



# 📡 Chapter - Frequency-Domain Analysis of Captured IQ Data

## 🔍 What We're Going to Do

When we capture I/Q samples from the RTL-SDR, we're essentially freezing the **raw RF baseband signal** (complex-valued time series). Time-domain plots are useful to check if data is valid, but most of the real insights come from transforming into the **frequency domain** using the **Fourier Transform (FFT)**.

**Our goals in this step:**

1. **Convert IQ samples → Spectrum**
    - Use FFT to transform from time domain → frequency domain.
    - This tells us *which frequencies are present* and *their relative power*.
  2. **Plot the Power Spectrum (FFT magnitude)**
    - x-axis: frequency bins (Hz)
    - y-axis: power (dB)
  3. **Plot the Spectrogram (Waterfall)**
    - Spectrum over time (sliding FFTs).
    - Useful for seeing bursts, frequency hopping, or temporal activity.
  4. **Extract Key Insights**
    - Where is the signal centered?
    - What's the occupied bandwidth?
    - Are there bursts or hopping activity?
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## 🔧 The Method

1. **Read IQ Data**
  - We'll start from your `.bin` capture (e.g., `Prac_1.bin`).
  - Reshape into I/Q pairs (complex signal).
2. **Compute FFT**
  - Choose FFT size ( `NFFT` ), typically power of 2 like 1024, 2048, 4096.

- Apply a window (Hann) to reduce spectral leakage.
- Use `fftshift` so that 0 Hz is in the center of the plot.

### 3. Convert FFT Magnitude to dB

- `20 * log10(|FFT|)`
- This gives us the familiar spectrum view.

### 4. Spectrogram

- Split samples into overlapping windows.
  - Compute FFT per window.
  - Stack results → gives time vs frequency power view.
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## First the Idea:

Imagine your IQ samples are like a song recorded in a tape.

You hear the sound in time — but you don't know which notes (frequencies) are being played strongly or softly.

👉 ### That's where the FFT comes in.

The FFT is like a magic prism: it takes your time-domain IQ samples (the messy wave) and splits them into frequency pieces.

Time domain → tells you what the wave looks like over time.

Frequency domain → tells you which frequencies are present and how strong they are.

That's the whole point:

Convert IQ samples → FFT → see the spectrum.

Capture samples (IQ = I + jQ, complex).

Run FFT (Fast Fourier Transform).

FFT is your spectrum analyzer in Python.

So, as Step 1: What are we trying to do?

We have IQ samples captured from the RTL-SDR.

In time-domain, IQ samples just look like noisy squiggles.

To know what frequencies are present, we need to look in the frequency domain.

That's where the FFT (Fast Fourier Transform) comes in → it's like turning a messy sound waveform into a "spectrum analyzer."

👉 ### So the job is simple:

IQ samples (time-domain) → FFT → Spectrum (frequency vs power).

Note:-

IQ data is stored as complex numbers:

I (In-phase) → real part

Q (Quadrature) → imaginary part

Together they form a complex number:  $I + jQ$ .

That's why we don't use float here → floats can't carry both I & Q at once.

complex64 = 32-bit real + 32-bit imaginary → compact and standard for SDR data.

Plot the magnitude → boom, you see the spectrum.

```
## np.fft.fft(raw)
```

Takes the IQ samples (time series) and transforms them to frequency domain.

The output array shows how much of each frequency is present.

Think of it as:

☞ “If I play all possible pure tones, how much does each tone match my captured signal?”

```
### np.fft.fftshift(spectrum)
```

By default, FFT puts 0 Hz (DC) at the left edge of the plot.

That looks odd, because we usually expect negative frequencies on the left and positive frequencies on the right.

fftshift just re-centers the spectrum:

Middle = 0 Hz

Left =  $-F_s/2$

Right =  $+F_s/2$

☞ This makes the spectrum look like what a spectrum analyzer shows.

```
### np.abs(spectrum_shifted) ** 2
```

FFT gives complex numbers (amplitude + phase).

But what we usually want is power (energy at each frequency).

`np.abs(x)` gives magnitude.

`**2` squares it to get power.

☞ So this is basically: “How strong is each frequency component?”

```
### Plotting
```

X-axis: Frequency bins (not yet converted to Hz, but we can later with `np.fft.fftfreq`)

Y-axis: Power

✓ That's it. You just built your first spectrum analyzer in ~10 lines of Python.

Input: messy IQ squiggles

Output: clear spectrum of frequencies present

## When you do:

```
spectrum = np.fft.fft(raw)
```

raw has N samples (say,  $N = 4096$ ).

The FFT output is also length N.

Each element `spectrum[k]` corresponds to a frequency bin, i.e., a tiny slice of the total frequency range.

How bins map to actual frequencies

Bin index k goes from 0 to N-1.

Frequency per bin ( $\Delta f$ ) =  $f_s / N$ , where  $f_s$  = sample rate.

Bin 0 = 0 Hz (DC)

Bin 1 =  $f_s/N$  Hz

...

Bin N-1 =  $(N-1) * (f_s/N)$  Hz

If you use `fftshift`, the bins get rearranged:

Left half → negative frequencies ( $-f_s/2 \dots 0$ )

Right half → positive frequencies ( $0 \dots f_s/2$ )

## Quick example :

$f_s = 2.4$  MHz

$N = 4096$

Then:

$\Delta f = f_s / N = 2\_400\_000 / 4096 \approx 585.94$  Hz per bin

Bin 0 → 0 Hz

Bin 1 → 585.94 Hz

Bin 2 →  $2 * 585.94 \approx 1171.88$  Hz

...

Bin 2048 (middle after `fftshift`) → 0 Hz

Bin 0 after `fftshift` →  $-f_s/2 \approx -1.2$  MHz

Why we talk in bins vs frequency

“Bin” is the index in the FFT array.

Frequency is what the bin represents in Hz:  $f = k * f_s / N$  (or after shift,  $f = (k-N/2)*f_s/N$ ).

## Python Practical



```
import numpy as np
import matplotlib.pyplot as plt
```

### 1. Load the raw IQ samples

```
raw = np.fromfile("Prac_1.bin", dtype=np.complex64)
```

### 2. Apply FFT to move from time → frequency domain

```
spectrum = np.fft.fft(raw)
```

### 3. Shift zero frequency (DC) to the center for better plotting

```
spectrum_shifted = np.fft.fftshift(spectrum)
```

### 4. Convert to power (magnitude squared)

```
power = np.abs(spectrum_shifted) ** 2
```

## 5. Plot the result

```
plt.plot(power)
plt.title("Frequency Spectrum")
plt.xlabel("Frequency Bin")
plt.ylabel("Power")
plt.show()
```

### ① Read the IQ samples

```
raw = np.fromfile(filename, dtype=np.uint8)
I = raw[0::2].astype(np.float32) - 127.5
Q = raw[1::2].astype(np.float32) - 127.5
IQ = I + 1j*Q
```

- `raw[0::2]` → picks all **I samples**
  - `raw[1::2]` → picks all **Q samples**
  - `astype(np.float32)` → converts 8-bit integers to floats for calculations
  - `-127.5` → centers the data around **0** (from `[0, 255]` → `[-127.5, 127.5]`)
  - `IQ = I + 1j*Q` → builds a **complex64 array** that FFT can understand
- 

### ② FFT: Time → Frequency domain

```
spectrum = np.fft.fft(IQ)
```

- Converts your **time-domain complex signal** into **frequency bins**.
  - Each element `spectrum[k]` corresponds to **the strength of a specific frequency component**.
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### 3 FFT Shift: Center 0 Hz

```
spectrum_shifted = np.fft.fftshift(spectrum)
```

- Moves **DC (0 Hz)** to the center of the array.
  - Makes plotting intuitive: negative frequencies on the left, positive on the right.
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### 4 Power Spectrum

```
power = np.abs(spectrum_shifted) ** 2
```

- FFT output is **complex** → has magnitude and phase.
  - `np.abs()` → magnitude
  - `**2` → power (energy at each frequency bin)
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### 5 Plotting

- **First plot** → raw FFT magnitude (not shifted)
- **Second plot** → shifted FFT for clear frequency visualization
- **Third plot** → power spectrum vs frequency bins

✓ Right now the X-axis is in **bins**, not Hz. Each bin corresponds to `fs/N` Hz.

## What You'll See & Analyze

- **FFT Plot (Spectrum)**
  - Peaks → carriers or strong tones.
  - Width of main lobe → occupied bandwidth.
  - Noise floor visible at bottom.
- **Spectrogram (Waterfall)**
  - Horizontal bright lines → continuous signals (carriers).
  - Bursts → packet transmissions.
  - Slanted lines → frequency drift or Doppler.

- Multiple bands → possible hopping or multi-signal environment.

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✓ This builds directly on your **basic signal properties**