Lab 3 – RC F	ilters and Basic Timer Functionality	February 24, 2017
Report Author:	Eric Rosenfeld	

Lab Team:Eric RosenfeldLab Date:February 24, 2017

# **Summary**

This report details the characteristics of RC filters and the results obtained from incorporating RC filters into circuits. An input voltage was supplied into the RC filter circuit by the function generator, then output voltage was measured using the oscilloscope. The capabilities and applications of the Arduino Microcontroller are further explored in this lab, and now incorporates using a speaker to become familiar with more inputs and outputs connected to the header pins on the Arduino. Additionally, the timer function on the function generator was utilized to create a tone for a duration of time. Version 4.5 of the EduShields YouKnow board was used for the experiments.

# **Results/Discussion**

## **Introduction**

## **RC Filter Theory: Low-Pass Filter**

In a given higher order circuit, transfer functions are derived to give the ratio of output voltage ( $V_{OUT}$ ) to input voltage ( $V_{IN}$ ) as well as the order of the circuit. For a low-pass filter, for example, the transfer function is (Equation 2 from the ME 106 Lab Manual):

$$\frac{V_{OUT}}{V_{IN}} = \frac{1}{1 + j\omega RC}$$

where:  $V_{OUT} = Output voltage$   $V_{IN} = Input voltage$   $j = a \text{ complex number } \sqrt{-1}$   $\omega = \text{Frequency of } V_{IN} \text{ in } \frac{rad}{s}$  R = Resistor ValueC = Capacitor Value

Because they are inversely related, as the frequency of the input voltage increases, the value of the transfer function  $\frac{V_{OUT}}{V_{IN}}$  decreases.

The low-pass filter found in the ME 106 manual was constructed on the breadboard for this exercise. The resistor value was 1.1 k $\Omega$  and the capacitor value was 0.1  $\mu$ F.

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Figure 1. RC Low-Pass Filter circuit constructed on the breadboard. V<sub>OUT</sub> is measured across the Capacitor.

The function generator was set up to output a sine wave at  $V_{IN} = 5 V_{P-P}$  for various frequencies. The amplitude of  $V_{IN}$  and  $V_{OUT}$  was then measured via the oscilloscope.

Frequency [Hz]	<b>Frequency</b> $\left[\frac{rad}{s}\right]$	<b>V</b> <sub>IN</sub> [ <b>mV</b> ]	Vout [mV]
<b>"I</b> "	$\omega = 2\pi f$		
500 Hz	$3141.593 \frac{rad}{s}$	506.3 mV	468.8 mV
1.6 kHz	$10053.1 \frac{rad}{s}$	506.3 mV	343.8 mV
10 kHz	$62831.9 \frac{rad}{s}$	506.3 mV	84.38 mV

Additionally, the ratio of the output voltage to the input voltage  $\frac{V_{OUT}}{V_{IN}}$  that were measured via the oscilloscope were compared with the magnitude of the transfer function from Equation 2 above. Substituting R = 1.1 k $\Omega$  and C = 0.1  $\mu$ F into the transfer function for low-pass filters from above, the magnitude of the transfer function becomes the following equation:

$$\frac{V_{OUT}}{V_{IN}} = \frac{1}{1 + j\omega \times (1.1 \ k\Omega) \times (0.1 \ \mu F)}$$

Each V<sub>IN</sub> angular frequency in  $\frac{rad}{s}$  is then substituted into the function. Afterwards, the function is evaluated. Lastly, the magnitude of the real and imaginary part of the function is taken.

Frequency $\left[\frac{rad}{s}\right]$ $\omega = 2\pi f$	Ratio $\frac{V_{OUT}}{V_{IN}}$	<b>Magnitude of Transfer Function</b> $\ \frac{V_{OUT}}{V_{IN}}\  = \sqrt{(REAL)^2 + (IMAGINARY)^2}$
$3141.593 \frac{rad}{s}$	0.92593	$\sqrt{(0.9988039104)^2 + (-0.34563839)^2} = 0.9994017763$
$10053.1  \frac{rad}{s}$	0.67904	$\sqrt{(0.4498659657)^2 + (-0.4974802294)^2} = 0.6707204826$
$62831.9 \frac{rad}{s}$	0.166660	$\sqrt{(0.0205048471)^2 + (-0.1417194353)^2} = 0.1431951364$

The percent differences are: 7.63212%, 1.23274%, and 15.1457% respectively. (Question 1) Although the percent difference corresponding to the 1.6 kHz was at a low 1.23274%, the other two percent differences are high probably due to the fluctuations in the output voltage seen on the oscilloscope during lab. The larger the frequencies became, the more that the output voltage was reduced. This circuit in which the resistor and capacitor are configured as such (with Vour being measured across the capacitor) is called a low-pass filter because lower frequencies are "passed" to the output with little attenuation [cut-off] and higher frequencies are significantly attenuated (i.e., not "passed") (Alciatore & Histand, 2011, p. 128).

The low-pass filter circuit was still in use for the following task: the phase lag of the low-pass circuit at 1.6 kHz was to be measured on the oscilloscope. The phase lag is the measure of the amount of time that the output voltage lags behind the input voltage, and it is measured in degrees (Alciatore & Histand, 2011, p. 129). First, the theoretical value of the phase lag is calculated for 1.6 kHz, using the Imaginary and Real parts of the Transfer Function:

$$\phi = \left(\frac{1MAGINARY}{REAL}\right)$$
$$= \left(\frac{0.4974802294}{0.4498659657}\right) = 47.877^{\circ}$$

Next, the observed value of the phase lag is calculated for 1.6 kHz. The procedure listed in the lab manual was followed in order to align the input voltage waveform and the output voltage waveform, as shown in the image below.

For  $V_{IN}$ , Period [T] = 625 µs For  $V_{OUT}$ , Period [T] = 625 µs

$$t_1 = 400 \ \mu s$$
$$t_2 = 484 \ \mu s$$
$$\Delta t = t_2 - t_1 = 84 \ \mu s$$
$$\phi = \frac{(360) \times (\Delta t_d)}{T}$$

where:  $\Phi$  = Phase lag, or phase angle

 $\Delta t_d$  = time displacement between the input and output signal [sec]

T = Period of the signals [sec]

$$=\frac{(360)\times(84\,\mu s)}{625\,\mu s}=48.384^{\circ}$$

(Alciatore & Histand, 2011, p. 129). (Question 2) One period on the waveform represents 360° (Furman, 2017, p. 3). The theoretical value of 47.877° agrees with the observed value of 48.384°, with a 1.05339% difference. This means that the values for the Transfer Function are correct, and the time displacement is correct.



Figure 2. The input voltage waveform and the output voltage waveform are depicted after aligning them so that it would be easier to measure the phase lag.

#### **RC Filter Theory: High-Pass Filter**

As stated earlier, transfer functions are derived to give the ratio of output voltage ( $V_{OUT}$ ) to input voltage ( $V_{IN}$ ) as well as the order of the circuit. Now, the transfer function for a high-pass filter is (Equation 4 from the ME 106 Lab Manual):

$$\frac{V_{OUT}}{V_{IN}} = \frac{j\omega RC}{1 + j\omega RC}$$

where:  $V_{OUT} = Output$  voltage

 $V_{IN} =$  Input voltage j = a complex number  $\sqrt{-1}$  $\omega =$  Frequency of  $V_{IN}$  in  $\frac{rad}{s}$ 

- R = Resistor Value
- C = Capacitor Value

Now, the high-pass filter found in the ME 106 manual was constructed on the breadboard for this exercise. The capacitor value and the resistor value were once again 0.1  $\mu$ F and 1.1 k $\Omega$ , respectively.



Figure 3. RC High-Pass Filter circuit constructed on the breadboard. Notice that the Resistor and Capacitor have switched places. Now, V<sub>OUT</sub> is measured across the Resistor.

Again, the function generator was set up to output a sine wave at  $V_{IN} = 5 V_{P-P}$  for various frequencies. The amplitude of  $V_{IN}$  and  $V_{OUT}$  was then measured via the oscilloscope.

Frequency [Hz]	<b>Frequency</b> $\left[\frac{rad}{s}\right]$	V <sub>IN</sub> [mV]	VOUT [mV]
<b>"I</b> "	$\omega = 2\pi f$		
500 Hz	$3141.593 \frac{rad}{s}$	506.3 mV	153.1 mV
1.6 kHz	$10053.1 \frac{rad}{s}$	493.8 mV	340.6 mV
10 kHz	$62831.9 \frac{rad}{s}$	493.8 mV	493.8 mV

Additionally, the ratio of the output voltage to the input voltage  $\frac{V_{OUT}}{V_{IN}}$  that were measured via the oscilloscope were compared with the magnitude of the transfer function from Equation 4

above. Substituting  $C = 0.1 \ \mu F$  and  $R = 1.1 \ k\Omega$  into the transfer function for high-pass filters from above, the magnitude of the transfer function becomes the following equation:

$$\frac{V_{OUT}}{V_{IN}} = \frac{j\omega \times (1.1 \ k\Omega) \times (0.1 \ \mu F)}{1 + j\omega \times (1.1 \ k\Omega) \times (0.1 \ \mu F)}$$

Each V<sub>IN</sub> angular frequency in  $\frac{rad}{s}$  is then substituted into the function. Afterwards, the function is evaluated. Lastly, the magnitude of the real and imaginary part of the function is taken.

Frequency $\left[\frac{rad}{s}\right]$ $\omega = 2\pi f$	<b>Ratio</b> $\frac{V_{OUT}}{V_{IN}}$	<b>Magnitude of Transfer Function</b> $\ \frac{V_{OUT}}{V_{IN}}\  = \sqrt{(REAL)^2 + (IMAGINARY)^2}$
$3141.593 \frac{rad}{s}$	0.30620	$\sqrt{(0.1066820325)^2 + (0.3087085621)^2} = 0.3266221553$
$10053.1  \frac{rad}{s}$	0.68975	$\sqrt{(0.5501340343)^2 + (0.4974802294)^2} = 0.7417102091$
$62831.9 \frac{rad}{s}$	1.0	$\sqrt{(0.9794951529)^2 + (0.1417194353)^2} = 0.9896944745$

The percent differences are: 6.45431%, 7.25975%, and 1.03589% respectively. (Question 1) The percent differences are now reasonably low, meaning the fluctuations in the output voltage seen on the oscilloscope were less, and the was overall more accuracy. In contrast with the low-pass filter, in which case the larger the frequencies became the more that the output voltage was reduced, this time, the lower the frequency the more that the output voltage was reduced. This circuit in which the capacitor and resistor are configured as such (with Vour being measured across the resistor) is called a high-pass filter because higher frequencies are now "passed" to the output with little attenuation and lower frequencies are now "not-passed" and are more attenuated [cut-off] (Alciatore & Histand, 2011, p. 128). The ratio of  $\frac{V_{OUT}}{V_{IN}}$  for 10 kHz is 1.0, which most likely means that the high-pass filter has reached its limit – it cannot attenuate any frequencies higher than 10 kHz.

The oscilloscope was then used to measure the phase lag of the high-pass filter circuit at 1.6 kHz. The same results as with the low-pass filter were obtained. (Question 2) The theoretical value of the phase lag was 47.877° and the observed value was 48.384°, leading to a percent difference of 1.05339%. Again, this means that the values for the Transfer Function are correct, and the time displacement is correct.

Using the high-pass filter circuit, a DC offset was to be added from the function generator output. (Question 3) With a +2.00 V DC Offset value for 10 kHz, the peak-to-peak voltage of  $V_{IN}$  was 487.5 mV and the peak-to-peak voltage of  $V_{OUT}$  was also 487.5 mV. Adding a DC offset voltage does not affect the output voltage  $V_{OUT}$  of the high-pass filter because the capacitor blocks the DC component of the signal.

#### Using the Arduino to Output a Frequency

#### **Basic Introduction to IC Timer-Counters**

(Question 4) Using a 16-bit counter (Timer1 on the ATmega328), a chosen prescaler value of 8, and a standard 16 MHz clock tick, it will take the counter 32768 microseconds to reach its maximum value.

215	214	213	212	211	210	29	28	27	26	2 <sup>5</sup>	24	2 <sup>3</sup>	2 <sup>2</sup>	21	20
$2^{16} = (65536) \times (0.5 \mu s  counter  tick) = 32768 \mu s$															

#### Driving a Speaker with the Function Generator

An auxiliary speaker was used for this exercise. (Question 5) Using the Digital Multimeter (DMM), the DC impedance of the auxiliary speaker was measured to be 3.628  $\Omega$ . The function generator was then set up with a 1 V peak-to-peak (V<sub>P-P</sub>), 0 V offset, sine wave signal. Also, it was set for a 50  $\Omega$  output impedance because the speaker is a low-impedance load. The function generator was attached to the speaker, and the oscilloscope was attached across the them. Using the knob on the function generator to increase the frequency, the tone could no longer be heard at a frequency of 16 kHz (outside of human hearing range).

Next, the function generator was set up to sweep frequencies from 200-15,000 Hz using the procedure listed in the lab manual. (Question 6A) After the sweep button on the generator is **pressed, a low police siren sound, a low** *"whooo"*, is emitted from the speaker. The waveform was then scaled vertically on the scope screen to get a better view of the behavior of the waveform. To do so, Auto-scale was pressed, and the horizontal scale was changed from 1.00 ms/div to 2.00 ms/div using the horizontal Time/Div knob. (Question 6B) On the oscilloscope, frequency increased continuously until the sweep reached the end of the range (15,000 Hz), at which point the waveform began the cycle again. Also, the amplitude of the waveform spikes near the beginning of the sweep cycle probably because a sudden increase in frequency causes a sudden increase in voltage. This continues in an endless cycle, and images of each instance are displayed below.



Figure 4. During the beginning of the sweep cycle, the sinusoidal waveform observed on the oscilloscope started with a low frequency and low amplitude.



Figure 5. Also near the beginning of the sweep cycle, the frequency of the sinusoidal waveform observed on the oscilloscope increased as it swept across the range. Amplitude also spiked.



Figure 6. Frequency continued to increase until the end of the sweep (15,000 Hz) was reached. Amplitude had eventually reduced as well.

(Question 6C) The waveform on the function generator was switched from a sine-wave to a square-wave. The tone then sounded different: a louder, higher-pitched tone that sounded similar to an evacuation alarm emitted from the speakers. But the same behavior was observed as with the sinusoidal wave: frequency increased continuously until the sweep reached the end of the range (15,000 Hz), at which point the waveform began the cycle again. Also, the amplitude of the waveform spikes near the beginning of the sweep cycle probably because a sudden increase in frequency causes a sudden increase in voltage. This continues in an endless cycle, and images of each instance are displayed below.

Because the square wave is not smooth like the sine wave, the input signal will also not be smooth, so the tone will not be as smooth. This is similar to a capacitor charging and discharging, which is represented by an exponential graph. The sine wave may be preferred for maintaining a constant low-sounding noise/tone, while the square wave might be preferred for applications that require sudden noises, such as a fire or evacuation alarm.

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Figure 7. During the beginning of the sweep cycle, the square waveform observed on the oscilloscope started with a low frequency and low amplitude.



Figure 8. Also near the beginning of the sweep cycle, the frequency of the square waveform observed on the oscilloscope increased as it swept across the range. Amplitude also spiked.



Figure 9. Frequency continued to increase until the end of the sweep (15,000 Hz) was reached. Amplitude had eventually reduced as well.

Now, instead of using the function generator, the Arduino was used to output audio to a real, auxiliary speaker. The circuit shown in the lab manual was constructed on the breadboard. (Question 7) With the 3.628  $\Omega$  speaker and a 176.36  $\Omega$  resistor connected, the current limited was calculated using Ohm's Law and series resistors:

 $V = I \times R$ 

Where V = 5 V from the USB cable And R =  $3.628 \Omega + 176.36 \Omega = 179.988\Omega$ 

$$I = \frac{V}{R} = \frac{5 V}{179.888 \Omega} = 0.02778 A = 27.78 mA$$

(Question 8) Now, an Arduino program was made using the Arduino IDE that generated a sweep tone from a minimum of 100 Hz to a maximum of 15,000 Hz in 5 seconds. Refer to Appendix A for the pitches.h source code and to Appendix B for the sketch pertaining to this exercise.

(Question 9) Now, the Arduino program was modified to time the sweep sequences and print the result on the Serial Monitor. Refer to Appendix C for the sketch pertaining to this exercise.

(Question 10) The breadboard was put aside for the last exercise. The function generator was set up to produce an input square wave with 5 V<sub>P-P</sub> amplitude, 2.5 V offset, 50% duty cycle, Hi-Z termination, for any frequency between 1 kHz and 5 kHz. The frequency chosen was 10 kHz, and the generator was connected to digital pin 11 on the Arduino. An Arduino sketch was created that used the pulseIn() function to time the length of the high portion of the signal pulses from the function generator and printed it out on the Serial Monitor. Refer to Appendix D for the sketch pertaining to this exercise.

486	985
486	985
486	985
494	985
494	985
494	985
492	986
492	986
492	986
500	986
500	986
500	986
486	986
486	986
494	986
494	986
Autoscroll	Autoscro

Figure 10. Serial Monitor displayed the duration of the high portions of the signal pulses from the function generator.

# **Conclusions and Recommendations**

Lab 3 demonstrated the characteristics and uses of RC filters: low-pass and high-pass filters in a circuit. For low-pass filters, the higher the frequency became, the more that the output voltage was reduced. Low-pass filters only "pass" lower frequencies to the output with little attenuation [cut-off] and higher frequencies are significantly attenuated (i.e., not "passed"). The opposite is true for high-pass filters: the lower frequencies resulted in reduced output voltage. Higher frequencies are now "passed" to the output with little attenuation and lower frequencies are now "not-passed" and are more attenuated [cut-off]. Additionally, using the function generator as an input signal for the Arduino's Digital pin 11 created a tone for the speaker. The sweep button on the generator will be useful when a range of frequencies is to be swept across.

For the future, it may be useful to use a RC filter to filter out high frequencies or low frequencies, respectively, depending on the situation. Additionally, using the technique covered in Question 10 may become useful because some sensors may not report their measurements as analog or digital values, and instead as pulse lengths and duty cycles.

# References

Furman, B. (2014). ME 106 Lab Manual. San Jose, CA: San Jose State University.

Alciatore, D. G. & Histand, M. B. (2011). Introduction to Mechatronics and Measurement

Systems (4th Ed.). New York, NY: McGraw-Hill.

# Appendix A: Pitches.h Source Code

/******	* * * * * * * * * * *	*************
* Publ	ic Constar	nts
* * * * * * * *	********	 **********************************
#define	NOTE BO	31
#define	NOTE C1	33
#define	NOTE_CS1	35
#define	NOTE_CSI	37
#define	NOTE_DI	30
#define	NOIE_DSI	41
#define	NOIE_EI	41
#deline	NOIE_FI	44
#derine	NOTE_FSI	40
#deline	NOTE_GI	49
#derine	NOTE_GSI	52
#define	NOTE_AI	55
#define	NOTE_ASI	58
#define	NOTE_BI	62
#define	NOTE_C2	65
#define	NOTE_CS2	69
#define	NOTE_D2	73
#define	NOTE_DS2	78
#define	NOTE_E2	82
#define	NOTE_F2	87
#define	NOTE_FS2	93
#define	NOTE_G2	98
#define	NOTE_GS2	104
#define	NOTE_A2	110
#define	NOTE_AS2	117
#define	NOTE_B2	123
#define	NOTE_C3	131
#define	NOTE CS3	139
#define	NOTE D3	147
#define	NOTE DS3	156
#define	NOTE E3	165
#define	NOTE F3	175
#define	NOTE FS3	185
#define	NOTE G3	196
#define	NOTE GS3	208
#define	NOTE A3	220
#define	NOTE AS3	233
#define	NOTE B3	247
#define	NOTE C4	2.62
#define	NOTE CS4	277
#define	NOTE D4	2.94
#define	NOTE DS4	311
#define	NOTE E4	330
#define	NOTE F4	349
#define	NOTE FSA	370
#define	NOTE C4	392
#dofine	NOTE COA	A15
#define		110
#define	NOTE ACA	166
#define	NOTE DA	
#dettile	NOTE_D4	191 500
#aeiine	NOLE_C2	525

#define	NOTE_CS5	554
#define	NOTE_D5	587
#define	NOTE_DS5	622
#define	NOTE_E5	659
#define	NOTE_F5	698
#define	NOTE_FS5	740
#define	NOTE_G5	784
#define	NOTE_GS5	831
#define	NOTE_A5	880
#define	NOTE_AS5	932
#define	NOTE_B5	988
#define	NOTE_C6	1047
#define	NOTE_CS6	1109
#define	NOTE_D6	1175
#define	NOTE_DS6	1245
#define	NOTE_E6	1319
#define	NOTE_F6	1397
#define	NOTE_FS6	1480
#define	NOTE_G6	1568
#define	NOTE_GS6	1661
#define	NOTE_A6	1760
#define	NOTE_AS6	1865
#define	NOTE_B6	1976
#define	NOTE_C7	2093
#define	NOTE_CS7	2217
#define	NOTE_D7	2349
#define	NOTE_DS7	2489
#define	NOTE_E7	2637
#define	NOTE_F7	2794
#define	NOTE_FS7	2960
#define	NOTE_G7	3136
#define	NOTE_GS7	3322
#define	NOTE_A7	3520
#define	NOTE_AS7	3729
#define	NOTE_B7	3951
#define	NOTE_C8	4186
#define	NOTE_CS8	4435
#define	NOTE_D8	4699
#define	NOTE_DS8	4978

## **Appendix B: Source Code for Question 8**

```
/*
 * Hello. This program was developed by Patrick Barrera and Jimmy He for a
laboratory project in one of
* the upper division college courses at San Jose State University:
Mechanical Engineering 106:
 * Fundamentals of Mechatronics. This is Lab 3 "RC Filters and Basic Timer
Functionality", Question 8.
* The program generates a sweep tone on the speaker from a minimum of 100
Hz to a maximum of 15,000 Hz
 * in 5 seconds.
 */
#include "pitches.h"
// I) Variables
/* Pin Assignments: Declare and Initialize Input, Output, and Variables
*/
int hertz = 100;
// II) setup() is analogous to Constructors
/* Configure Pins. Set up Input and Output with the setup() function.
     pinMode(12, INPUT); function initializes pin 12 (Button0) to INPUT
 *
     pinMode(SPEAKER, OUTPUT); function initializes the SPEAKER (pin 5) to
OUTPUT
*
     pinMode(SW0, INPUT PULLUP); function initializes switch 0 (pin 12) to
INPUT PULLUP.
 *
       INPUT PULLUP initializes switch 0 to default at 0 Volts.
 *
     MyServo.attach(10); Attaches the servo on pin 10 to the servo object
 */
void setup()
{
 pinMode(11, OUTPUT);
}
// III) loop() is analogous to main()
/* Loop Forever
 */
void loop()
{
  for(hertz = 10; hertz <= 15000; hertz+=50)</pre>
  {
    //int noteDuration = 1000 / noteDurations[hertz];
    tone(11, hertz);
    //digitalWrite(11, HIGH);
    delay(15);
  }
```

```
delay(15);
for(hertz = 15000; hertz >= 10; hertz-=50)
{
    tone(11, hertz);
    //digitalWrite(11, HIGH);
    delay(15);
}
digitalWrite(11, LOW);
}
```

## **Appendix C: Source Code for Question 9**

```
/*
  * Hello. This program was developed by Patrick Barrera and Jimmy He for a
laboratory project in one of
  * the upper division college courses at San Jose State University:
Mechanical Engineering 106:
  * Fundamentals of Mechatronics. This is Lab 3 "RC Filters and Basic Timer
Functionality", Question 9.
  * The program generates a sweep tone on the speaker from a minimum of 100
Hz to a maximum of 15,000 Hz
  * in 5 seconds, times the sweep sequence, and prints it to the Serial
Monitor.
  */
 #include "pitches.h"
 // I) Variables
 /* Pin Assignments: Declare and Initialize Input, Output, and Variables
  */
 int hertz = 100;
 unsigned long time;
 // II) setup() is analogous to Constructors
 /* Configure Pins. Set up Input and Output with the setup() function.
      pinMode(12, INPUT); function initializes pin 12 (Button0) to INPUT
  *
       pinMode(SPEAKER, OUTPUT); function initializes the SPEAKER (pin 5) to
OUTPUT
  *
       pinMode(SW0, INPUT PULLUP); function initializes switch 0 (pin 12) to
INPUT PULLUP.
  *
          INPUT PULLUP initializes switch 0 to default at 0 Volts.
       MyServo.attach(10); Attaches the servo on pin 10 to the servo object
  */
 void setup()
 {
   pinMode(11, OUTPUT);
   // initialize serial communication at 9600 bits per second:
   Serial.begin(9600);
 }
 // III) loop() is analogous to main()
 /* Loop Forever
  */
 void loop()
 {
   time = millis();
   for(hertz = 10; hertz <= 15000; hertz+=50)</pre>
```

```
{
  //int noteDuration = 1000 / noteDurations[hertz];
  tone(11, hertz);
  //digitalWrite(11, HIGH);
  Serial.println(time);
  delay(15);
}
delay(15);
time = 0;
for(hertz = 15000; hertz >= 10; hertz-=50)
{
  tone(11, hertz);
 //digitalWrite(11, HIGH);
 delay(15);
}
digitalWrite(11, LOW);
```

```
}
```

#### **Appendix D: Source Code for Question 10**

```
/*
 * Hello. This program was developed by Patrick Barrera and Jimmy He for a
laboratory project in one of
* the upper division college courses at San Jose State University:
Mechanical Engineering 106:
* Fundamentals of Mechatronics. This is Lab 3 "RC Filters and Basic Timer
Functionality", Question 10.
* This program uses the pulseIn() function to time the length of the high
portion of the signal pulses
 * from the function generator and prints it out on the Serial Monitor. The
Arduino takes in an input
 * signal from the function generator via Digital pin 11 (and ground) on the
Arduino.
 */
#include "pitches.h"
// I) Variables
/* Pin Assignments: Declare and Initialize Input, Output, and Variables
*/
int hertz = 100;
unsigned long duration;
// II) setup() is analogous to Constructors
/* Configure Pins. Set up Input and Output with the setup() function.
     pinMode(12, INPUT); function initializes pin 12 (Button0) to INPUT
 *
     pinMode (SPEAKER, OUTPUT); function initializes the SPEAKER (pin 5) to
OUTPUT
*
     pinMode(SW0, INPUT PULLUP); function initializes switch 0 (pin 12) to
INPUT PULLUP.
 *
         INPUT PULLUP initializes switch 0 to default at 0 Volts.
 *
     MyServo.attach(10); Attaches the servo on pin 10 to the servo object
 */
void setup()
{
  // take in input signal from function generator
  pinMode(11, INPUT);
  // initialize serial communication at 9600 bits per second:
  Serial.begin(9600);
}
// III) loop() is analogous to main()
/* Loop Forever
*/
```

```
void loop()
{
   duration = pulseIn(11, HIGH);
   Serial.println(duration);
}
```