

Project_Progress

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1.1 Amplifier, Microphone circuit

Breadboard Connections: Vcc = +5V input: - microphone : 47E to 48E - connect mic to ground : 48A to Ground - connect mic to amplifier : 47C to 30H

amplifier - connect op-amp LM386 pin(1, 2[-in], 3[+in], 4[GND], 5[V_out], 6[Vcc], 7, 8) -> (26E, 27E, 28E, 29E, 29F, 28F, 27F, 26F) - 10 kOhm resistor : 30G to 34D - 10 kOhm resistor : 33H to Ground - 1 microF capacitor : 34B to 36B - 1 microF capacitor : 34C to 36C - 1 microF capacitor : 29I to 31I - 1 microF capacitor : 31J to 33J - 10 microF capacitor : 30F to 28C - 10 microF capacitor : 27G to 22G - connect 28I to Vcc - connect 34A to Vcc - connect 22F to Ground - connect 27A to Ground - connect 29A to Ground - connect 36A to Ground

amplifier output at op-amp pin5: connect 29J to 20J

positive clamper circuit to add 2.5V offset to amplifier output - diode : 14F to Ground - 1 kOhm resistor : 14G to Ground - 10 kOhm resistor : 14I to 4I - 100 microF capacitor : 20H to 24H

final output signal at 4J

LaunchPad Connections: - Connect LaunchPad 5V to Vcc - Connect LaunchPad GND to ground - Connect LaunchPad P6.0 to 4J

1.2 C Code

```
[ ]: #include <msp430f5529.h>
#include<stdio.h>

# define TXD BIT4           // TXD on P4.4
# define RXD BIT5           // RXD on P4.5
# define LED1 BIT0           // P1.0 LED
# define LED2 BIT7           // P4.7 LED
# define INPUT BIT0          // input from microphone on P6.0

unsigned int TXByte;

int main(void) {
    WDTCTL = WDTPW + WDTHOLD;    // stop watchdog timer
```

```

// configure ADC
ADC12CTL0 = ADC12SHT02 + ADC12ON; // Sampling time, ADC12 on
ADC12CTL1 = ADC12SHP; // sampling timer
ADC12CTL0 |= ADC12ENC; // ADC enable
P6SEL |= 0x01; // P6.0 allow ADC on pin 6.0

ADC12MCTL0 = ADC12INCH_0;
//selects which input results are
//stored in memory ADC12MEM0. Input
//one is selected on reset so this line is not needed
//Must be written before enabling conversions

ADC12CTL0 |= ADC12ENC; // ADC enable

/* Configure hardware UART */
UCA1CTL1 = UCSWRST; // Recommended to place USCI in reset first
P4SEL |= BIT4 + BIT5;
UCA1CTL1 |= UCSSEL_2; // Use SMCLK
UCA1BR0 = 109; // Set baud rate to 9600 with 1.048MHz clock
(Data Sheet 36.3.13)
UCA1BR1 = 0; // Set baud rate to 9600 with 1.048MHz clock
UCA1MCTL = UCBRS_2; // Modulation UCBRSx = 2
UCA1CTL1 &= ~UCSWRST; // Initialize USCI state machine
/* if we were going to receive, we would also:
   IE2 |= UCA1RXIE; // Enable USCI_A1 RX interrupt
*/

// output pin to computer
P4DIR |= TXD;
P4OUT |= TXD;

// LED for function testing
// P1DIR |= LED1;
// P1OUT |= LED1;

P4DIR |= LED2;
P4OUT &= ~LED2;

while (1) {
    ADC12CTL0 |= ADC12SC; // Start sampling
    while (ADC12CTL1 & ADC12BUSY); // while bit ADC12BUSY in register
    ADC12CTL1 is high wait

    while (! (UCA1IFG & UCTXIFG)); // wait for TX buffer to be ready for
    new data

```

```

        TXByte = ADC12MEM0/10;    // input voltage proportional to the volume
of the audio signal
        UCA1TXBUF = TXByte;    // Transmit TXByte;

//        P1OUT ^= LED1;        // toggle LED1
        P4OUT ^= LED2;        // toggle LED2

        _delay_cycles(10000); // wait for 20 milliseconds before repeating
    }
}

```

1.3 Python Code to receive data and save it into a csv file with time

```

[ ]: import matplotlib.pyplot as plt
import serial # for serial port
import numpy as np # for arrays, numerical processing
import time
import csv

#define the serial port
port = '/dev/tty.usbmodem112203'

audiodata = []
times = []
csv_savename = "audiodata3.csv"
create_plot = True # set true to create updating plot

# try:
#     import IPython
#     shell = IPython.get_ipython()
#     shell.enable_matplotlib(gui='qt')
# except:
#     print("Need to enter '%matplotlib qt' manually in console.")
#     pass

# initialize figures
if create_plot:
    plt.figure()
    plt.xlabel("Time (s)")
    plt.ylabel("ADC output")
    plt.title("Audio Signal ADC")
    # plt.ylim(0, 100)

try:

```

```

with serial.Serial(port,9600,timeout = 0.1) as ser:
    print(ser.name)
    print("Flushing serial...")
    ser.flushInput()
    ser.flushOutput()
    print("Flushed")
    start_time = time.time()
    while(1): #loop forever
        data = ser.read(1)
        if len(data) > 0: #was there a byte to read?
            print(ord(data))
            # get current time and save in list
            current_time = time.time()-start_time
            times.append(current_time)
            audiodata.append(ord(data))

            # if create_plot:
            #     plt.pause(0.0001)# wait for plot to update (INCREASE IF
↳PLOT ISN'T UPDATING PROPERLY)
            #     plt.clf() # clear figure
            #     plt.plot(times, audiodata)

#on ctrl-c (in console), save to CSV in specified location
except KeyboardInterrupt:
    create_plot = False
    print("Collection stopped - saving to CSV...")
    startcsv = time.time()
    with open(csv_savename, "w", newline='', encoding='utf-8') as f:
        writer = csv.writer(f)
        rows = zip(times, audiodata)
        for row in rows:
            writer.writerow(row)
    endcsv = time.time()
    print("Data saved to " + csv_savename)
    print("Took " + str(endcsv - startcsv) + " s")

```

1.4 Python Code to analyze the data from saved csv file

```

[ ]: import numpy as np
import matplotlib.pyplot as plt
import scipy as sp

# import data
def count_peaks(filename):
    print("processing file: {}".format(filename))

```

```

data = np.loadtxt(filename, delimiter=',')
time = data[:, 0]
power = data[:, 1]

# print(np.mean(power), np.max(power))
fs = 1/(time[1]-time[0]) # sampling frequency

# find peaks in smoothed signal
# peaks, props = sp.signal.find_peaks(power, distance=0.2*fs, height=np.
↪max(power[-500:]))
# find peaks in noisy signal using wavelet decomposition
# peaks = sp.signal.find_peaks_cwt(power, widths=np.arange(5, 15))
peaks, props = sp.signal.find_peaks(power, distance=40, height=np.
↪mean(power)+1*np.std(power))
print("time_arr, power_arr length", time.shape, power.shape)
print("peak array shape", peaks.shape)
print("fs=" ,fs)
print("peaks idx", peaks)

plt.figure(figsize=(12, 6))
plt.plot(time, power)
plt.scatter(time[peaks], power[peaks], marker='x', c='red', label="peaks")
plt.legend()
plt.xlabel("Time (s)")
plt.ylabel("ADC output")
plt.title("total peaks: {}".format(len(peaks)))
plt.show()

return len(peaks)

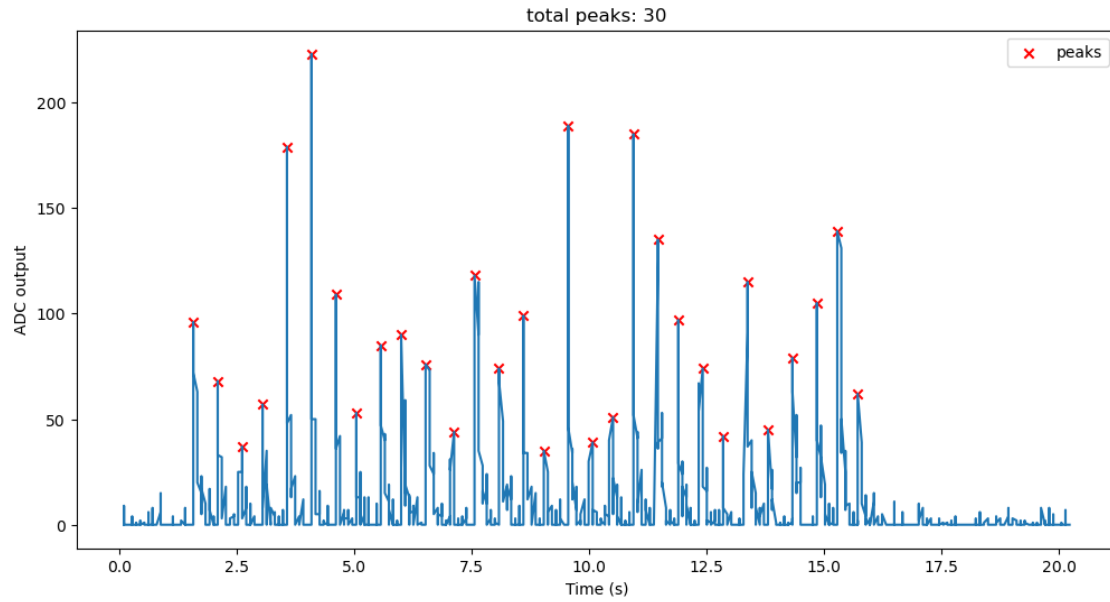
file1 = "audiodata_approx30_peaks.csv"
file2 = "audiodata_approx100_peaks.csv"
count_peaks(file1)
count_peaks(file2)

```

```

processing file: audiodata_approx30_peaks.csv
time_arr, power_arr length (2097,) (2097,)
peak array shape (30,)
fs= 53092.45569620253
peaks idx [ 160  210  264  313  363  417  472  520  571  620  674  729  782  834
  884  935  982 1035 1080 1129 1180 1227 1278 1326 1378 1430 1483 1535
1583 1628]

```



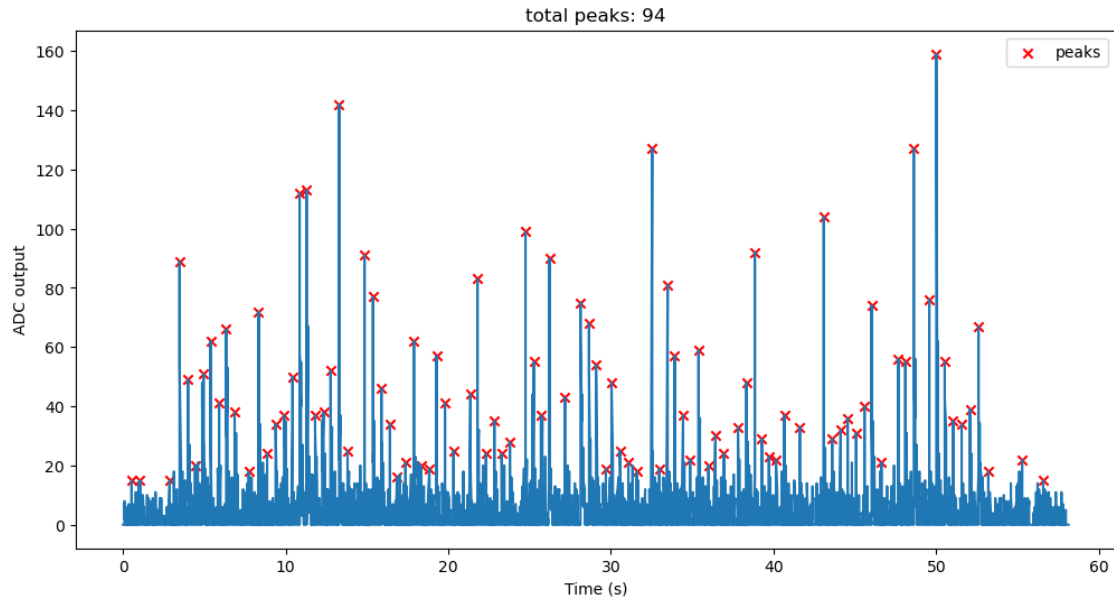
processing file: audiodata_approx100_peaks.csv

time_arr, power_arr length (6040,) (6040,)

peak array shape (94,)

fs= 10.704124132298897

```
peaks idx [ 51 101 289 355 407 458 507 552 605 656 710 798 862 913
 967 1026 1073 1121 1169 1219 1270 1318 1372 1428 1536 1589 1642 1698
1738 1799 1848 1897 1945 1997 2050 2101 2206 2257 2311 2360 2412 2461
2559 2611 2662 2712 2813 2915 2968 3015 3073 3114 3168 3220 3271 3367
3419 3466 3515 3565 3617 3668 3734 3774 3827 3924 3977 4025 4070 4121
4167 4216 4315 4468 4522 4573 4624 4675 4727 4777 4840 4943 4993 5046
5145 5193 5245 5294 5349 5402 5459 5520 5735 5876]
```



[]: 94

1.5 Progress summary

Hardware: done. The circuit are working properly, but sometime generates a lot of noise randomly, but everything goes back to normal after I connected it to the oscilloscope.

Software: I finished most of the code, and is trying to optimize the peak finding algorithm. The code for the real time plot in the heartbeat sensor doesn't seems to work for my project, it creates a huge delay in the plot.