# lab\_symfseries

September 17, 2019

## 1 Fourier series

This workbook explores the Fourier series representation of signals. It is shown how symbolic methods can be used to calculate series coefficients, and the accuracy of the time-domain reconstruction for different numbers of coefficients is also considered.

## 1.1 Background

Any signal x(t) that is periodic with a period T can be written in a Fourier series form

$$x(t) = \sum_{k=-\infty}^{\infty} c_k e^{jk\omega_0 t}$$

with  $\omega_0 = 2\pi/T = \pi$  radians per second. The coefficients satisfy

$$c_k = \frac{1}{T} \int_{-T/2}^{T/2} x(t) e^{-jk\omega_0 t} dt.$$

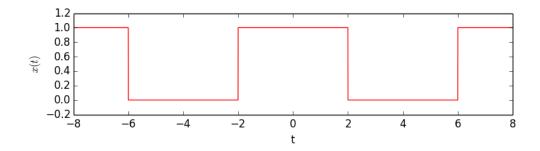
For real signals with  $x(t) = x^*(t)$  one can show that  $c_{-k} = c_k^*$ . Writing in polar form  $c_k = |c_k|e^{j\angle c_k}$  the series becomes

$$x(t) = c_0 + \sum_{k=1}^{\infty} (c_k e^{jk\omega_0 t} + c_{-k} e^{-jk\omega_0 t}) = c_0 + \sum_{k=1}^{\infty} |c_k| (e^{jk\omega_0 t} e^{j\angle c_k} + e^{-jk\omega_0 t} e^{-j\angle c_k}) = c_0 + \sum_{k=1}^{\infty} 2|c_k| \cos(\omega_0 t + \angle c_k)$$

The coefficient  $c_k$  corresponds to a complex exponential with frequency  $k\omega_0$ . We call the component of the signal with frequency  $k\omega_0$  the kth harmonic. The first harmonic is also called the fundamental.

## 1.2 Signal definition and analysis

The following periodic signal is considered throughout this workbook:



The signal is has period T=8, and  $\omega_0=2\pi/8=\pi/4$ . We can find the corresponding Fourier series coefficients:

$$c_k = \frac{1}{T} \int_{-T/2}^{T/2} x(t) e^{-jk\omega_0 t} dt = \frac{1}{T} \int_{-2}^2 e^{-jk\omega_0 t} dt = \frac{1}{jk\omega_0 T} [-e^{-jk\omega_0 t}]_{t=-2}^2 = \frac{2}{k\omega_0 T} \sin(2k\omega_0).$$

Additionally, the DC coefficient  $c_0 = 4/8$ .

#### 1.3 Fourier series reconstruction

For a given set of coefficients  $c_k$  we want to be able to plot the corresponding x(t). The function defined in the cell below takes a set of Fourier series coefficients for a real signal ckv and a fundamental frequency omega0, and calculates reconstructed values xv at time instants tv.

```
xv = ckv[0]*np.ones(tv.shape);
#tv.shape returns the dimensions of the matrix (or, in this case, array) tv.
#np.ones creates an array of the input size populated with 1's
plt.figure(1)
plt.plot(tv,xv)
for k in range(1,len(ckv)):
    kh = 2*np.abs(ckv[k])*np.cos(k*omegaO*tv + np.angle(ckv[k])); #create kth harr
    plt.plot(tv,kh);

    xv = xv + kh; #add kth harmonic to x
plt.show()
    return(np.real(xv));
# end def
```

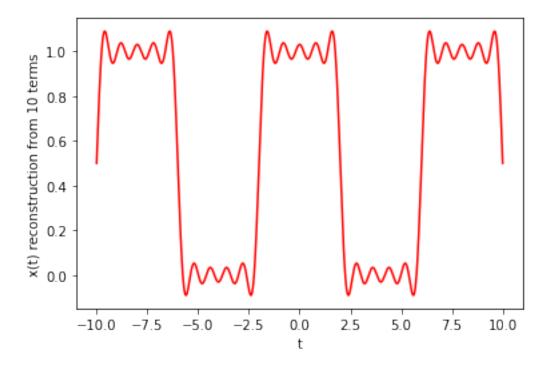
The cell below uses the derived expression for the coefficients of the signal and stores them in the vector ckv. The kth element of ckv contains the coefficient  $c_k$ .

```
In [3]: # Fourier series coefficients for rectangular pulse train
    T = 8; # period
    N = 10; # maximum number of terms
    omega0 = 2*np.pi/T;
    ckv = np.zeros(N+1, dtype=np.complex64);
    #np.zeros is like np.ones but with 0's

for k in range(1,N+1): ckv[k] = 2/(k*omega0*T)*np.sin(2*k*omega0);
    ckv[0] = 4/8;
```

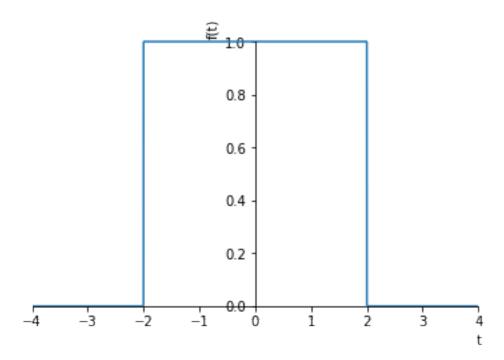
We can use the fsrrec function to find the time-domain representation of the signal x(t) using a finite number of terms in the reconstruction. The code below does this and plots the result.

```
In [4]: # Reconstruct from series representation and plot
    tv = np.linspace(-10,10,10000);
    xv = fsrrec(ckv,omega0,tv);
    #xv = fsrrec_plots(ckv,omega0,tv);
    plt.figure(2)
    plt.plot(tv,xv,'r');
    plt.xlabel('t'); plt.ylabel('x(t) reconstruction from ' + str(N) + ' terms');
```



## 1.4 Finding the coefficients using symbolic math

This section will show how we can calculate the Fourier series coefficients of a signal using symbolic manipulation. The first thing to do is symbolically define the signal. The Piecewise function lets you define a signal over different pieces.



The following cell defines the symbolic integral for computing the coefficients.

```
In [6]: Ts, k, w0 = sp.symbols('Ts k w0');
    w0 = 2*sp.pi/Ts;
    expt = sp.exp(-1j*k*w0*t);
    cke = 1/Ts*sp.integrate(x*expt, (t, -Ts/2, Ts/2));
    #cke = sp.integrate(x*expt, (t, -sp.oo, T/2)) - sp.integrate(x*expt, (t, -sp.oo, -T/2));
    ck = cke.subs(Ts,T).doit(); # set value for period and evaluate
    ck
```

#### Out[6]:

$$\begin{cases}
-\frac{4.0ie^{0.5i\pi k}}{\pi k} + \frac{4.0ie^{-0.5i\pi k}}{\pi k} & \text{for } k > -\infty \land k < \infty \land k \neq 0 \\
4 & \text{otherwise}
\end{cases}$$

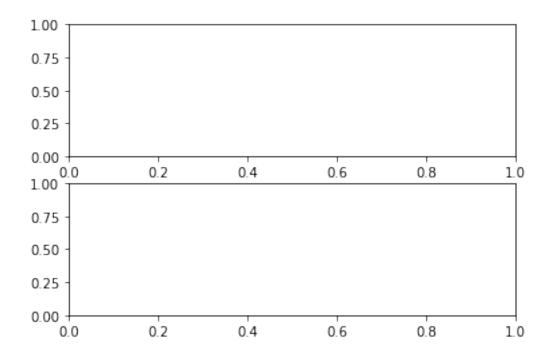
We now define a vector kv of coefficients of interest, and populate corresponding elements of ckv with the coefficient values.

```
for i in range(len(kv)):
           ki = kv[i];
           ckvs[i] = ck.subs({k:ki}).evalf();
        ckvs
Out[7]: array([ 0.
                         +0.j, 0.03536776+0.j, -0.
                                                          +0.j, -0.04547284+0.j,
                         +0.j, 0.06366198+0.j, -0.
                                                          +0.j, -0.10610329+0.j,
               0.
                         +0.j, 0.31830987+0.j, 0.5
                                                          +0.j, 0.31830987+0.j,
               0.
                         +0.j, -0.10610329+0.j, -0.
                                                          +0.j, 0.06366198+0.j,
               0.
                         +0.j, -0.04547284+0.j, -0.
                                                          +0.j, 0.03536776+0.j,
                         +0.j], dtype=complex64)
               0.
```

Now we can plot the frequency-domain representation of the signal x(t) by displaying the value of  $c_k$  for each value k of interest. Since  $c_k$  can in general be complex we need two plots: one for magnitude and one for phase.

```
In [8]: fh, ax = plt.subplots(2);
        ax[0].stem(kv, np.abs(ckvs), c='g'); ax[0].set_ylabel(r'$|c_k|$');
        ax[1].stem(kv, np.angle(ckvs), c='g'); ax[1].set_ylabel(r'$\angle c_k$');
        plt.xlabel('$k$');
        TypeError
                                                  Traceback (most recent call last)
        <ipython-input-8-07e1feb2452a> in <module>
          1 fh, ax = plt.subplots(2);
   ----> 2 ax[0].stem(kv, np.abs(ckvs), c='g'); ax[0].set_ylabel(r'$|c_k|$');
          3 ax[1].stem(kv, np.angle(ckvs), c='g'); ax[1].set_ylabel(r'$\angle c_k$');
          4 plt.xlabel('$k$');
        ~/anaconda3/lib/python3.7/site-packages/matplotlib/__init__.py in inner(ax, data, *arg
                def inner(ax, *args, data=None, **kwargs):
       1587
                    if data is None:
       1588
   -> 1589
                        return func(ax, *map(sanitize_sequence, args), **kwargs)
       1590
       1591
                    bound = new_sig.bind(ax, *args, **kwargs)
```

TypeError: stem() got an unexpected keyword argument 'c'



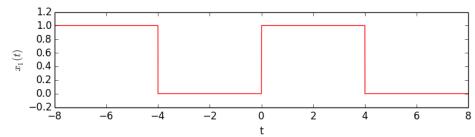
We could also have created a lambda function from the symbolic expression. This function takes an array of values for k and calculates  $c_k$  directly. Note though that at least on my version of Python lam\_ck(0) generates a divide-by-zero error.

With numerical values for the coefficients, obtained via symbolic computation, we can plot the partial sum for the time-domain reconstruction as before. Recall that our fsrrec function only takes the coefficients for nonnegative index values.

### 2 Tasks

These tasks involve writing code, or modifying existing code, to meet the objectives described.

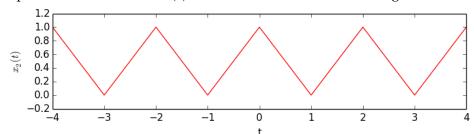
1. Find Fourier and plot the series frequency-domain representation for the signal  $x_1(t)$ below range k  $-8, \dots, 8$ : over the



Do this using

both symbolic processing and by evaluating the integral for the coefficients by hand. Compare the result with that displayed earlier for x(t). Note that the signals are related in time by  $x_1(t) = x(t-2)$ . You should observe that shifting a signal only changes the phase in the frequency domain, while the magnitude remains unchanged.

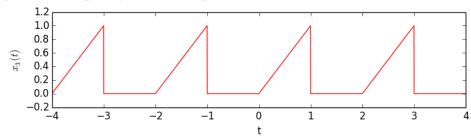
• Use symbolic processing find the frequency-domain to and plot of representation the k  $x_2(t)$ below range  $-8, \dots, 8$ : over



Also plot the

reconstruction over the range t = -4 to t = 4 using only components up to and including the 5th harmonic. You should find that as k increases the magnitude of the coefficients in this case falls off much faster than those of x(t). This is because  $x_2(t)$  is smoother (it is at least continuous, while x(t) is discontinuous). The reconstruction is therefore also more accurate with a smaller number of terms.

• Find and plot the frequency-domain representation of  $x_3(t)$  below over the range k =



-8, ..., 8.

In []: