

DSP Homework 10

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Abstract

1 Videos

2 DSP Hardware Platforms

3 Relationship Between FT and DTFT

We can first put the expression of perfect reconstruction here

$$x_s(t) = \sum_n x(nT) \delta(t - nT) = x_a(t) * s(t) = x_a(t) * \sum_n [\delta(t - nT)] \quad (1)$$

We put (1) into FT

$$\begin{aligned} \tilde{x}_s(f) &= \int_{-\infty}^{\infty} \sum_n [x(nT) \delta(t - nT)] e^{-j2\pi f t} dt \\ &= \sum_n x(nT) e^{-j2\pi f nT} \\ &= \sum_n x(n) e^{-j2\pi n f'} \end{aligned} \quad (2)$$

The value of T (also $\frac{1}{f_s}$) doesn't matter, so we can make

$$f' = \frac{f}{f_s}$$

The (2) can be changed into

$$\tilde{x}_s(f) = \sum_n x(n) e^{-j2\pi n f'}$$

Therefore

$$\begin{aligned}
\tilde{x}_s(f) &= \mathcal{F}[x_s(t)] \\
&= \mathcal{F}[x_a(t) * s(t)] \\
&= \tilde{x}_a(f) \times \tilde{s}(f) \\
&= \tilde{x}_a(f) \times \left[f_s \sum_n \delta(f - nf_s) \right]
\end{aligned}$$

Therefore, the relationship can be described as

$$\tilde{x}(f) = \tilde{x}_s(f) = \tilde{x}_a(f) \times \left[f_s \sum_n \delta(f - nf_s) \right] \quad (3)$$

4 From Analog Signal to DFT

Assuming there's an analog signal $x_a(t)$, computer cannot deal with analog values, so we can sample it use Shannon/Nyquist sampling method.

$$x(n) = x_s(t) = x_a(t) * \sum_n [\delta(t - nT)] = \sum_n x(nT) \delta(t - nT) \quad (4)$$

Then, we should analyze $x(n)$ in frequency domain, calculate the FT of $x(n)$ as below. Also, because computer cannot deal with analog values, so we should assume $n \in [0, N)$ to make $x_a(t)$ time-limited.

$$\begin{aligned}
\tilde{x}_s(f) &= \int_{-\infty}^{\infty} \sum_{n=0}^{N-1} [x(nT) \delta(t - nT)] e^{-j2\pi f t} dt \\
&= \int_{-\infty}^{\infty} \sum_{n=0}^{N-1} [x(nT) e^{-j2\pi f nT} \delta(t - nT)] dt \\
&= \sum_{n=0}^{N-1} \left[x(nT) e^{-j2\pi f nT} \int_{-\infty}^{\infty} \delta(t - nT) dt \right] \\
&= \sum_{n=0}^{N-1} x(nT) e^{-j2\pi nT \cdot f} \\
&= \sum_{n=0}^{N-1} x(n) e^{-j2\pi n \cdot f}
\end{aligned} \quad (5)$$

Because now $\tilde{x}_s(f)$ is discrete and we don't care about the sampling period T , so we make $f = f \times T$ in (5).

However, $\tilde{x}_s(f)$ is still not discrete. So, as what we did to $x_a(t)$, we should sample $\tilde{x}_s(f)$ in frequency domain again. Rewrite the summation of (5) as below:

$$\tilde{x}_s(f) = x(0) + x(1)e^{-j2\pi f} + x(2)e^{-j2\pi \cdot 2f} + \dots + x(N-1)e^{-j2\pi \cdot (N-1)f}$$

The period is decided by $x(1)e^{-j2\pi f}$ and is 1 here. We should sample it in $[0, 1)$. Assuming $f = \frac{k}{N}$ $k \in [0, N)$, we have

$$\tilde{x}_s\left(\frac{k}{N}\right) = \tilde{x}_s(k) = \sum_{n=0}^{N-1} x(n) e^{-j2\pi n \cdot \frac{k}{N}}$$

$x_s(k)$ is also discrete now, so that it can be processed with the computer.

5 Analog Frequency in DFT

Variables in (4) are n and k .

First, here's my thoughts about n . Since $\tilde{x}_s(k)$ is just a sampling of $\tilde{x}_s(f)$, if we can find high analog frequency in $\tilde{x}_s(f)$, the same method works for $\tilde{x}_s(k)$. Go back and see (3), it tells us that $\tilde{x}_s(f)$ is just many copies and frequency-shift of $\tilde{x}_a(f)$. Therefore, the bigger n gets, the higher $\tilde{x}_s(f)$ reach.

Then, k also matters. k has its source from f , when k (also f) gets bigger, no doubt the frequency gets bigger. However, n decides on frequency domain which period we are in. If n is fixed, we cannot reach other other frequency out of this specific interval.

6 Conclusion