ckground Fourier Series Distinguishing Noise filtering

Research Methods

Robert E Simpson

Singapore University of Technology & Design robert_simpson@sutd.edu.sg

November 9, 2018



Useful Noes

Rob[72:144]=0 Sets the index range 72 to 144 of list Rob to 0. numpy.fft.fft computes the fast Fourier transform numpy.fft.ifft computes the inverse fast Fourier transform



Case Problem 1: Fourier Series

1. Write a function to plot the amplitude spectrum for the signal described by:

$$f(t) = \sum_{k}^{n} (n-k)\sin(2\pi kt) \tag{1}$$

- 2. Plot the waveform for this signal for n = 20 and k = 10 between t = 0 and t = 4 ensuring that the signal is not undersampled.
- 3. Redo the plot, only this time under and over sample the signal at $f_s/2$ and $2f_s$ respectively. What do you notice?
- 4. Use numpy's built in FFT function to compute and then plot the magnitude of the frequency spectrum for the signal. Make sure that the signal is not under sampled. What do you notice?

```
def w9cp1(t, n, k):
    out=0
    for kk in range(k,n):
        out=out+ (n-kk)*np.sin(2*np.pi*kk*t)
    return out
Ns=600 #number of samples
t=np.linspace(0,4, Ns)
Maxtime = 4.0
time_step=Maxtime/Ns
FreqStep = 1./(Maxtime)
print "max freq is", 1/time_step
freq = []
for i in range(Ns):
    freq . append(i*FreqStep)
ft=w9cp1(t,20,10)
```



```
w=np.fft.fft(ft) #compute forward Fourier transform
plt.figure()
plt.plot(t, ft, lw=1)
mag = []
for line in w:
    mag.append(np.linalg.norm(line)) #compute the magnitude
plt.figure()
plt.plot(freq, mag)
plt.xlabel('Freq (Hz)')
plt.ylabel('FT Magnitude')
plt.xlim(0,50)
```

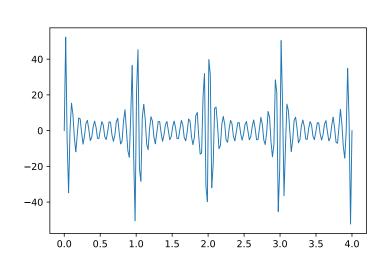


2.

The maximum frequency component in the signal is k=20. Therefore we need at least 40 points ($f_s=>40$ Hz) along one wavelength to reconstruct the signal.

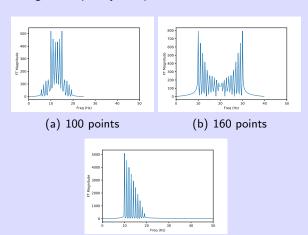
We are asked to plot the wave from 0 to 4. When k is 1 this would be four wavelengths, but k goes up to 20. Therefore, we need to multiply this by 40 (because $f_s =>$ 40 Hz). So the minimum number of data points is: $4 \times 40 = 160$. But let's add more points to ensure that we capture all the features of the signal, so we make a wave with 200 points and scale the x-axis from 0 to 4.







3. When we over sample the signal shape doesn't change too much, but when we under sample there is big change in the signal because we have removed the higher frequency components.





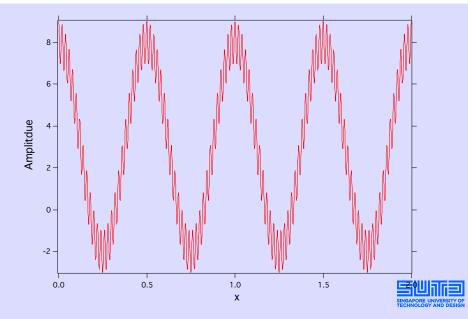


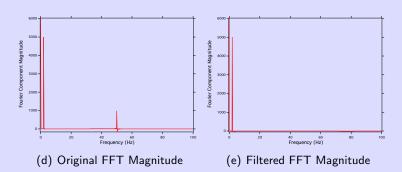
4. As we would expect, the signal is built from sinusoids at frequencies ranging from 10 Hz, 11 Hz, ..., 20 Hz.



- Plot the wave: $5\cos((2\pi)2x) + \cos((2\pi)50x) + 3$ between x=0 and x=2.
- Plot the amplitude of the Fourier components that are present in the signal
- Now create a new wave of the complex FFT signal.
- Filter out the 50 Hz signal in frequency space by seting the amplitudes of the frequency spectrum to zero for frequencies close to 50 Hz (I created a 'window' function to do this).
- Now synthesise the signal from the modified/filtered Fourier component spectrum.
- Plot and compare the 'filtered' and original signals.
- Repeat the process and filter out the low frequency signals.

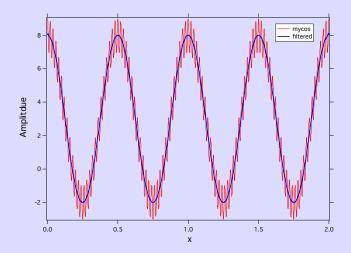








The filtered signal now does not have the high frequency noise.





```
Ns=1001 #number of samples
x=np.linspace(0,2,Ns)
y=5*np.cos((2*np.pi)*2*x) + np.cos((2*np.pi)*50*x)
Maxtime=2.
time_step=Maxtime/Ns
FreqStep = 1./(Maxtime)
print "freq step", FreqStep
print "max freq is", 1./time_step
freq = []
for i in range(Ns):
    freq . append ( i * FreqStep )
```



```
w=np.fft.fft(y)
plt.figure()
plt.plot(x,y)

plt.figure()
plt.plot(freq,w)
plt.xlim(0,60)
```



```
win = []
def window(freq, spec, f0, Delta_f):
                                      #Create a widow function
    indx0=freq.index(f0)
    indx02=len(freq)-indx0
    print indx0, indx02
    freqstep=freq[1]-freq[0]
    print freqstep
    Deltaindex=Delta_f/freqstep
    print "Delta index", Deltaindex
    IndexStartDelete0=indx0-(Deltaindex/2)
    IndexEndDelete0=indx0+(Deltaindex/2)
    IndexStartDelete02=indx02 - (Deltaindex /2)
    IndexEndDelete02=indx02+(Deltaindex/2)
    F_spec=spec
    F_spec[int(IndexStartDelete0): int(IndexEndDelete0)]=0
    F_spec[int(IndexStartDelete02):int(IndexEndDelete02)]=0
    return F_spec
```

```
filtered50=window(freq,w, 50,10)
plt.figure()
plt.plot(x,y)
plt.figure()
plt.plot(freq, filtered50)
plt.xlim(0,60)
F_spec=np.fft.ifft(filtered50)
plt.figure()
plt.plot(X, F_spec)
```



Often a signal can be distinguished from noise by analysing the frequency components in the signal. E.g. The signal may contain low frequencies whereas the noise may be at a high frequency.



ackground Fourier Series Distinguishing **Noise** filtering

Case Problem 3

- Set the scale to go from 0 to 10 seconds
- Set a sinusoidal wave with an amplitude of 10 and a frequency of 2 Hz
- Add random white noise to the signal (numpy.random.normal)
- Now you have a noisy signal, perform Fourier filtering by setting all Fourier components to zero except for the component with the largest contribution to the signal.
- Compare the synthesised signal with the original signal.

