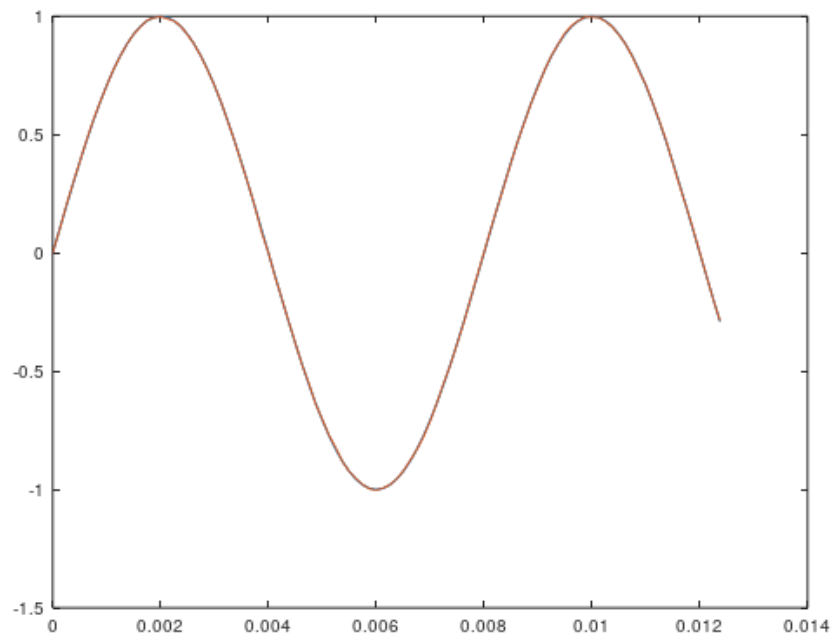


TX00CQ31 –Digital Signal Processing

Study 3: Signal Spectrum	Name(s): Arsi Arola	ID(s): 1706768	Deadline: Exercise 4
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Read these instructions first!

Number	Write your answer in this column	Question
Q1	<p>Write the commands!</p> <pre> SSS = 768; A = 1; Fs = 8000; F = Fs / 2^(mod(SSS, 5) + 3); SNR = 50; tt = 0:(1/Fs):1/Fs * (2^12-1); ss = sin(2*pi*F*tt); ss = ss * A; % set amplitude sound(ss,Fs); var(ss) nvar = var(ss) / 10^(SNR/10); nn = randn(size(ss))*sqrt(nvar); var(nn) mean(nn) ssn = ss + nn; plot(tt(1:100), ss(1:100)); hold on; plot(tt(1:100), ssn(1:100)); hold off; % becuase the SNR is so high we cant even detect the noise from the image just by looking </pre>	<p>Generate wave sam</p> <ul style="list-style-type: none"> • A • S fr 8 % • S ra • S le s <p>Signal to level +5</p>

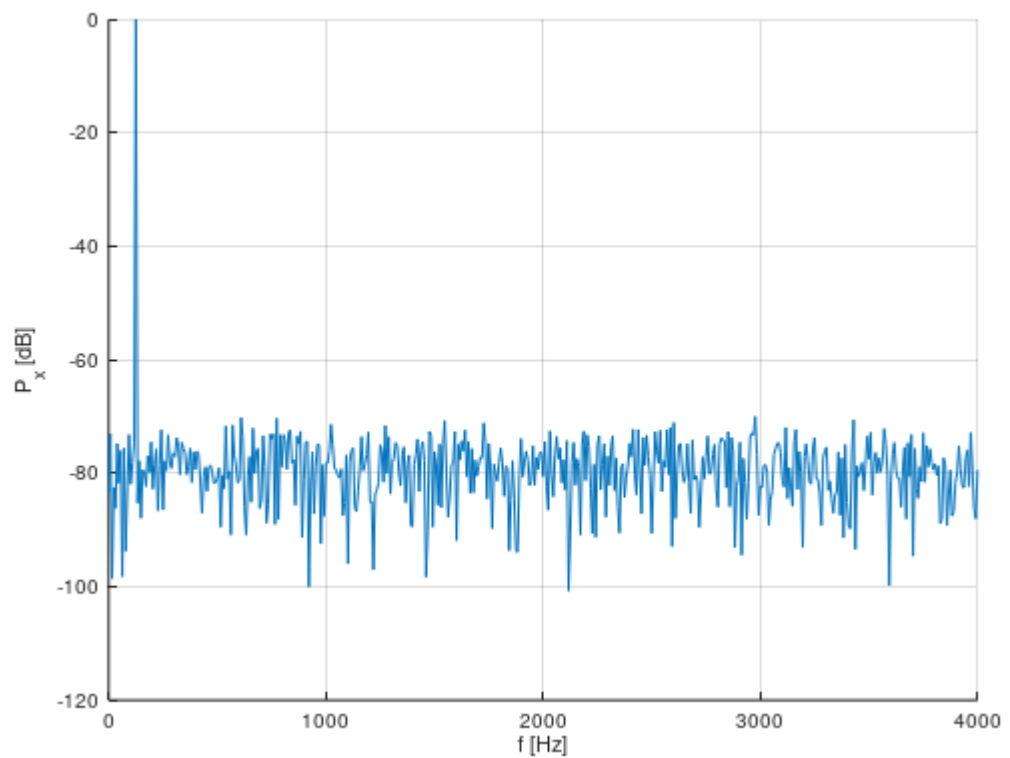


sound(ssn, Fs)
sound(nn, Fs)

Q2

Insert the figure here!

```
hold on;
spk(ssn, Fs);
hold off;
```



Use Four
Transfor
generate
spectrum
previous

- U
fr
F
h
a

Use desi
vertical a

Explain why the signal level in spectrum seems higher than approx. 83 dB compared to the noise level.

It is because of spectral leakage. This make the amplitude increasing. Bin effect also affects the noise

Q3

Write the commands!

```
SS = 68;
```

```
Gain = SS + 2;
```

```
ssnGain = ssn * Gain;
```

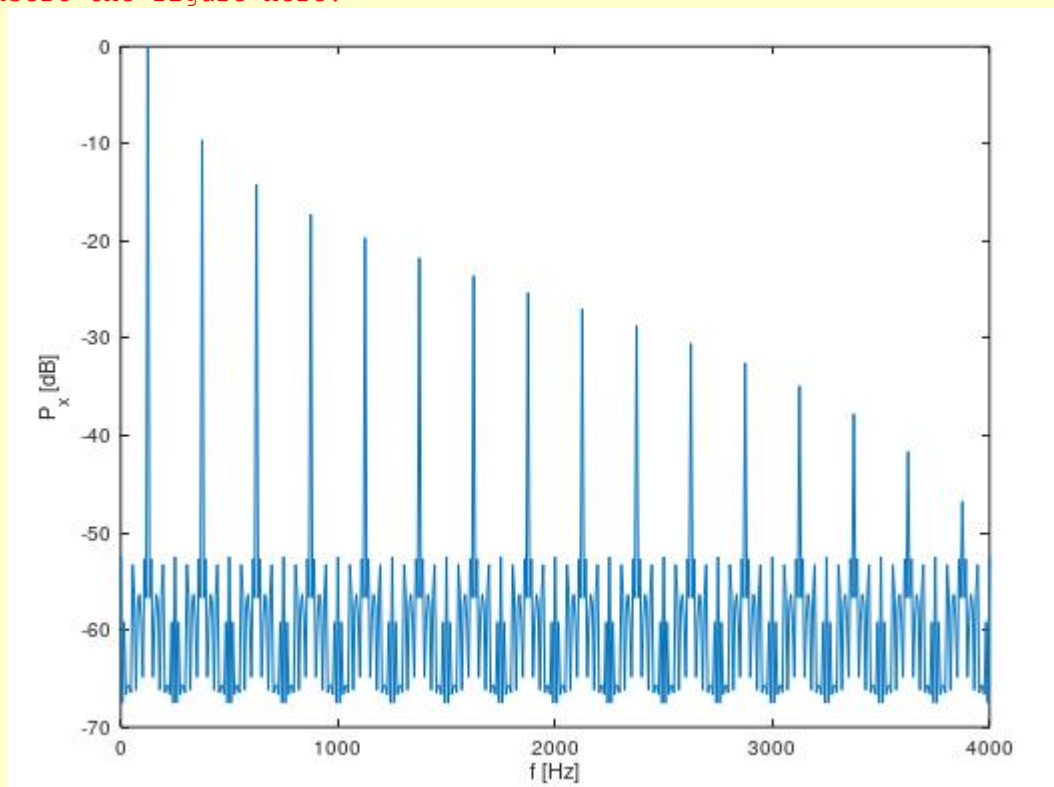
```
ssnGain( ssnGain > 1 ) = 1;
```

```
ssnGain( ssnGain < -1 ) = -1;
```

```
spk(ssnGain,Fs);
```

```
grid; hold off
```

Insert the figure here!



Explain the main differences of the power spectra in this and previous case.

Since power is gained the steps between audio differences is bigger so the power differences are quite jarring and also a lot of the random variance disappears.

Simulate
distortion
sinusoid
Q1 to pro
saturated

- C
- D
- a
- le

Generate
spectrum
sample a
it with th
in Q2.

Q4

Write the commands!

```
SSS = 768;

Fs = SSS * 10;

tt = 0:1/Fs:1-1/Fs;

F = chirp(tt, 100, 1, 200)

A = exp(-tt/0.2);

S = A .* F .+ (randn(size(tt)) .* 0.02);
```

Produce
second lo
sample v

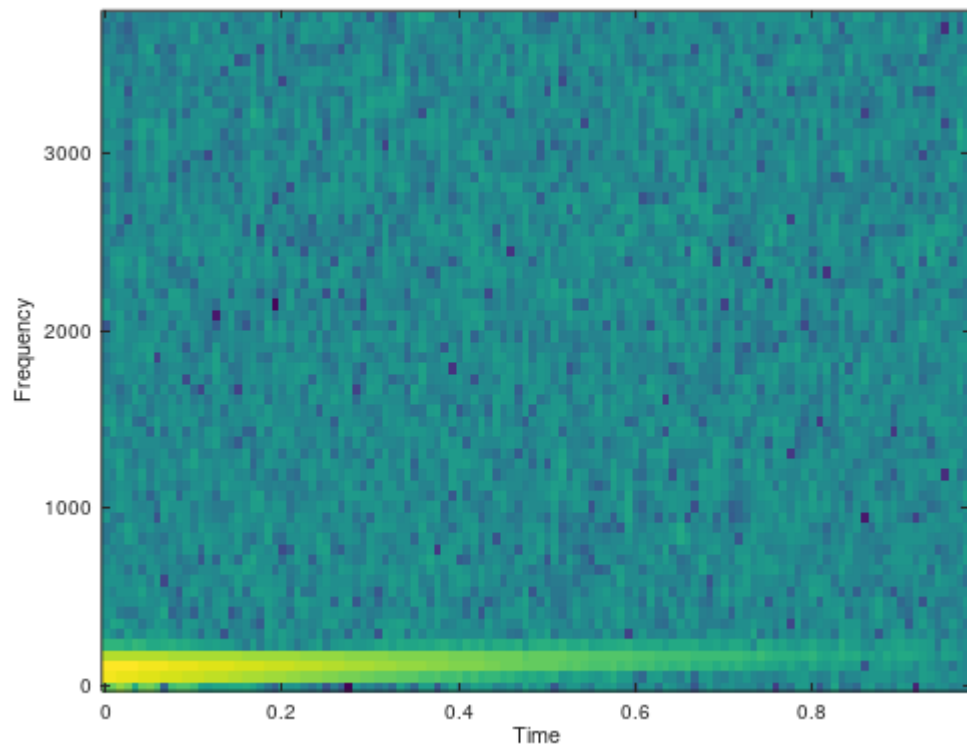
- s
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s
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c
0
• n
a
1
• a
n
s
d
0

sampling
[SSS*10](#)

Q5

Insert the figure here!

Generate
spectrogr
previous
Use frequ
as your v
and time
as your h
axis.



Q6

Insert the figure here!

```
fs = 1e6; % 1Mhz
n = 0:2^12;
u4 = (n >= 0) - (n >= 4);
u8 = (n >= 0) - (n >= 8);
u16 = (n >= 0) - (n >= 16);
```

```
hold on;
spk(u4, fs);
spk(u8, fs);
spk(u16, fs);
hold off;
```

% When the pulse width doubles the spectral content of the pulse halves.

Generate pulses of 4μs, 8μs, 16μs when the frequency is 1MHz. Then evaluate the spectrum of the pulses in MATLAB. How the spectral content changes and its spectral content a

