## EECS 16B Designing Information Devices and Systems II Spring 2016 Anant Sahai and Michel Maharbiz Discussion 13A

## 1. Revisiting the DFT basis

In lecture, we show that the Discrete Fourier Fransform (DFT) represents a projection of a length $n$ signal $\vec{x}$ onto the set of $n$ sampled complex sinusoids generated by the $n$-th roots of unity. Here we want to discuss interpolation over DFT basis.
Recap of DFT: We can think of a real-world signal that is a function of time $x(t)$. By recording its values at regular intervals, we can represent it as a vector of discrete samples $\vec{x}$, of length $n$.

$$
\vec{x}=\left[\begin{array}{c}
x[0]  \tag{1}\\
x[1] \\
\vdots \\
x[n-1]
\end{array}\right]
$$

Let $\vec{X}=\left[\begin{array}{lll}X[0] & \ldots & X[n-1]\end{array}\right]^{T}$ be the signal $\vec{x}$ represented in the frequency domain, that is

$$
\begin{equation*}
\vec{X}=U^{-1} \vec{x}=U^{*} \vec{x} \tag{2}
\end{equation*}
$$

where $U$ is a matrix of the DFT basis vectors $\left(\omega=e^{i \frac{2 \pi}{n}}\right)$.

$$
U=\left[\begin{array}{ccc}
\mid & & \mid  \tag{3}\\
\vec{u}_{0} & \cdots & \vec{u}_{n-1} \\
\mid & & \mid
\end{array}\right]=\frac{1}{\sqrt{n}}\left[\begin{array}{ccccc}
1 & 1 & 1 & \cdots & 1 \\
1 & \omega & \omega^{2} & \cdots & \omega^{n-1} \\
1 & \omega^{2} & \omega^{4} & \cdots & \omega^{2(n-1)} \\
\vdots & \vdots & \vdots & & \vdots \\
1 & \omega^{n-1} & \omega^{2(n-1)} & \cdots & \omega^{(n-1)(n-1)}
\end{array}\right]
$$

Alternatively, we have that $\vec{x}=U \vec{X}$ or more explicitly

$$
\begin{equation*}
\vec{x}=X[0] \vec{u}_{0}+\cdots+X[n-1] \vec{u}_{n-1} \tag{4}
\end{equation*}
$$

In other words, $\vec{x}$ is a linear combination of the complex exponentials $\vec{u}_{i}$ with coefficients $X[i]$.
(a) For $x(t)=e^{i \frac{2 \pi}{3}}$, sketch the real and complex parts versus $t$.
(b) Sample $x(t)$ at $t=0,1,2$. How are those sampling points related to our DFT basis for $n=3$ ?
(c) Show that for any $n$ and $k, \vec{u}_{-k}=\vec{u}_{n-k}$
(d) According to (c), the DFT matrix can be represented as

$$
U=\left[\begin{array}{cccccc}
\mid & \mid & \mid & & \mid & \mid \\
\vec{u}_{0} & \vec{u}_{1} & \vec{u}_{2} & \cdots & \vec{u}_{-2} & \vec{u}_{-1} \\
\mid & \mid & \mid & & \mid & \mid
\end{array}\right]
$$

(e) Compute the DFT coefficients $\vec{X}$ for the following signal:

$$
\vec{x}=\left[\sin \left(\frac{2 \pi}{3}(0)\right) \quad \sin \left(\frac{2 \pi}{3}(1)\right) \quad \sin \left(\frac{2 \pi}{3}(2)\right) \quad \sin \left(\frac{2 \pi}{3}(3)\right) \quad \sin \left(\frac{2 \pi}{3}(4)\right) \quad \sin \left(\frac{2 \pi}{3}(5)\right)\right]^{T}
$$

(f) Now consider a real discrete-time signal:

$$
\vec{x}=\left[\begin{array}{c}
1 \\
x[1] \\
x[2] \\
1 \\
1-\frac{\sqrt{3}}{2} \\
x[5]
\end{array}\right]^{T}
$$

We know $X[m]=0$ for $m= \pm 2$. Show that we could recover the whole signal $\vec{x}$ based on the information above. What is $\vec{X}$ ?
(g) What if we don't know $x[4]$ is $1-\frac{\sqrt{3}}{2}$ ? Is $\vec{x}$ unique?
(h) Given a continues time sinusoidal signal $x(t)=\sin \left(\frac{2 \pi}{3} t\right)$, what is its frequency? What is the sampling rate for creating the discrete signal $\vec{x}$ in (e)?
(i) Sample $x(t)=\sin \left(\frac{2 \pi}{3} t\right)$ with the sample rate $=2 \mathrm{~Hz}$ between $0 \leq t<3$. How many data points do you get? Collect those sample points as a discrete signal $\vec{y}$. Compare the DFT coefficients of $\vec{y}$ with the result in (e). Explain their relationship.
(j) Sample $x(t)=\sin \left(\frac{2 \pi}{3} t\right)$ with the sample rate $=2 \mathrm{~Hz}$ between $0 \leq t<6$. How many data points do you get? Collect those sample points as a discrete signal $\vec{z}$. Compare the DFT coefficients of $\vec{z}$ with the result in (e). Explain their relationship.
(k) Consider a length $n$ discrete-time signal $\vec{x}$, along with its DFT coefficients, $\vec{X}$. If we know $X[m]=0$, for all $|m|>k$, what is the minimum number of sampling points we need to interpolating the $\vec{x}$ ?
(l) Sample $x(t)=\sin \left(\frac{2 \pi}{3} t\right)$ with the sample rate $=2 / 3 \mathrm{~Hz}$ between $0 \leq t<3$. Are you able to reconstruct the signal based on the sample points?

