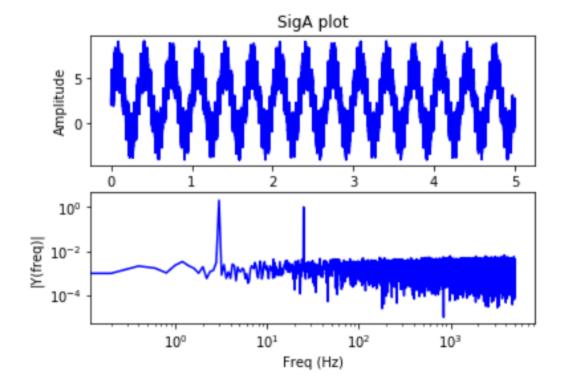
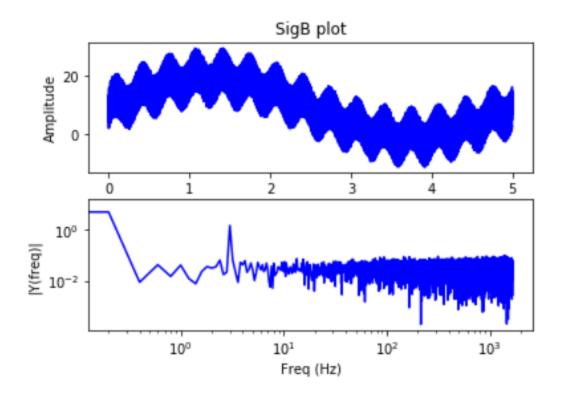
Note: Wasn't sure what signal was wanted for some of these plots. For example for sigB it appears to be the combination of 2 sin waves, the plot was smoothed so that both were visible. For sigD I used values which kept the oscillations and just smoothed them out. Code is also in the separate jupyter notebook.

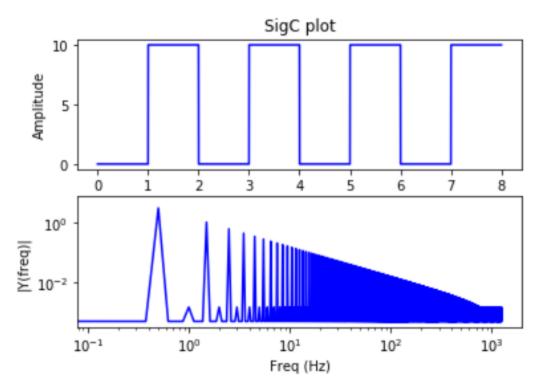
## Question 4:

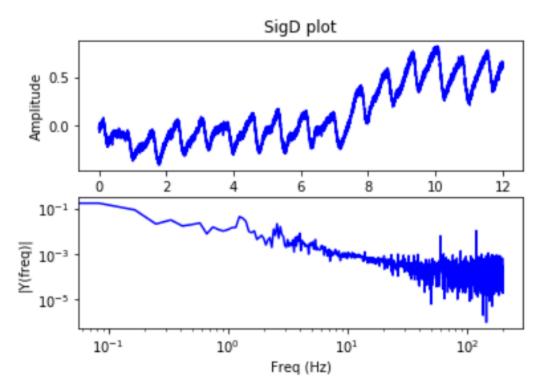
```
#auestion 4
import matplotlib.pyplot as plt
import numpy as np
import csv
t = [] # column 0
data1 = [] # column 1
data2 = [] # column 2
with open('sigA.csv') as f:
  # open the csv file
  reader = csv.reader(f)
  for row in reader:
     # read the rows 1 one by one
     t.append(float(row[0])) # leftmost column
     data1.append(float(row[1])) # second column
     #data2.append(float(row[2])) # third column
endtime=t[-1]
print(endtime)
nsamples=len(data1)
print(nsamples)
sam_rate=nsamples/endtime
sam_rate=round(sam_rate)
print(sam_rate)
Fs = sam rate # sample rate
Ts = 1.0/Fs; # sampling interval
ts = np.arange(0,t[-1],Ts) # time vector
y = data1 # the data to make the fft from
n = len(y) # length of the signal
k = np.arange(n)
T = n/Fs
frq = k/T # two sides frequency range
frq = frq[range(int(n/2))] # one side frequency range
Y = np.fft.fft(y)/n # fft computing and normalization
Y = Y[range(int(n/2))]
fig, (ax1, ax2) = plt.subplots(2, 1)
ax1.plot(t,y,'b')
ax1.set xlabel('Time')
ax1.set_ylabel('Amplitude')
ax2.loglog(frq,abs(Y),'b') # plotting the fft
ax2.set xlabel('Freq (Hz)')
ax2.set_ylabel('|Y(freq)|')
```

# plt.show()





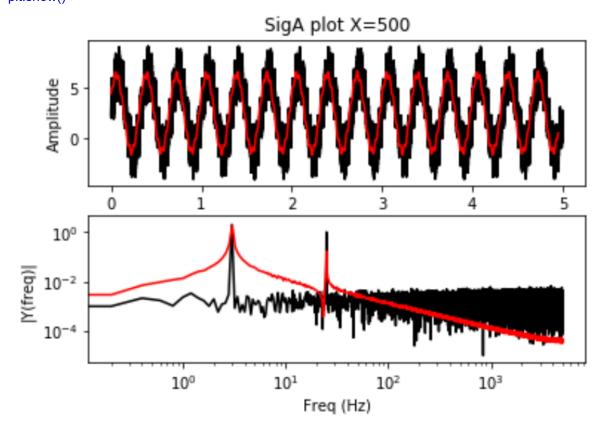


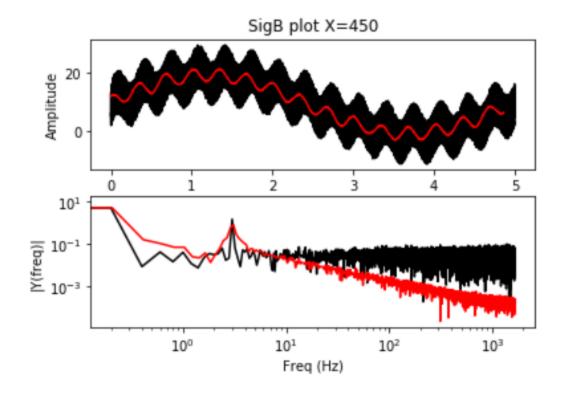


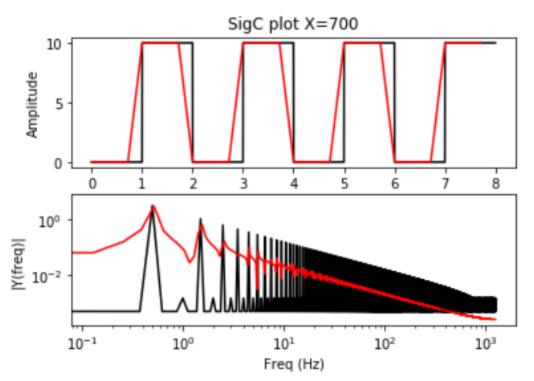
## **Question 5 MAF Transform:**

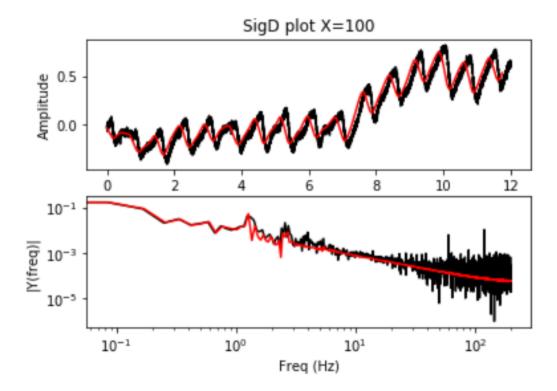
```
#question 5
import matplotlib.pyplot as plt
import numpy as np
import csv
t = [] # column 0
data1 = [] # column 1
data2 = [] # column 2
X=500;
newdata=[]#the averaged vector
newtime=[]#the shortened time vector
tempdata=0;#the temporary averaged value
with open('sigC.csv') as f:
  # open the csv file
  reader = csv.reader(f)
  for row in reader:
     # read the rows 1 one by one
     t.append(float(row[0])) # leftmost column
     data1.append(float(row[1])) # second column
     #data2.append(float(row[2])) # third column
  for i in range(0,(len(data1)-X)):
     tempdata=0
     for ii in range(0,X):
       tempdata=tempdata+data1[i+ii]
     newdata.append(tempdata/X)
     newtime.append(t[i])
```

```
#unfiltered data
endtime=t[-1]
nsamples=len(data1)
sam rate=nsamples/endtime
sam rate=round(sam rate)
Fs = sam rate # sample rate
Ts = 1.0/Fs; # sampling interval
ts = np.arange(0,t[-1],Ts) # time vector
y = data1 # the data to make the fft from
n = len(y) # length of the signal
k = np.arange(n)
T = n/Fs
frq = k/T # two sides frequency range
frq = frq[range(int(n/2))] # one side frequency range
Y = np.fft.fft(y)/n # fft computing and normalization
Y = Y[range(int(n/2))]
fig, (ax1, ax2) = plt.subplots(2, 1)
ax1.plot(t,y,color="black")
ax1.set xlabel('Time')
ax1.set_ylabel('Amplitude')
ax1.set title('Unfiltered sample')
ax2.loglog(frq,abs(Y),color="black") # plotting the fft
ax2.set xlabel('Freq (Hz)')
ax2.set_ylabel('|Y(freq)|')
plt.show()
y2=y
Y2=Y
frq2=frq
endtime=newtime[-1]
nsamples=len(newdata)
sam rate=nsamples/endtime
sam rate=round(sam rate)
Fs = sam rate # sample rate
Ts = 1.0/Fs; # sampling interval
ts = np.arange(0,newtime[-1],Ts) # time vector
y = newdata # the data to make the fft from
n = len(y) # length of the signal
k = np.arange(n)
T = n/Fs
frq = k/T # two sides frequency range
frq = frq[range(int(n/2))] # one side frequency range
Y = np.fft.fft(y)/n # fft computing and normalization
Y = Y[range(int(n/2))]
fig. (ax1, ax2) = plt.subplots(2, 1)
ax1.plot(t,y2,color="black")
ax1.plot(newtime,y,'r-')
ax1.set_xlabel('Time')
ax1.set_ylabel('Amplitude')
ax1.set_title('Filtered sample X='+str(X))
ax2.loglog(frq2,abs(Y2),color="black") # plotting the fft
```









## **Question 6: IIR filter**

for i in range(0,len(data1)):

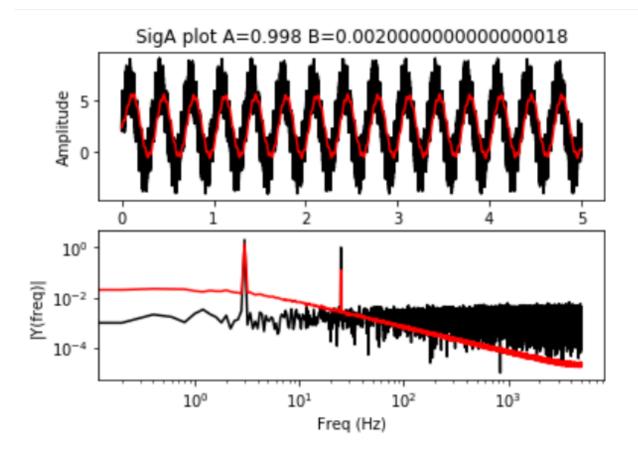
new\_avg=A\*new\_avg+B\*data1[i]

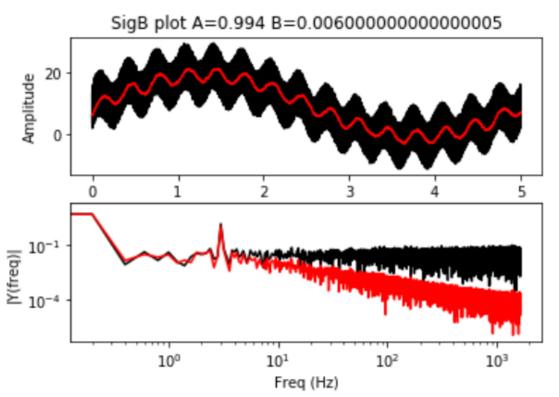
```
#question 6
import matplotlib.pyplot as plt
import numpy as np
import csv
t = [] # column 0
data1 = [] # column 1
data2 = [] # column 2
A=.95
B=1-A
with open('sigD.csv') as f:
  # open the csv file
  reader = csv.reader(f)
  for row in reader:
    # read the rows 1 one by one
    t.append(float(row[0])) # leftmost column
    data1.append(float(row[1])) # second column
    #data2.append(float(row[2])) # third column
endtime=t[-1]
nsamples=len(data1)
sam_rate=nsamples/endtime
sam_rate=round(sam_rate)
newdata=[]#the new vector
new avg=data1[0]
```

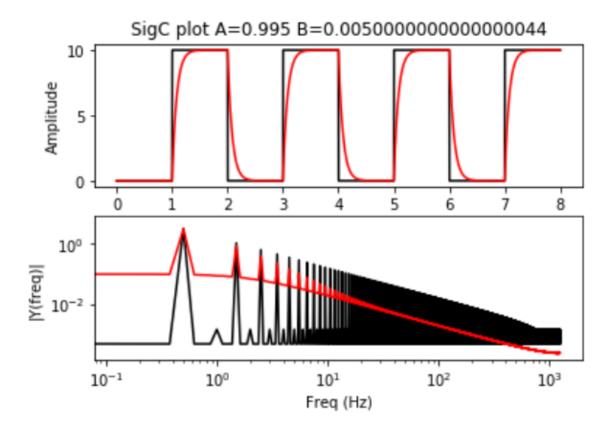
#### newdata.append(new\_avg)

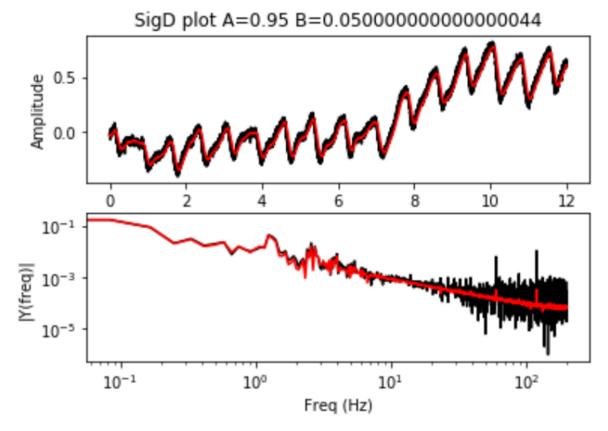
plt.show()

```
Fs = sam rate # sample rate
Ts = 1.0/Fs; # sampling interval
ts = np.arange(0,t[-1],Ts) # time vector
y = data1 # the data to make the fft from
n = len(y) # length of the signal
k = np.arange(n)
T = n/Fs
frq = k/T # two sides frequency range
frq = frq[range(int(n/2))] # one side frequency range
Y = np.fft.fft(y)/n # fft computing and normalization
Y = Y[range(int(n/2))]
Fs = sam rate # sample rate
Ts = 1.0/Fs; # sampling interval
ts = np.arange(0,t[-1],Ts) # time vector
y = newdata # the data to make the fft from
n = len(y) # length of the signal
k = np.arange(n)
T = n/Fs
frq2 = k/T # two sides frequency range
frq2= frq[range(int(n/2))] # one side frequency range
Y2 = np.fft.fft(y)/n # fft computing and normalization
Y2 = Y2[range(int(n/2))]
fig, (ax1, ax2) = plt.subplots(2, 1)
ax1.plot(t,data1,color="black")
ax1.plot(t,newdata,color="red")
ax1.set_xlabel('Time')
ax1.set_ylabel('Amplitude')
ax1.set_title('SigD plot A='+str(A)+" B="+str(B))
ax2.loglog(frq,abs(Y),color="black") # plotting the fft
ax2.loglog(frq2,abs(Y2),color="red") # plotting the fft
ax2.set xlabel('Freq (Hz)')
ax2.set_ylabel('|Y(freq)|')
```









#question 7 FIR Response (SIG A data loaded import matplotlib.pyplot as plt import numpy as np import csv

t = [] # column 0 data1 = [] # column 1 data2 = [] # column 2

newdata=[]#the averaged vector newtime=[]#the shortened time vector tempdata=0;#the temporary averaged value Constants=[] #The FIR contstants h = [

= [
0.00000000000000000000,
0.000009291614166548,
0.000050548523936133,
0.000082648554461720,
0.000000000000000000,
-0.000263718332932162,
-0.000598895948257723,
-0.000683824606726953,
-0.000158386273069903,
0.001027219811730566,
0.002318936748160005,
0.002640811218668270,
0.001013754286002316,
-0.002523719254668862,

```
-0.006325631253687232,
  -0.007561897310826583.
  -0.003831874832884219.
  0.004738477060165078,
  0.014323274838318484.
  0.018447375737723894,
  0.011433631732265579,
  -0.007231490196232454.
  -0.030231505621782617,
  -0.043749386035555130,
  -0.033153126834597833,
  0.009232303164279268,
  0.078287872718538265,
  0.155824099666343502.
  0.216872056812350694.
  0.240022308028222775,
  0.216872056812350694.
  0.155824099666343502,
  0.078287872718538265,
  0.009232303164279266.
  -0.033153126834597826,
  -0.043749386035555143,
  -0.030231505621782628,
  -0.007231490196232453,
  0.011433631732265581.
  0.018447375737723908,
  0.014323274838318490,
  0.004738477060165081.
  -0.003831874832884219,
  -0.007561897310826589,
  -0.006325631253687240,
  -0.002523719254668863,
  0.001013754286002317,
  0.002640811218668273,
  0.002318936748160003.
  0.001027219811730566.
  -0.000158386273069903.
  -0.000683824606726953,
  -0.000598895948257722.
  -0.000263718332932162,
  0.000082648554461720,
  0.000050548523936133,
  0.000009291614166548,
  X=len(h)
with open('sigD.csv') as f:
  # open the csv file
  reader = csv.reader(f)
  for row in reader:
    # read the rows 1 one by one
    t.append(float(row[0])) # leftmost column
    data1.append(float(row[1])) # second column
```

1

```
#data2.append(float(row[2])) # third column
  for i in range(0,(len(data1)-X)):
    tempdata=0
    for ii in range(0.X):
       tempdata=tempdata+ data1[i+ii]*h[ii]
    newdata.append(tempdata)
    newtime.append(t[i])
#unfiltered data
endtime=t[-1]
nsamples=len(data1)
sam rate=nsamples/endtime
sam rate=round(sam rate)
print(sam rate)
Fs = sam rate # sample rate
Ts = 1.0/Fs; # sampling interval
ts = np.arange(0,t[-1],Ts) # time vector
y = data1 # the data to make the fft from
n = len(y) # length of the signal
k = np.arange(n)
T = n/Fs
frq = k/T # two sides frequency range
frq = frq[range(int(n/2))] # one side frequency range
Y = np.fft.fft(y)/n # fft computing and normalization
Y = Y[range(int(n/2))]
fig, (ax1, ax2) = plt.subplots(2, 1)
ax1.plot(t,y,color="black")
ax1.set xlabel('Time')
ax1.set ylabel('Amplitude')
ax1.set title('Unfiltered sample')
ax2.loglog(frq,abs(Y),color="black") # plotting the fft
ax2.set_xlabel('Freq (Hz)')
ax2.set ylabel('|Y(freq)|')
plt.show()
y2=y
Y2=Y
frq2=frq
endtime=newtime[-1]
nsamples=len(newdata)
sam rate=nsamples/endtime
sam rate=round(sam rate)
Fs = sam rate # sample rate
Ts = 1.0/Fs; # sampling interval
ts = np.arange(0,newtime[-1],Ts) # time vector
y = newdata # the data to make the fft from
n = len(y) # length of the signal
k = np.arange(n)
T = n/Fs
frq = k/T # two sides frequency range
frq = frq[range(int(n/2))] # one side frequency range
```

```
Y = np.fft.fft(y)/n # fft computing and normalization
Y = Y[range(int(n/2))]

fig, (ax1, ax2) = plt.subplots(2, 1)
ax1.plot(t,y2,color="black")
ax1.plot(newtime,y,'r-')
ax1.set_xlabel('Time')
ax1.set_ylabel('Amplitude')
ax1.set_title('FIR Coeffeicents='+str(X))
ax2.loglog(frq2,abs(Y2),color="black") # plotting the fft
ax2.loglog(frq,abs(Y),'r-') # plotting the fft
ax2.set_xlabel('Freq (Hz)')
ax2.set_ylabel('|Y(freq)|')
plt.show()
```

NOTE: For SigA the website does not let me go below a 100Hz cutoff frequency. This is about the best I can get under these conditions.

