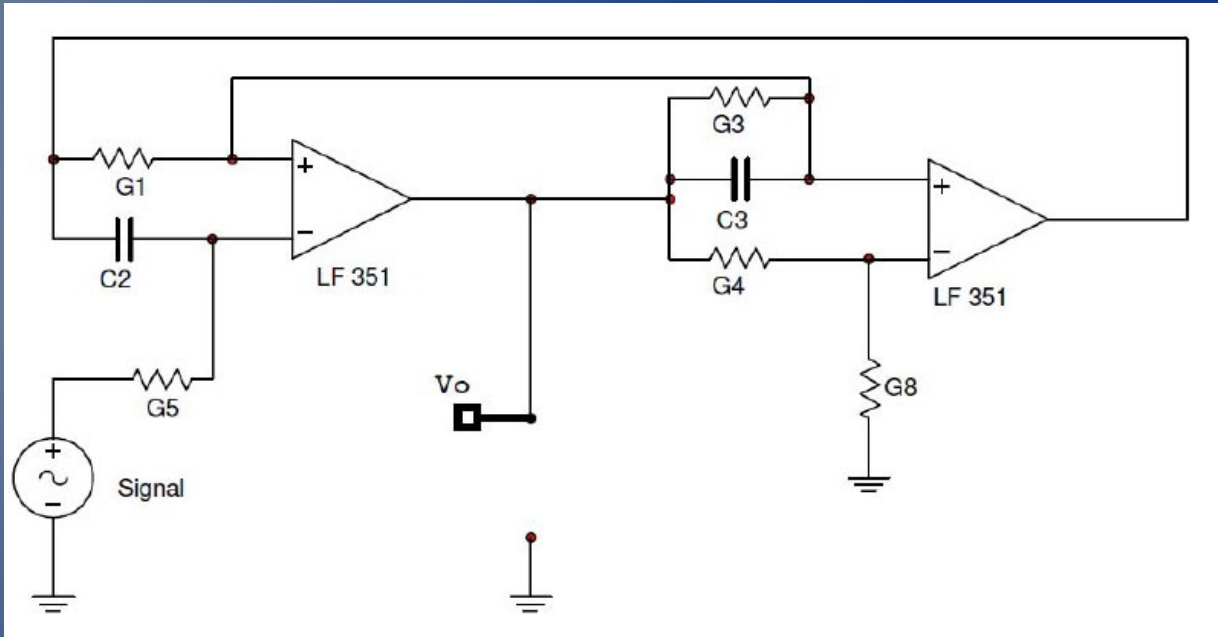


# Cutoff and Resonance Filter

- A low pass filter where the user has the ability to tune where the cutoff frequency is (between 10 Hz and 1.2Khz).
- A filter that has the ability for the user to tune how much resonance is desired on the output.
-

# CGIC

- Based on Dr. Wasfy Mikhael CGIC LP filter design.
- If  $G_3$  is tuned only the resonance changes.
- If  $G_4$  or  $G_8$  is tuned the gain, cutoff and resonance are affected.

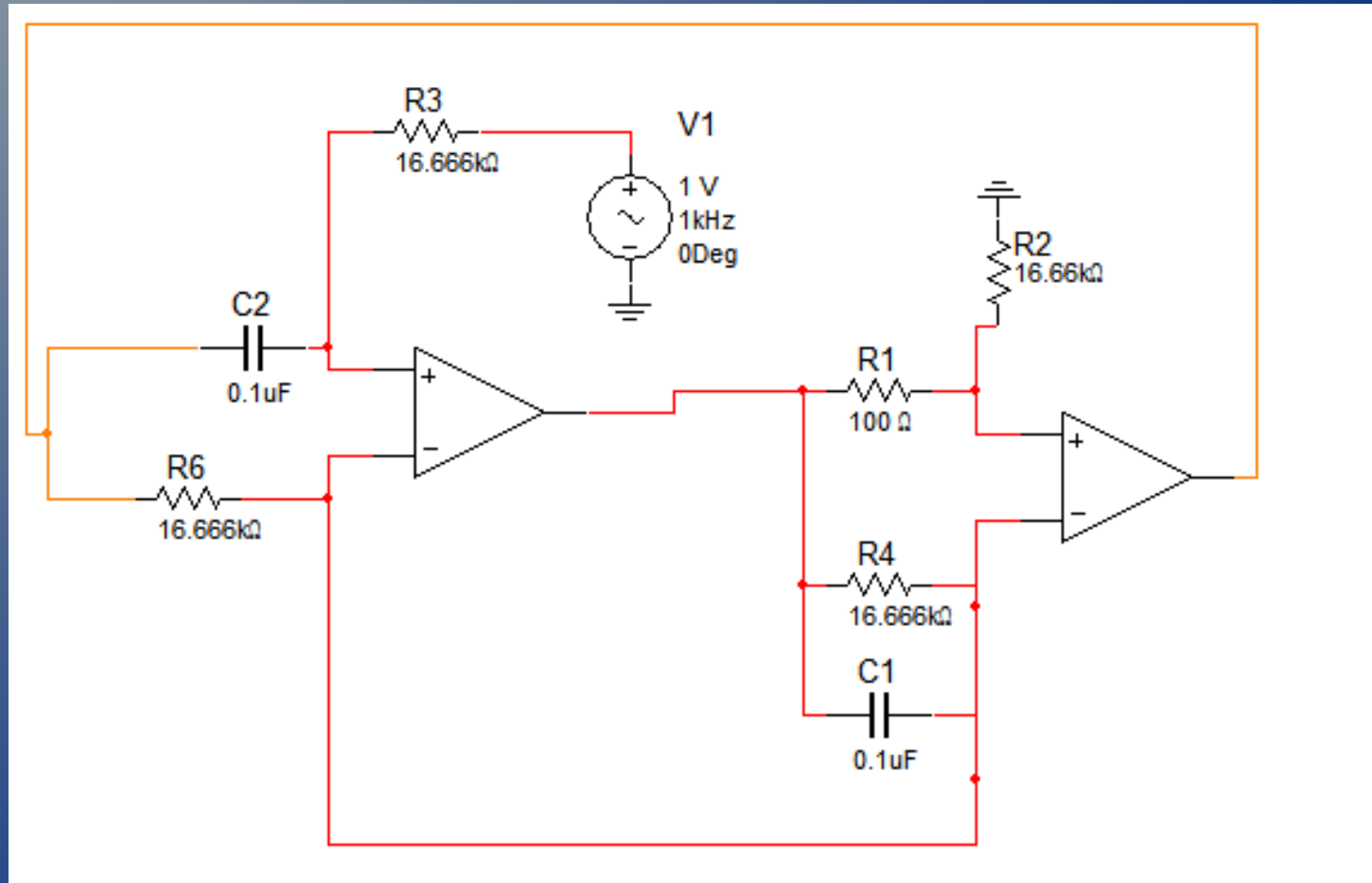


$$T(s) = \frac{1 + \frac{G_8}{G_4}}{s^2 \frac{C_2 C_3 G_8}{G_1 G_5 G_4} + s \frac{C_2 G_3 G_8}{G_1 G_5 G_4} + 1}$$

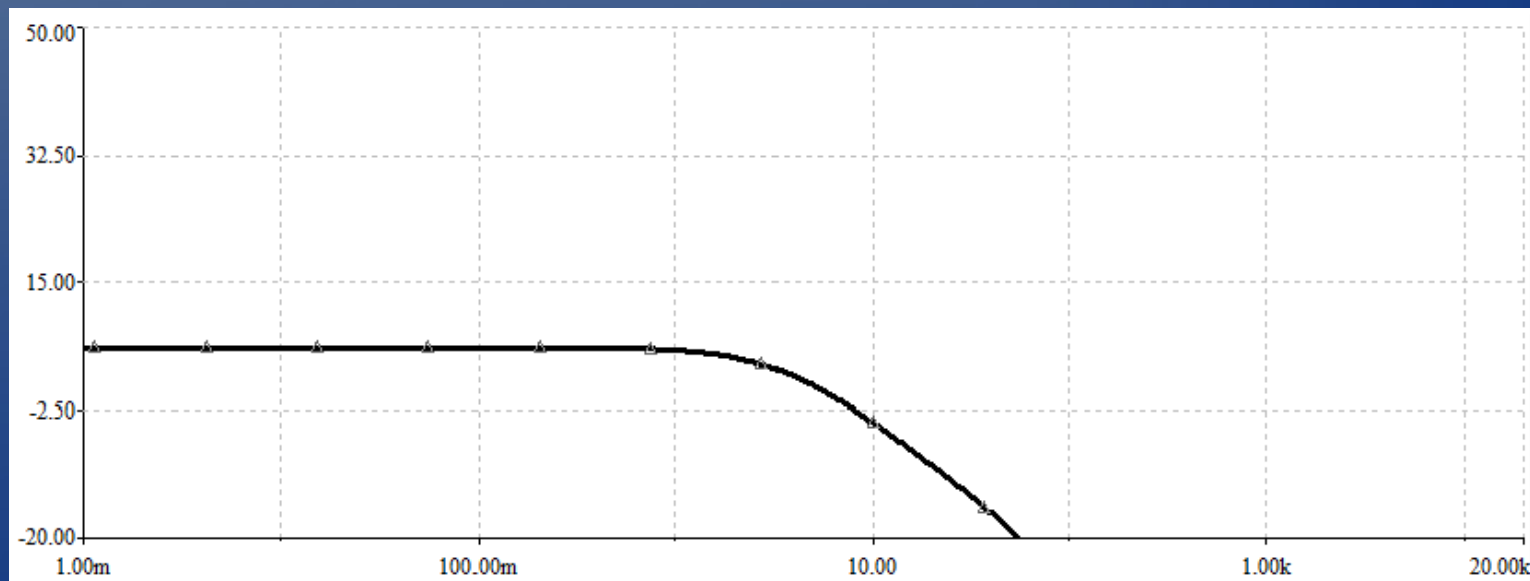
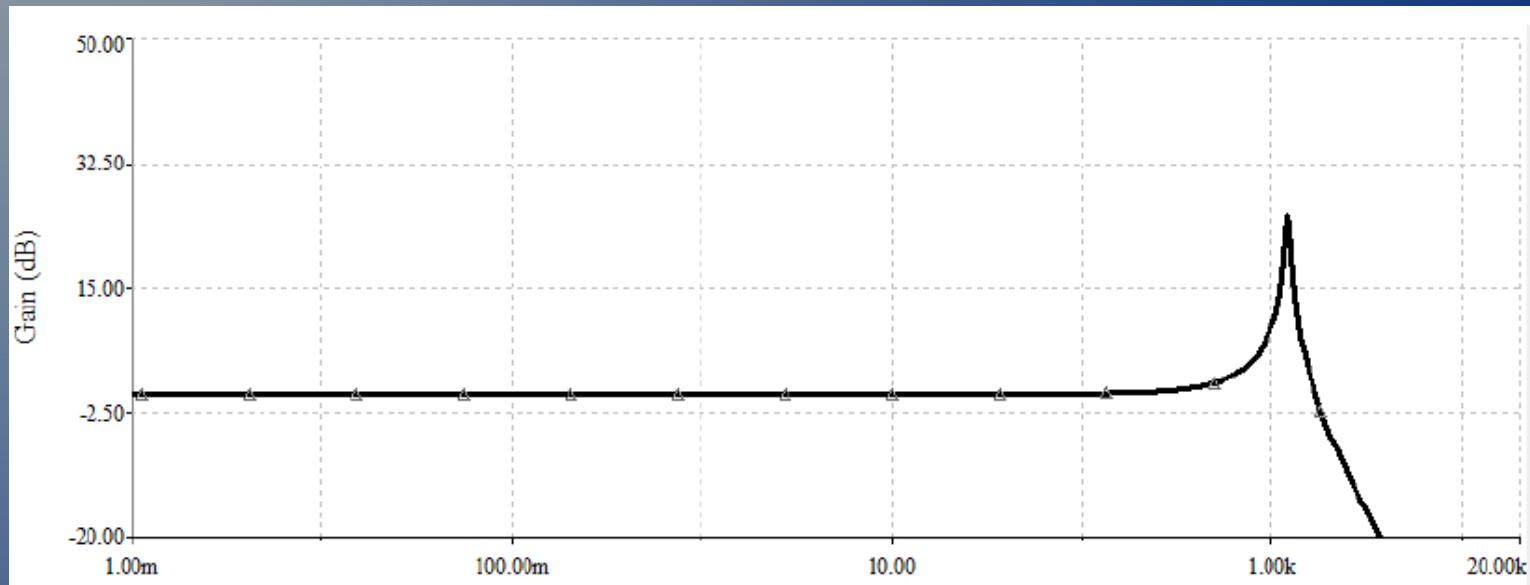
$$\omega_p^2 = \frac{G_1 G_5 G_4}{C_2 C_3 G_8}$$

$$Q_p^2 = \frac{C_3 G_1 G_4 G_5}{C_2 G_3^2 G_8}$$

# CGIC



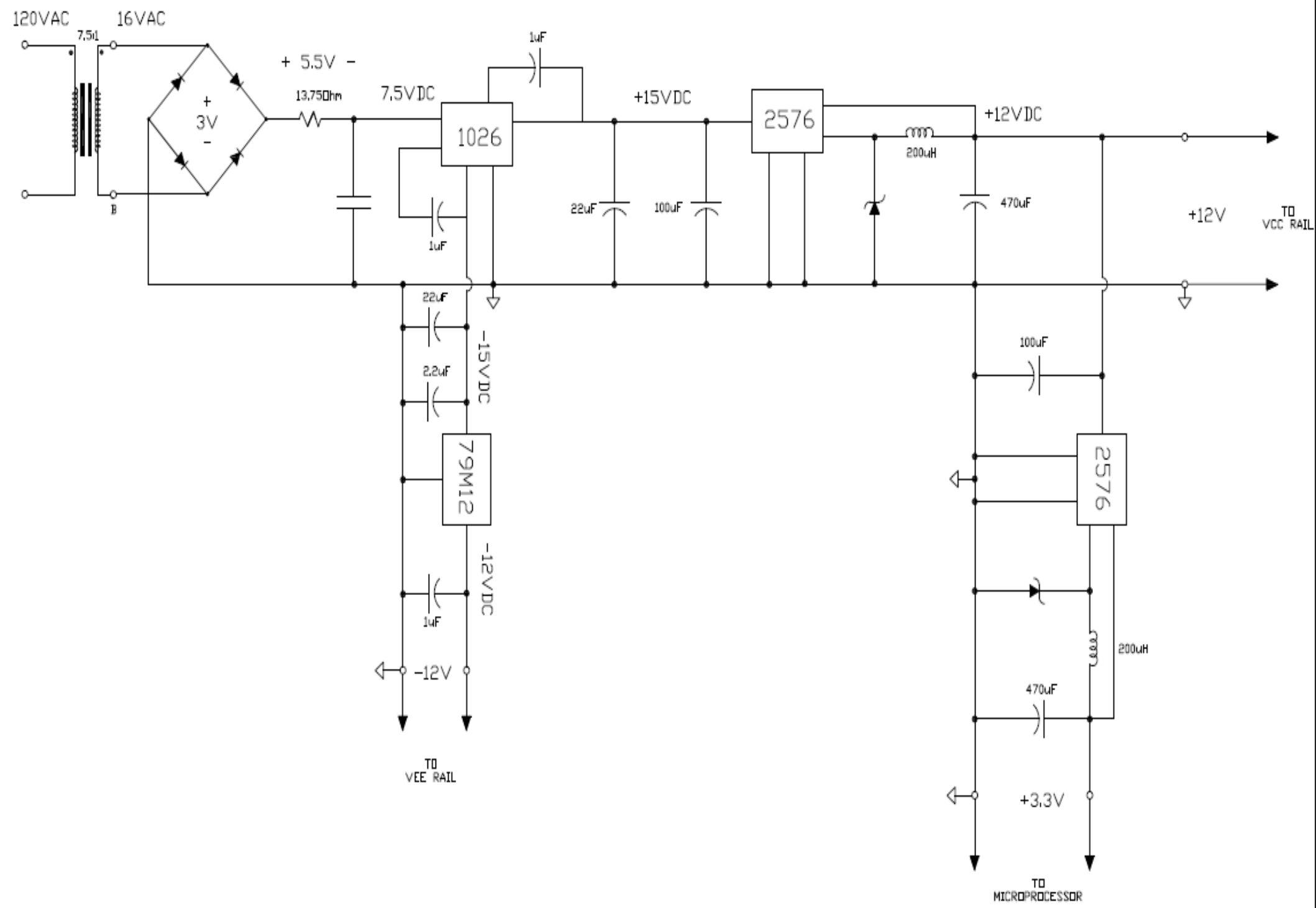
# Cutoff and Resonance



# Power Supply Requirements

- Supply +/-12Vdc and +3.3Vdc
- $1.0A < A < 2.0A$  from +/-12V supply
- $250mA < A < 1.0A$  from +5V supply

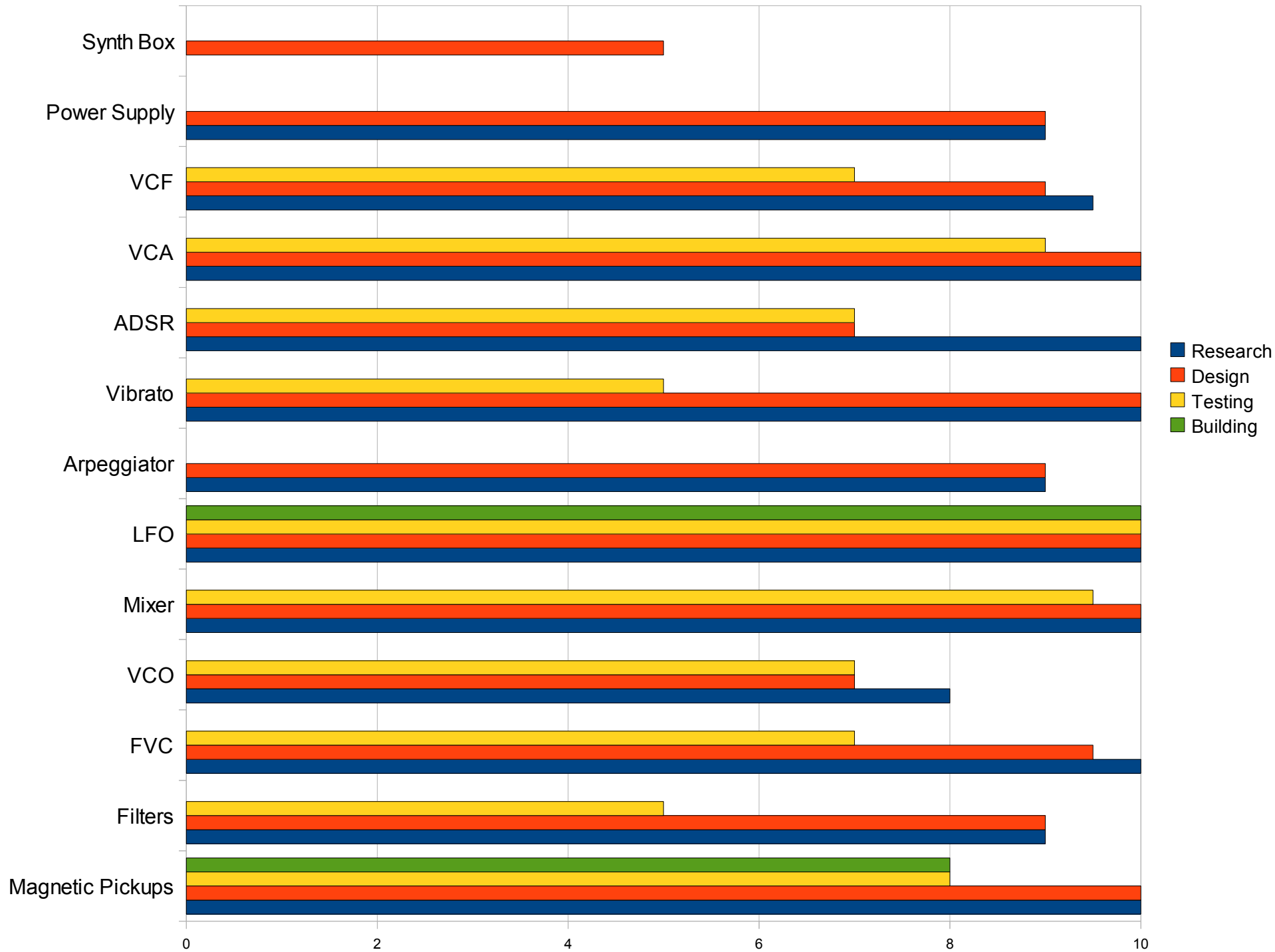




# Parts Required

- 1 PCB transformer (120VAC:16VDC)
- 2 LM2576 switching regulators
- 1 LM79M12 negative voltage regulator
- LT1026 DC-DC voltage converter
- 10 Capacitors
- 1 Resistor
- 2 Inductors
- 2 Schottky diodes
- 3 Heat sinks--8.9degC/W

Progress



<u>Part</u>	<u>Qty.</u>	<u>Cost ea.</u>	<u># Bought</u>	<u>\$ Spent</u>	<u>Projected Cost</u>
<b>ICs</b>					
LM13700	8	\$1.50	8	\$12.00	\$12.00
511-TL081CN	40	\$0.58	20	\$11.16	\$23.20
LM2902KNE4	6	\$0.51	6	\$3.06	\$3.06
LM2907	6	\$1.86	2	\$3.72	\$11.16
TL072	2	\$0.69	0	\$0.00	\$1.38
TL074	5	\$0.99	0	\$0.00	\$4.96
NJM13700M	3	\$1.02	3	\$3.06	\$3.06
LM339	3	\$0.49	0	\$0.00	\$1.47
MSP430F4132	1	\$4.56	1	\$4.56	\$4.56
UA78L09ACLP	1	\$0.58	0	\$0.00	\$0.58
KA7909TU	1	\$0.63	0	\$0.00	\$0.63
LM2576	2	\$2.24	2	\$4.48	\$4.48
LM79M12	1	\$0.61	1	\$0.61	\$0.61
LT1026	1	\$2.50	1	\$2.50	\$2.50
4:1 Multiplexer	1	\$7.00	1	\$7.00	\$7.00
Total				\$52.15	\$80.65
<b>Small Components</b>					
Resistors	556	\$0.06	0	\$0.00	\$33.34
Capacitors	73	\$0.19	0	\$0.00	\$13.87
Inductors	2	\$0.28	2	\$0.56	\$0.56
Schottky Diodes	2	\$1.00	2	\$2.00	\$2.00
1N4148 Diodes	10	\$0.02	0	\$0.00	\$0.20
1N4001	2	\$0.16	0	\$0.00	\$0.32
LEDs	12	\$0.10	0	\$0.00	\$1.20
Total				\$0.00	\$51.49
<b>Potentiometers</b>					
Trim Pot	32	\$0.50	0	\$0.00	\$16.00
Panel Pot	4	\$1.25	0	\$0.00	\$5.00
Slide Pot	12	\$2.08	12	\$24.96	\$24.96
Total				\$24.96	\$45.96

<b>Boards</b>					
Dev Board	1	\$4.30	1	\$4.30	\$4.30
LQFP 64 PCB Adapter	1	\$1.75	1	\$1.75	\$1.75
Through-hole PCB	5	\$3.00	1	\$3.00	\$15.00
Copper-plated Board	6	\$5.00	0	\$0.00	\$30.00
Total				\$9.05	\$51.05
<b>Hardware</b>					
DPDT Toggle Switch	7	\$2.92	6	\$17.52	\$20.44
SPDT Toggle Switch	1	\$0.00	1	\$0.00	\$0.00
Heat Sinks	3	\$0.85	3	\$2.55	\$2.55
Transformer	1	\$8.35	1	\$8.35	\$8.35
AlNiCo Magnets and Magnetic wire		\$10.00	1	\$10.00	\$10.00
VGA Connectors	1	\$6.00	1	\$6.00	\$6.00
VGA Cable	1	\$0.00	1	\$0.00	\$0.00
Power Supply Box	1	\$6.00	0	0	\$6.00
Power Supply Tri-Cable	1	\$8.00	0	\$0.00	\$8.00
Total				\$44.42	\$61.34
			Initial Estimate	Total Spent	Projected Total
			\$212.04	\$130.58	\$290.49

