# A Basic Example of Python

```
#####################
import numpy as np
from scipy import signal
import matplotlib.pyplot as plt
from matplotlib.ticker import MaxNLocator
import ipywidgets as widgets
from ipywidgets import interact, interactive, fixed, interact manual
from IPython.display import display
######################
rng = np.random.default rng(seed=42) # Meaning of Life
### OPSK Constellation Plotter ###
## Signal Generation ##
num symbols = 1000
x int = rng.integers(\frac{0}{4}, num_symbols) # 0 to 3 random symbol set
x 	ext{ degrees} = x 	ext{ int*} \frac{360}{4.0} + \frac{45}{45} # 45, 135, 225, 315 degrees because
I don't like complex math
x radians = x degrees*np.pi/180.0 # sin() and cos() takes in radians
x symbols = np.cos(x radians) + 1j*np.sin(x radians) # Produce QPSK
complex symbols
## Add Noise and Plot ##
@interact(awgn=(0.0, 1.0, 0.01), phase=(0.0, 1.0, 0.01))
def signal with noise(awgn=0.0, phase=0.0):
   ## Noise Generation ##
   # AWGN with unity power
   n = (rng.integers(num symbols) +
1j*rng.integers(num symbols))/np.sqrt(2)
   noise power = awgn
```

```
# Phase Noise
phase_noise = rng.integers(len(x_symbols)) * phase

# Add noise to signal
    recv = (x_symbols * np.exp(lj*phase_noise)) + (n *
np.sqrt(noise_power))

plt.plot(np.real(recv), np.imag(recv), '.')
plt.grid(True)
plt.show()

## Alternative method to decorator - toggle comments with decorator to use
# interactive_plot = widgets.interactive(signal_with_noise)
# output = interactive_plot.children[-1]
# interactive_plot

{"model_id":"dde7b837209346ffb3c52fab04e73b18","version_major":2,"version_minor":0}
```

# **Background**

### Digital Signals

Digital signals use symbols to represent one or more bits of information.

Generaly, we want to transmit the most data with the least spectrum.

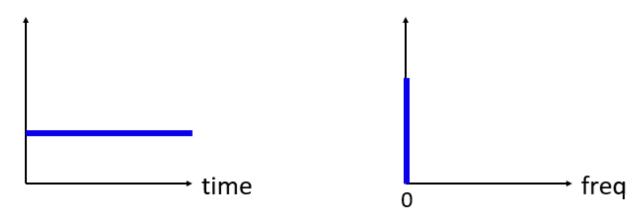
The frequency domain tells us how much spectrum/bandwidth our signal will use.

We usually want to maxamize throughput and minimize spectrum use.

Why can't we just transmit binary data at a faster speed?

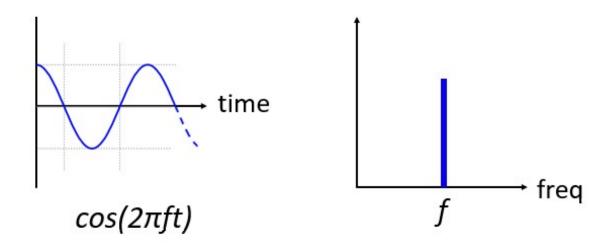
# Time-Frequency Pairs

## Constant Signal



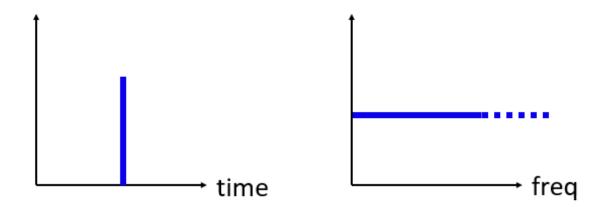
Because there is no frequency, in the frequency domain we have a spike at 0 Hz ("DC").

### Sinusoidal Wave



*Key Point*: The signal has a single frequency, which is why we see a single peak(impulse) in the frequency domain.

#### Unit Impulse



An impulse in the time domain is a horizontal line (all frequencies) in the frequency domain.

Note that the signal theoretically contains every frequency, although it would have to be infinently short in the time domain for this to occur.

Key Point: Quick changes in the time domain result in many frequencies occurring.

### **Fourier Properties**

Time Scaling Property

Scaling in the time domain causes inverse scaling in the frequency domain.

$$F(x(at)) = \frac{1}{|a|} X(j\frac{\omega}{a})$$

Transmitting 1's and 0's faster will increase the bandwidth of our signal and create undesireable frequency components.

#### Modulation

Our goal: to maximize "spectral efficiency" in units of bits/sec/Hz.

We do this with modulation.

Modify the properties of a sinusoid(the carrier) to transmit information.

- Amplitude
- Phase
- Frequency

Using schemes such as:

- ASK
- PSK
- FSK
- QAM

The spectral characteristics (frequency domain) of the baseband symbols do not change when we modulate a carrier.

Carrier modulation just shifts the baseband up in frequency while the shape stays the same.

The amount of bandwidth we use stays the same.

# Inter-Symbol-Interference (ISI)

### **Definition**

ISI occurs when a symbol interferes with subsequent symbols.

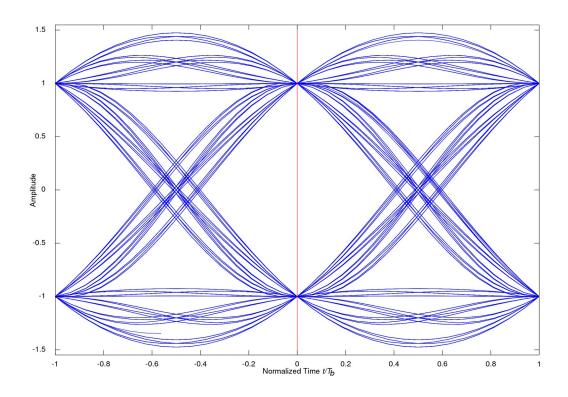
This causes distortion similar to noise and potentially introduces error in the receiver decision device.

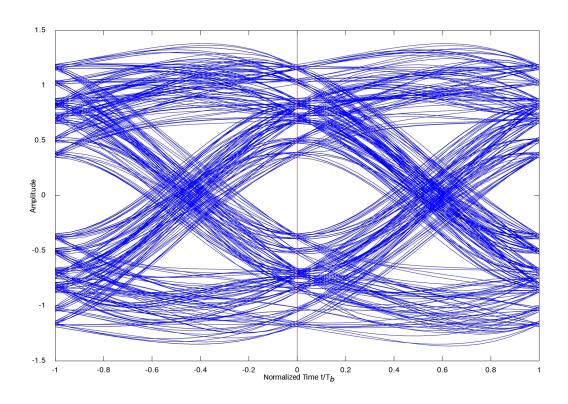
Successive symbols appear to "blur" or "bleed" together.

#### Causes

- Bandlimited Channel
  - Channel Frequency Response Characteristics
  - Artificial Restrictions (spectrum allocation)
- Multipath Propogation
  - Reflection
  - Refraction
  - Atmospheric Ducting
  - Ionospheric Reflection

# BPSK Multipath Eye Diagram





## Nyquist ISI Criterion

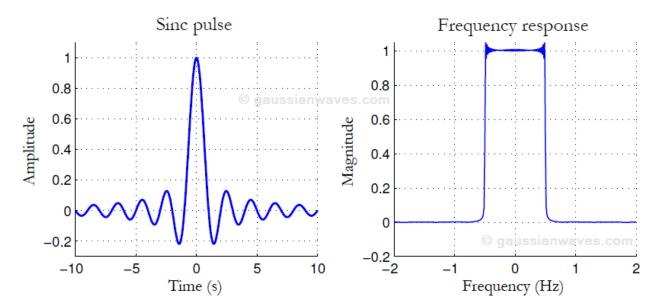
To achieve zero intersymbol interference, samples must have only one non-zero value at each sampling instant.

##The Rectangular Pulse

The simplest signal that avoids ISI is the rectangular pulse with width  $T_{sym}=1/F_{sym}$  but it consumes infinite bandwidth. This is because its Fourier transform (the sinc function) extends infinitely on either side of the spectrum.

### The Sinc Pulse

The signal that avoids ISI with the least amount of bandwidth is the sinc pulse with bandwidth  $F_{\mathrm{sym}}/2$ 



The major drawback of the sinc function is the slow decay rate, which can lead to ISI between symbols that are far apart.

The infinite impulse response makes it practically unrealizable.

### Solutions

- Error Correcting Codes
- Adaptive Equalization
- Symbol Guard Periods
- Receiver Sequence Detector

# **Basic Pulse Shaping Using BPSK**

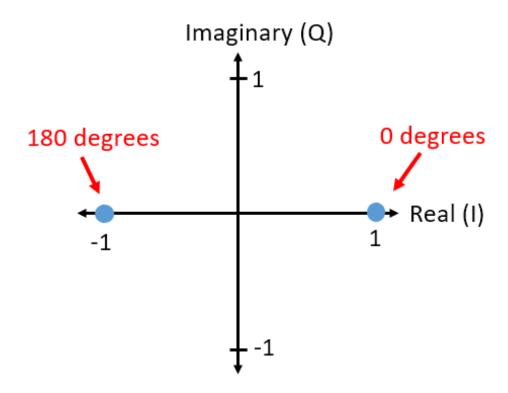
### **BPSK Fundamentals**

Using 1 bit per symbol is equivalent to transmitting square pulses.

BPSK is a square wave consisting of 1's and 0's.

There are two phase levels:

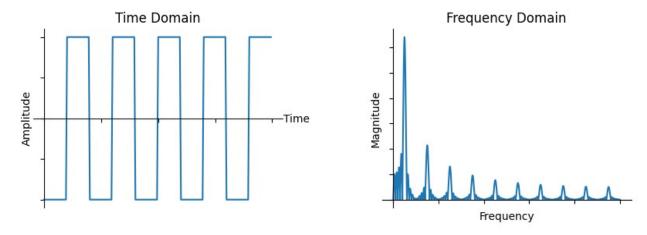
- 0° for binary 1
- $180^{\circ}$  for binary 0



But square pulses are not efficient because they use an excess amount of spectrum.

### Square wave Fourier Transform

```
#########################
## Signal Generation ##
#############################
y = np.repeat(np.arange(10) % 2, 20) * 2 - 1 # Create a repeating
square wave pattern
y = y.astype(float)
Y = np.abs(np.fft.rfft(y, 10000)) # Perform real-valued DFFT
#################
## Basic Plot ##
##################
fig, (ax1, ax2) = plt.subplots(1, 2, figsize=(10, 3))
plt.subplots adjust(wspace=0.4)
ax1.plot(y)
ax1.text(len(y) + 10, -0.05, 'Time')
ax1.set ylabel("Amplitude")
ax1.title.set text('Time Domain')
ax2.plot(Y)
ax2.set xlabel("Frequency")
ax2.set ylabel("Magnitude")
ax2.title.set text('Frequency Domain')
##################
## Fancy Format ##
###################
# set the x-spine
ax1.spines['left'].set_position('zero')
ax2.spines['left'].set position('zero')
# turn off the right spine/ticks
ax1.spines['right'].set_color('none')
ax1.yaxis.tick left()
ax2.spines['right'].set_color('none')
ax2.yaxis.tick left()
# set the v-spine
ax1.spines['bottom'].set position('zero')
ax2.spines['bottom'].set position('zero')
# turn off the top spine/ticks
ax1.spines['top'].set color('none')
ax1.xaxis.tick bottom()
ax2.spines['top'].set color('none')
ax2.xaxis.tick bottom()
```

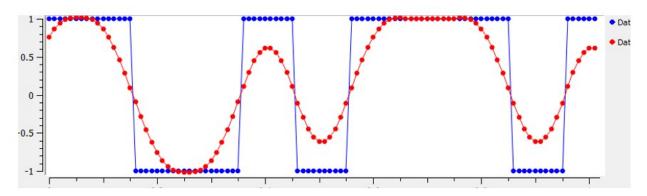


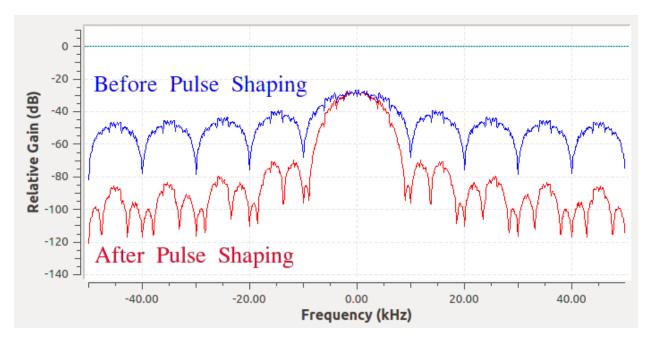
We want to "shape" the blocky-looking symbols so that they take up less bandwidth in the frequency domain.

How?

### The Solution

Use a lowpass filter to discard the higher frequency components of our symbols.





The sidelobes are ~30 dB lower after pulse shaping.

The main lobe is narrower, so less spectrum is used for the same amount of bits per second.

# **Pulse Shaping Theory**

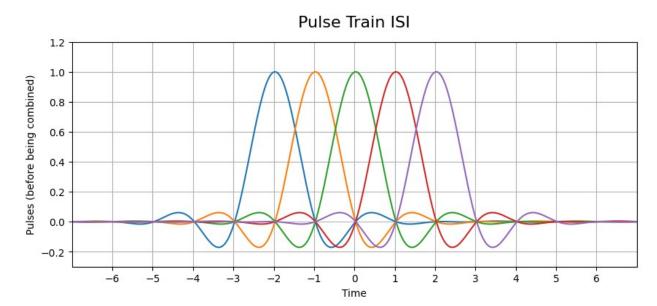
### Filters and Symbol Timing

When we apply a pulse-shaping filter, it elongates the pulse in the time domain (in order to condense it in frequency).

This causes adjacent symbols to begin to overlap one another.

This is okay if all pulses but one add up to zero at every multiple of our symbol period T.

```
t = np.arange(num taps//-2, num taps//2) # remember it's not inclusive
of final number
h = np.sinc(t/Ts) * np.cos(np.pi*beta*t/Ts) / (1 - (2*beta*t/Ts)**2)
## Basic Plot ##
N = 2000
fig, (ax) = plt.subplots(1, 1, figsize=(10, 4))
for delay in np.arange(-2,3):
    x = np.zeros(N)
    x[N//2 + delay*sps] = 1
    x 	ext{ shaped} = 	ext{np.convolve}(x, h, 'same')
    t = (np.arange(N) - N/2)/sps
    ax.plot(t, x shaped, '-')
## Plot Settings ##
ax.axis([-7, 7, -0.3, 1.2])
ax.set xlabel("Time")
ax.set_ylabel("Pulses (before being combined)")
ax.text(s="Pulse Train ISI", x=0,y=1.3, fontsize=16,
fontweight='roman', horizontalalignment='center')
#ax.xaxis.set major locator(MaxNLocator(integer=True))
major ticks = np.arange(-6, 7)
ax.set xticks(major ticks)
plt.grid(True)
plt.show()
```



At every interval T, one pulse peaks while the other pulses are at 0.

We want the receiver to sample the signal at the perfect time (at our pulse peaks), so it must synchronize with the symbol period (somehow).

We only care about symbol alignment at the receiver when we perform our sampling.

#### Matched Filters

#### **Basic Concept**

We (normally) want to low pass filter at the transmitter to reduce the amount of spectrum we use.

We also want to low pass filter at the receiver to elimate as much adjacent noise/interference next to the signal as possible.

Theoretically, the optimal linear filter for maximizing the SNR in the presence of AWGN is to use the same filter at both the Tx and Rx.

The receiver correlates the received signal with the known template signal. The template signal is essentially the pulses the transmitter sends, irrespective of the phase/amplitude shifts applied to them. Filtering is performed via convolution, and, in this case, correlation.

In matched filtering, we convolve the unknown signal with a conjugated time-reversed version of the template.

#### Filter Splitting

So we split the pulse shaping filter between the transmit and receive sides.

### Filter Design

#### Requirements:

- Reduce our signal bandwidth (to use less spectrum)
- All pulses except one should sum to zero at every symbol interval
- Split the filter into TX and RX portions

# **Common Pulse-Shaping Filters**

Common pulse-shaping filters include:

Sinc filter

- Raised-cosine filter
- Root raised-cosine filter
- Gaussian filter

Note that the sinc filter is equivalent to the raised-cosine filter when  $\beta = 0$ . The sinc filter is an ideal filter as there is no transition region.

#### Raised-Cosine Filter

The most popular pulse-shaping filter

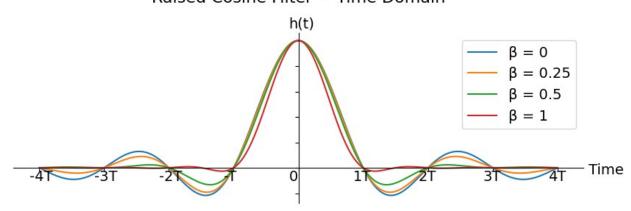
```
################
## Time Plot ##
################
fig, (ax1) = plt.subplots(1, 1, figsize=(10, 3))
 plt.subplots adjust(wspace=0.4)
sps = 100
t = np.linspace(-4 * sps, 4 * sps, 1000)
leq = []
 for beta in [0, 0.25, 0.5, 1]:
                h = np.sinc(t / sps) * np.cos(np.pi * beta * t / sps) / (1 - (2 * sps) / (1 + (2 * sps) / (2 * sps) / (1 + (2 * sps) / (1 + (2 * sps) / (1 + (2 * sps) / (2 * sps) / (1 + (2 * sps) / (2 * sps) / (1 + (2 * sps) / (2 * sps) / (1 + (2 * sps) / (2 * sps) / (1 + (2 * sps) / 
beta * t / sps) ** 2)
                ax1.plot(t, h)
                leg.append('\beta = ' + str(beta))
plt.legend(leg, fontsize=14)
ax1.text(max(t) * 1.12, -0.05, 'Time', fontsize=14)
ax1.text(-4 * sps - 15, -0.1, '-4T', fontsize=14)
ax1.text(-3 * sps - 15, -0.1, '-3T', fontsize=14)
ax1.text(-2 * sps - 15, -0.1, '-2T', fontsize=14)
ax1.text(-1 * sps - 15, -0.1, '-T', fontsize=14)
ax1.text(0 * sps - 15, -0.1, '0', fontsize=14)
ax1.text(1 * sps - 15, -0.1, '1T', fontsize=14)
ax1.text(2 * sps - 15, -0.1, '2T', fontsize=14)
ax1.text(3 * sps - 15, -0.1, '3T', fontsize=14)
ax1.text(4 * sps - 15, -0.1, '4T', fontsize=14)
ax1.text(-15, 1.1, 'h(t)', fontsize=14)
# set the x-spine (see below for more info on `set position`)
ax1.spines['left'].set_position('zero')
```

```
# turn off the right spine/ticks
ax1.spines['right'].set color('none')
ax1.yaxis.tick left()
# set the v-spine
ax1.spines['bottom'].set position('zero')
# turn off the top spine/ticks
ax1.spines['top'].set color('none')
ax1.xaxis.tick bottom()
# Turn off tick numbering/labels
ax1.set xticklabels([])
ax1.set yticklabels([])
ax1.text(s='Raised Cosine Filter \sim Time Domain', x=0, y=1.3,
fontsize=16, fontweight='roman', horizontalalignment='center')
plt.show()
#########################
## Frequency Plot ##
#######################
fig, (ax1) = plt.subplots(1, 1, figsize=(10, 3))
plt.subplots adjust(wspace=0.4)
sps = 100
leg = []
for beta in [0.01, 0.25, 0.5, 1]:
    f = np.linspace(-1.5 / sps, 1.5 / sps, 1000)
    H = []
    for fi in f:
        if np.abs(fi) \ll (1 - beta) / 2 / sps:
            H.append(1)
        elif (np.abs(fi) \le (1 + beta) / 2 / sps) and (np.abs(fi) > (1 + beta) / 2 / sps)
- beta) / 2 / sps):
            H.append(0.5 * (1 + np.cos(np.pi * sps / beta *
(np.abs(fi) - (1 - beta) / (2 * sps)))))
        else:
            H.append(0)
    ax1.plot(f, H)
    if beta == 0.01:
        beta = 0
    leq.append('\beta = ' + str(beta))
plt.legend(leg, fontsize=14)
ax1.text(max(f) * 1.12, -0.05, 'Freq', fontsize=14)
ax1.text(-1.5 / sps - 0.0015, -0.1, '-3/2T', fontsize=14)
ax1.text(-1 / sps - 0.0015, -0.1, '-1/T', fontsize=14)
```

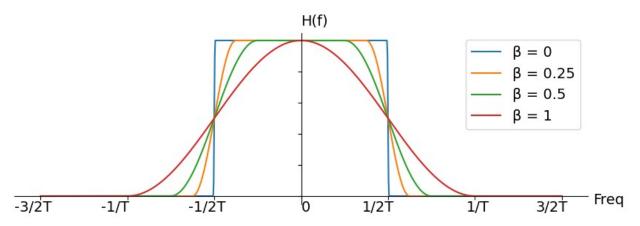
```
ax1.text(-0.5 / sps - 0.0015, -0.1, '-1/2T', fontsize=14)
ax1.text(0, -0.1, '0', fontsize=14)
ax1.text(1.5 / sps - 0.0015, -0.1, '3/2T', fontsize=14)

ax1.text(1 / sps - 0.0005, -0.1, '1/T', fontsize=14)
ax1.text(0.5 / sps - 0.0015, -0.1, '1/2T', fontsize=14)
ax1.text(0, 1.1, 'H(f)', fontsize=14)
# set the x-spine (see below for more info on `set position`)
ax1.spines['left'].set_position('zero')
# turn off the right spine/ticks
ax1.spines['right'].set color('none')
ax1.yaxis.tick_left()
# set the y-spine
ax1.spines['bottom'].set position('zero')
# turn off the top spine/ticks
ax1.spines['top'].set color('none')
ax1.xaxis.tick bottom()
# Turn off tick numbering/labels
ax1.set xticklabels([])
ax1.set_yticklabels([])
ax1.text(s='Raised Cosine Filter ~ Frequency Domain', x=0, y=1.3,
fontsize=16, fontweight='roman', horizontalalignment='center')
plt.show()
```

#### Raised Cosine Filter ~ Time Domain



#### Raised Cosine Filter ~ Frequency Domain



The  $\beta$  parameter is the only parameter for the raised-cosine filter, and it determines how quickly the filter tapers off in the time domain, which will be inversely proportional with how quickly it tapers off in frequency.

The reason it's called the raised-cosine filter is because the frequency domain when  $\beta$  = 1 is a half-cycle of a cosine wave, raised up to sit on the x-axis.

```
@interact(beta=(0.0,1.0,0.01),n_taps=(1, 300, 1))
def rrc_filter(beta = 0.35, n_taps = 101):
    Ts = sps # Assume sample rate is 1 Hz, so sample period is 1, so
*symbol* period is 8
    t = np.arange(n_taps) - (n_taps-1)//2
    h = 1/Ts*np.sinc(t/Ts) * np.cos(np.pi*beta*t/Ts) / (1 -
(2*beta*t/Ts)**2)
    plt.figure(1)
    plt.plot(t, h, '.')
    plt.plot(t, h, alpha=0.5)
    plt.title("Raised-Cosine Filter")
    plt.grid(True)
    plt.show()

{"model_id": "73bdb5af12824e0bb92e08b3efced8e1", "version_major":2, "version_minor":0}
```

## Square-Root Raised-Cosine Filter

The equally split filter portion of the TX/RX raised-cosine filter.

$$h(t) = \begin{cases} \frac{1}{T_s} \left( 1 + \beta (\frac{4}{\pi} - 1) \right) & t = 0 \\ \frac{\beta}{T_s \sqrt{2}} \left[ \left( 1 + \frac{2}{\pi} \right) \sin \left( \frac{\pi}{4\beta} \right) + \left( 1 - \frac{2}{\pi} \right) \cos \left( \frac{\pi}{4\beta} \right) \right] & t = \pm \frac{T_s}{4\beta} \\ \frac{1}{T_s} \frac{\sin \left[ \pi \frac{t}{T_s} \left( 1 - \beta \right) \right] + 4\beta \frac{t}{T_s} \cos \left[ \pi \frac{t}{T_s} \left( 1 + \beta \right) \right]}{\pi \frac{t}{T_s} \left[ 1 - \left( 4\beta \frac{t}{T_s} \right)^2 \right]} & \text{otherwise} \end{cases}$$

#### Raised-Cosine Roll-Off Factor

 $\beta$  is called the "roll-off" factor or "excess bandwidth."

It determines how fast (in the time domain) the filter rolls off to zero.

It is a number between 0 and 1.

More taps are required to lower  $\beta$ .

The lower the roll-off, the more compact the frequency of the signal for a given symbol rate.

Bandwidth can be approximated for a given symbol rate and roll-off factor as:

 $BW = R_s(\beta + 1)$ , where  $R_s$  is the symbol rate in Hz.

# Advanced BPSK Pulse Shaping

BPSK involves transmissitng 1's and -1's with the "Q" portion equal to zero, so we only need to plot the in-phase portion.

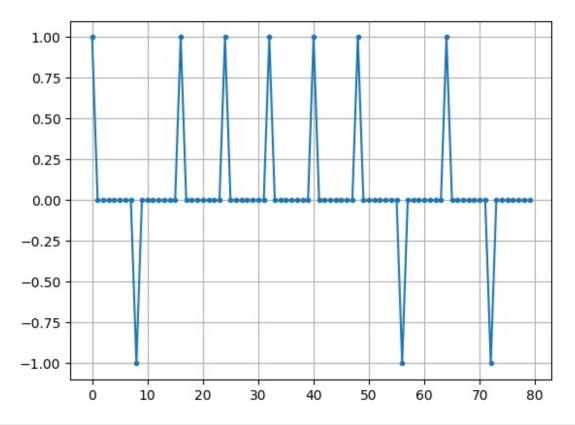
This is equivalent to bipolar 2-ASK, as a 180 $\square$ ° phase shift is equivalent to multiplying the sinusoid by -1.

### The Impulse Train

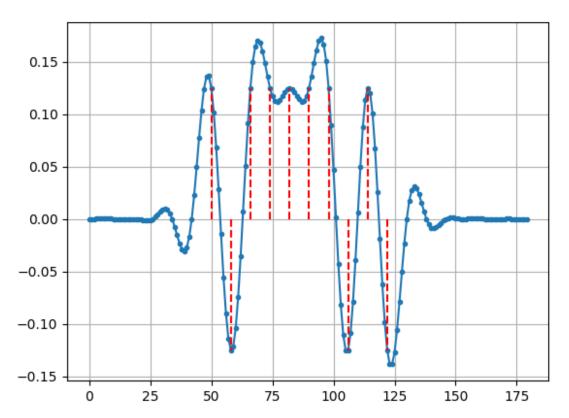
An Example Generated Setup:

```
bits: [0, 0, 1, 1, 1, 1, 0, 0, 1, 1]
BPSK symbols: [-1, -1, 1, 1, 1, -1, -1, 1, 1]
Applying 8 samples per symbol: [-1, 0, 0, 0, 0, 0, 0, 0, -1, 0, 0, 0, 0, 0, 0, 0, 0, ...]
```

```
###########################
### Impulse Response ###
##############################
num symbols = 10
sps = 8
# Transmission data of 1's and 0's
bits = rng.integers(0, 2, num symbols)
x = np.array([])
for bit in bits:
    pulse = np.zeros(sps)
    pulse[0] = bit*2-1 # set the first value to either a 1 or -1
    x = np.concatenate((x, pulse)) # add the 8 samples to the signal
plt.figure(0)
plt.plot(x, '.-')
plt.grid(True)
plt.show()
```



```
num taps = 101
beta = 0.35
Ts = sps # Assume sample rate is 1 Hz, so sample period is 1, so
*symbol* period is 8
t = np.arange(num taps) - (num taps-1)//2
h = \frac{1}{Ts*np.sinc(t/Ts)} * np.cos(np.pi*beta*t/Ts) / (1 -
(2*beta*t/Ts)**2)
# Filter our signal, in order to apply the pulse shaping
x 	ext{ shaped} = np.convolve(x, h)
plt.figure(1)
# Plot filtered signaal with sample points
plt.plot(x shaped, '.-')
# Mark the Ts intervals
for i in range(num_symbols):
    plt.plot([i*sps+num_taps//2,i*sps+num_taps//2], [0,
x shaped[i*sps+num taps//2]], '--', color = 'red')
plt.grid(True)
plt.show()
```



# BPSK in GNU Radio

