Sygnals and Systems - Dr. Akhavan

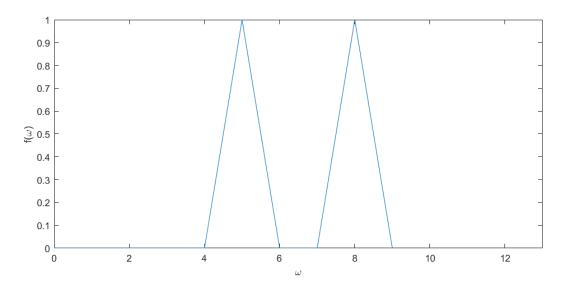
CA5 - Matin Bazrafshan

Part 0

if we consider a signal with $f_s = 20\,\mathrm{Hz}$, $T = 1\,s$ and N = 100 the frequency resolution will be $\delta_s = \frac{f_s}{N} = 1$, so we can not detect picks with distance less than 1.

for example signal $x_1(t) = e^{2\pi j(5t)} + e^{2\pi j(8t)}$ have the picks on 5 and 8 respectively as the fourier transform of it is $\mathcal{F}(e^{2\pi j(5t)} + e^{2\pi j(8t)}) = \delta(\omega - 5) + \delta(\omega - 8)$. the distance between two pulse is 3 so it will be shown well.

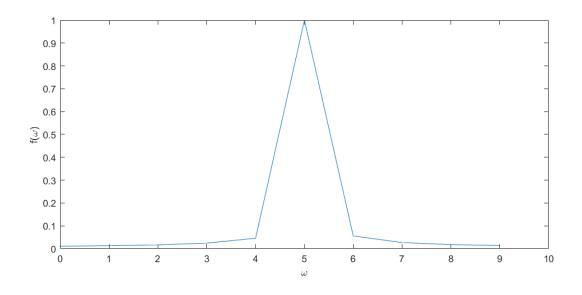
```
clearvars;
fs = 20;
ts = 1 / fs;
tstart = 0;
tend = 1;
t = tstart : ts : tend - ts;
T = tend - tstart;
N = T * fs;
f = -fs/2 : fs/N : fs/2 - fs/N;
x1 = exp(1i * 2 * pi * 5 * t) + exp(1i * 2 * pi * 8 * t);
y1 = fftshift(fft(x1));
y1 = y1 / max(y1);
figure('Position', [0 0 900 400]);
plot(f,abs(y1));
xlim([0 13]);
xlabel('\omega');
ylabel('f(\omega)');
```



but in signal $x_2(t) = e^{2\pi j(5.1t)} + e^{2\pi j(5.1t)}$ we have $\mathcal{F}(e^{2\pi j(5t)} + e^{2\pi j(8t)}) = \delta(\omega - 5) + \delta(\omega - 5.1)$ so the distance is 0.1 and can not be detected.

```
x2 = exp(1i * 2 * pi * 5 * t) + exp(1i * 2 * pi * 5.1 * t);
y2 = fftshift(fft(x2));
y2 = y2 / max(y2);

figure('Position', [0 0 900 400]);
plot(f,abs(y2));
xlim([0 10]);
xlabel('\omega');
ylabel('f(\omega)');
```



Part 1

1.1

first we plot $x_1(t) = \cos(10\pi t)$ with parameters:

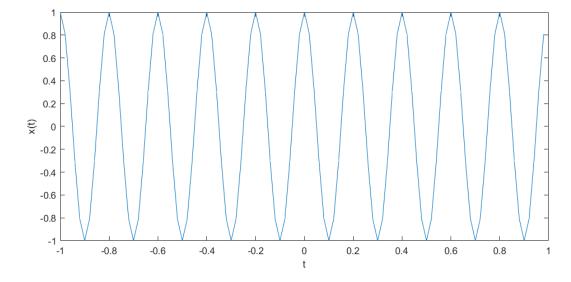
```
• f_s = 50
```

•
$$t = (-1, 1)$$

- T = 2
- $N = T \times f_s$

```
clearvars;
fs = 50;
ts = 1 / fs;
tstart = -1;
tend = 1;
t = tstart : ts : tend - ts;
T = tend - tstart;
N = T * fs;
f = -fs/2 : fs/N : fs/2 - fs/N;

figure('Position', [0 0 900 400]);
x1 = cos(10 * pi * t);
plot(t,x1);
xlabel('t');
ylabel('x(t)')
```



first we need to calculate the fourier transform:

•
$$x_1(t) = \cos(10\pi t) \rightarrow \mathcal{F}(x_1(t)) = \pi(\delta(\omega - 10\pi) + \delta(\omega + 10\pi))$$

after that we normalize the output by dividing it to the maximum absolute value of it:

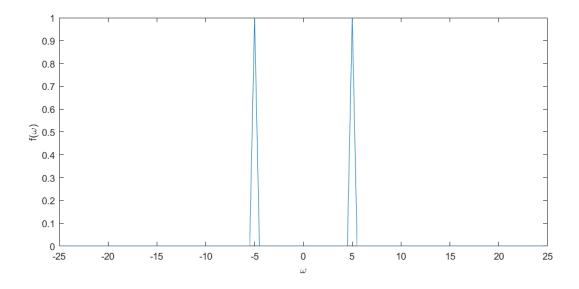
$$\hat{x}_1(\omega) = \frac{\hat{x}_1(\omega)}{\max(|\hat{x}_1(\omega)|)}$$

then we consider the $f = \frac{\omega}{2\pi}$ so:

•
$$\hat{x}_1(f) = \delta(f-5) + \delta(f+5)$$

```
y1 = fftshift(fft(x1));
y1 = y1 / max(abs(y1));

figure('Position', [0 0 900 400]);
plot(f,abs(y1));
xlabel('\omega');
ylabel('f(\omega)')
```



1.2

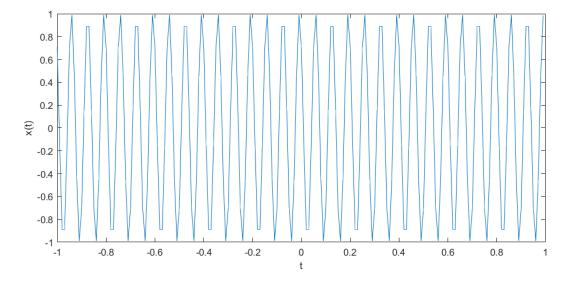
first we plot $x_2(t) = \cos\left(30\pi t + \frac{\pi}{4}\right)$ with parameters:

- $f_s = 100$
- t = (0, 1)
- T = 1
- $N = T \times f_s$

```
clearvars;
fs = 100;
ts = 1 / fs;
tstart = -1;
tend = 1;
```

```
t = tstart : ts : tend - ts;
T = tend - tstart;
N = T * fs;
f = -fs/2 : fs/N : fs/2 - fs/N;

figure('Position', [0 0 900 400]);
x2 = cos(30 * pi * t + pi / 4);
plot(t,x2);
xlabel('t');
ylabel('x(t)')
```



first we need to calculate the fourier transform:

•
$$x_2(t) = \cos\left(30\pi t + \frac{\pi}{4}\right) \rightarrow \mathcal{F}(x_1(t)) = \pi \left(e^{\frac{j\pi\omega}{4}}\delta(\omega - 30\pi) + e^{-\frac{j\pi\omega}{4}}\delta(\omega - 30\pi)\right)$$

after that we normalize the output by dividing it to the maximum absolute value of it:

$$\hat{x}_2(\omega) = \frac{\hat{x}_1(\omega)}{\max(|\hat{x}_1(\omega)|)}$$

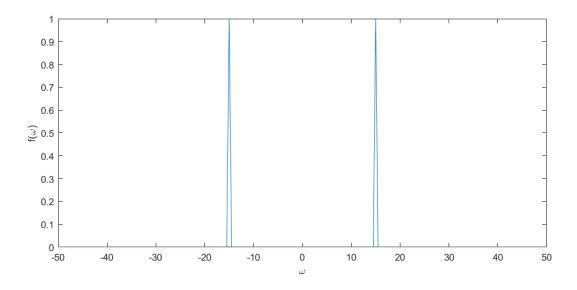
then we consider the $f = \frac{\omega}{2\pi}$ so:

•
$$\hat{x}_2(f) = e^{j\frac{\pi}{4}} \delta(f - 15) + e^{-j\frac{\pi}{4}} \delta(f + 15)$$

We expect two picks on ± 15 , as we plot $|\hat{x}_2(f)|$

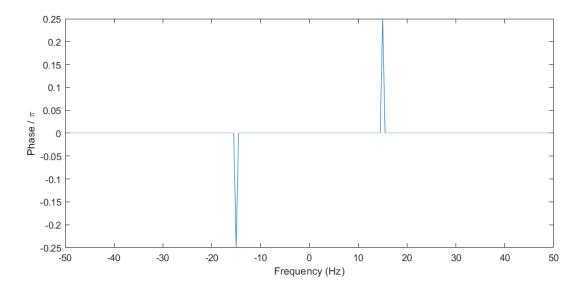
```
y2 = fftshift(fft(x2));
y2 = y2 / max(abs(y2));
```

```
figure('Position', [0 0 900 400]);
plot(f,abs(y2));
xlabel('\omega');
ylabel('f(\omega)')
```



And in plotting phase of $\hat{x}_2(f)=e^{\frac{j\pi}{4}}\delta(f-15)+e^{-\frac{j\pi}{4}}\delta(f+15)$, we consider y-axis ticks is π , so there should be a $-\frac{1}{4}$ on -15 and a $\frac{1}{4}$ on 15.

```
threshold = 1e-6;
y2(abs(y2) < threshold) = 0;
theta = angle(y2);
plot(f,theta/pi);
xlabel('Frequency (Hz)')
ylabel('Phase / \pi')</pre>
```



Part 2

2.1 Creating Mapset

```
clearvars;
mapset = create_mapset()

mapset = 2×32 cell
'a' 'b' 'c' 'd' 'e' 'f' 'g...
'00000' '00001' '00010' '00010' '00100' '00101' '0
```

2.2 - 2.3 Encoding message

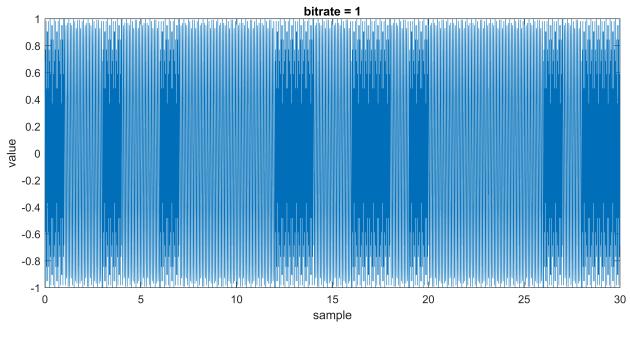
To encode message we need to find a proper f to encode bitstrings. Consider bitrate = n so there is 2^n possibilities for our bitstrings. So we divide our valid frequency range by 2^n . Valide range for frequency is $(0,\frac{f_s}{2}]$, because sine function has two symmetric pulses, there is no need to use negative frequencies. so each distance(we call them step) between two used frequencies is $f_{\text{step}} = \frac{f_s}{2}$, and initial frequency is $\frac{f_{\text{step}}}{2}$. so $f = f_{\text{init}} + i \times f_{\text{step}}$ where $i \in [0, 2^n - 1]$.

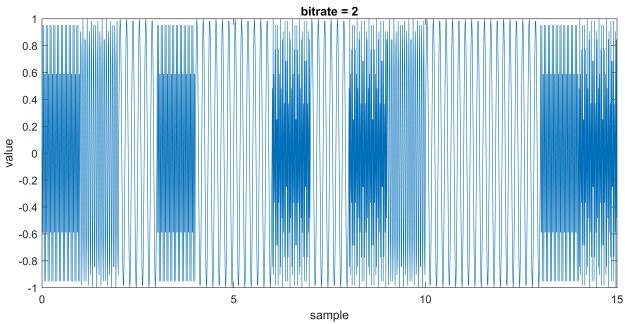
Another thing I consider is that if length of a message is not dividable by bitrate, I add as many as needed semicolons to fix that issue, so if you see semicolons at the end of decoded messages, it is not noise or miscoding.

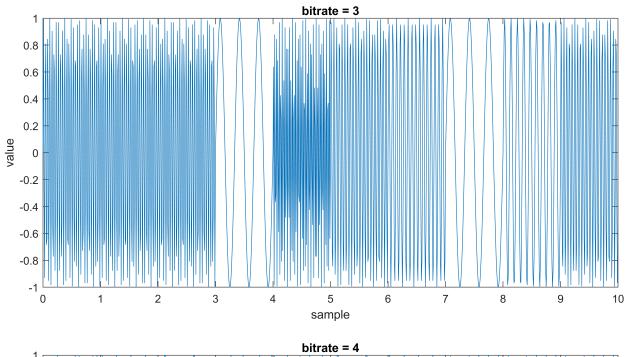
```
fs = 100;
ts = 1 / fs;

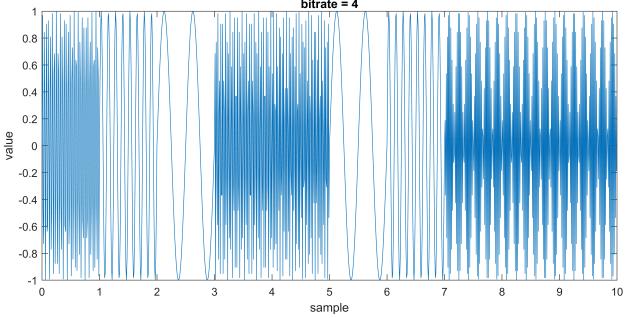
message = 'signal';

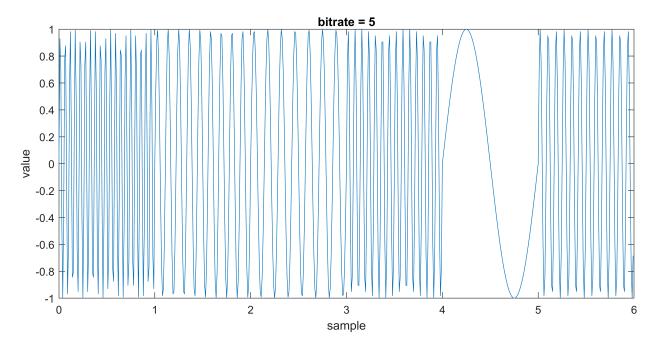
bit_rates = [1,2,3,4,5];
for i = 1:length(bit_rates)
    y = coding_freq(message, bit_rates(i), mapset, fs);
    t = 0:ts:length(y)/fs - ts;
    figure('Position', [0 0 900 400]);
    plot(t, y);
    xlabel('sample');
    ylabel('value');
    title("bitrate = " + num2str(bit_rates(i)));
end
```











2.4 Decoding message

To decode message, first we slice our bitstrings to strings with length of bitrate, and then we do fft and fftshift respectively, and then find max index, but the fftshift default is $\left[-\frac{f_s}{2},\frac{f_s}{2}-1\right]$ the expected frequency is $\left|\max_{1}-\frac{f_s}{2}-1\right|$.

Another thing I consider is that if length of a message is not dividable by bitrate, I add as many as needed semicolons to fix that issue, so if you see semicolons at the end of decoded messages, it is not noise or miscoding.

```
for i = 1:length(bit_rates)
    y = coding_freq(message, bit_rates(i), mapset, fs);
    decoded = decoding_freq(y, bit_rates(i), mapset, fs);
    fprintf("bitrate: %d - decoded message: %s\n", bit_rates(i), decoded);
end

bitrate: 1 - decoded message: signal
bitrate: 2 - decoded message: signal
bitrate: 3 - decoded message: signal
```

2.5 Adding noise with variance 1e-4

bitrate: 4 - decoded message: signal;;
bitrate: 5 - decoded message: signal

The variance is 10^{-4} so the standard deriviation is 10^{-2} , we need to multiply the std to randn. and then add it to the main signal to simulate noise.

```
std = 0.01;
bit_rates = [1,2,3,4,5];
for i = 1:length(bit_rates)
    y = coding_freq(message, bit_rates(i), mapset, fs);
```

```
y = y + std * randn(size(y));
decoded = decoding_freq(y, bit_rates(i), mapset, fs);
fprintf("bitrate: %d - decoded message: %s\n", bit_rates(i), decoded);
end
```

```
bitrate: 1 - decoded message: signal
bitrate: 2 - decoded message: signal
bitrate: 3 - decoded message: signal
bitrate: 4 - decoded message: signal;;
bitrate: 5 - decoded message: signal
```

2.6 Increasing noise and comparing bitrates

we do increase the noise in order to see the maximum amount of standard deriviation of noise added to signal where signal still is decode correctly.

Note: I tested multiple times with many different settings and different hyperparameters, but all resulted that bitrate 5 works better than bitrate 1 in $f = 100 \,\mathrm{Hz}$ but we get expected results in $f = 150 \,\mathrm{Hz}$. Also I repeat every test for every bitrate 100 times(I also tested up to 500 times but also got the same result).

fs = 100 Hz

```
bit rates = [1,2,3,4,5];
maxstd = zeros(size(bit_rates));
fs = 100;
times = 100;
for i = 1:length(bit_rates)
    for t = 1:times
        for variance = 0.1:0.1:2
            y = coding_freq(message, bit_rates(i), mapset, fs);
            y = y + variance * randn(size(y));
            decoded = decoding freq(y, bit rates(i), mapset, fs);
            if(decoded ~= "signal" && decoded ~= "signal;;")
                maxstd(i) = maxstd(i) + variance - 0.1;
                break:
            end
        end
    end
    maxstd(i) = maxstd(i) / times;
end
fprintf("");
fprintf("maximum std where signal is decoded correctly: \n\n\t bitrate 1: %.3f \t bitrate 2: %.
    " %.3f \t bitrate 4: %.3f \t bitrate 5: %.3f", ...
        maxstd(1), maxstd(2), maxstd(3), maxstd(4), maxstd(5));
```

maximum std where signal is decoded correctly:

bitrate 1: 1.361 bitrate 2: 1.395 bitrate 3: 1.436 bitrate 4: 1.421 bitrate 5: 1.471

```
fs = 150 Hz
```

```
fs = 150;
```

```
for i = 1:length(bit rates)
    for t = 1:times
        for variance = 0.1:0.1:2
            y = coding freq(message, bit rates(i), mapset, fs);
            y = y + variance * randn(size(y));
            decoded = decoding_freq(y, bit_rates(i), mapset, fs);
            if(decoded ~= "signal" && decoded ~= "signal;;")
                maxstd(i) = maxstd(i) + variance - 0.1;
                break;
            end
        end
    end
    maxstd(i) = maxstd(i) / times;
end
fprintf("");
fprintf("maximum std where signal is decoded correctly: \n\n\t bitrate 1: %.3f \t bitrate 2: %.
    " %.3f \t bitrate 4: %.3f \t bitrate 5: %.3f", ...
        maxstd(1), maxstd(2), maxstd(3), maxstd(4), maxstd(5));
maximum std where signal is decoded correctly:
```

maximum std where signal is decoded correctly:

bitrate 1: 1.651 bitrate 2: 1.573 bitrate 3: 1.557 bitrate 4: 1.565 bitrate 5: 1.463

We can see the output of signals with $f_s = 150 \,\mathrm{Hz}$ shows what we expected, the more noise the get, the less signal is proofed to noise.

2.8 - 2.9 Increasing frequency(fs) and make signal more persistant

Increasing bandwidth can help the signals to be more persistant against noise, also increasing f_s can help us doing that. because when f_s is increased, we have a larger rnge of frequency to choose for encoding messages, and that will result in increasing distance between two picks after doing fft and fftshif and that makes the decoding easier and less sensetive to noises.

Note that increasing f_s without increasing bandwidth does not improve the noise resistancy of signal.

```
break;
                 end
             end
        end
        maxstd(i) = maxstd(i) / times;
    fprintf("frequency(fs): %d", frequencies(freq));
    fprintf("maximum std: \n");
    fprintf("\t bitrate 1: %.3f \t bitrate 2: %.3f \t bitrate 3: %.3f \t bitrate 4: %.3f \t bit
        maxstd(1), maxstd(2), maxstd(3), maxstd(4), maxstd(5));
end
frequency(fs): 80
maximum std:
 bitrate 1: 1.236 bitrate 2: 1.267 bitrate 3: 1.274 bitrate 4: 1.291 bitrate 5: 1.363
frequency(fs): 90
maximum std:
 bitrate 1: 1.293 bitrate 2: 1.343 bitrate 3: 1.382 bitrate 4: 1.367 bitrate 5: 1.421
frequency(fs): 100
maximum std:
 bitrate 1: 1.366 bitrate 2: 1.401 bitrate 3: 1.447 bitrate 4: 1.440 bitrate 5: 1.494
frequency(fs): 110
maximum std:
 bitrate 1: 1.431 bitrate 2: 1.459 bitrate 3: 1.504 bitrate 4: 1.504 bitrate 5: 1.543
frequency(fs): 120
maximum std:
 bitrate 1: 1.482 bitrate 2: 1.522 bitrate 3: 1.551 bitrate 4: 1.527 bitrate 5: 1.541
frequency(fs): 130
maximum std:
 bitrate 1: 1.528 bitrate 2: 1.534 bitrate 3: 1.589 bitrate 4: 1.590 bitrate 5: 1.586
frequency(fs): 140
maximum std:
 bitrate 1: 1.578 bitrate 2: 1.606 bitrate 3: 1.587 bitrate 4: 1.563 bitrate 5: 1.481
_____
frequency(fs): 150
maximum std:
 bitrate 1: 1.625 bitrate 2: 1.578 bitrate 3: 1.606 bitrate 4: 1.578 bitrate 5: 1.395
```

Functions

implemented functions:

```
function [mapset] = create_mapset()
    mapset = cell(2, 32);
    list = char('a':'z');
    list = [list ,' ', '.', ','!', '"', ';'];

for i = 1:32
    character = list(i);
    bit_string = dec2bin(i - 1, 5);
```

```
mapset{1, i} = character;
        mapset{2, i} = bit_string;
    end
end
function [coded_string] = encode_string(message, mapset)
    coded string = '';
    for i = 1:length(message)
        bit_string = mapset{2, strcmp(mapset(1,:), message(i))};
        coded_string = [coded_string, bit_string];
    end
end
function [coded_freq] = coding_freq(message, bit_rate, mapset, fs)
    while(mod(length(message) * 5, bit_rate) ~= 0)
        message = [message, ';'];
    end
    ts = 1 / fs;
    tstart = 0;
    tend = 1;
    t = tstart : ts : tend - ts;
    fstep = floor((fs / 2) / (2 ^ bit_rate));
    fstart = ceil(fstep / 2);
    bit_string = encode_string(message, mapset);
    coded freq = [];
    for i = (1:bit rate:length(bit string))
        f = fstart + fstep * bin2dec(bit_string(i:i+bit_rate-1));
        new_signal = sin(2 * pi * f * t);
        coded_freq = [coded_freq, new_signal];
    end
end
function [message] = decoding_freq(coded_freq, bit_rate, mapset, fs)
    code string = '';
    fstep = floor((fs / 2) / (2 ^ bit_rate));
    fstart = ceil(fstep / 2);
    for i = (1:fs:size(coded freq,2))
        sliced_freq = coded_freq(i:i+fs-1);
        [~, fmax] = max(abs(fftshift(fft(sliced_freq))));
        fmax = abs(fmax - fs / 2 - 1);
        for j = 0:(2 ^ bit_rate - 1)
            th_left = j * fstep + fstart - fstep / 2;
            th right = j * fstep + fstart + fstep / 2;
            if(fmax >= th_left && fmax <= th_right)</pre>
                code_string = [code_string, dec2bin(j, bit_rate)];
                break;
            elseif(j == 2 ^ bit_rate - 1)
                code_string = [code_string, dec2bin(j, bit_rate)];
            end
        end
    end
```

```
message = '';
for i = 1:5:length(code_string)
    char = mapset{1, strcmp(mapset(2,:), code_string(i:i+4))};
    message = [message, char];
end
end
```