# EECS 16B Designing Information Devices and Systems II Spring 2017 Murat Arcak and Michel Maharbiz Discussion 14B

## Notes

#### **Recap: LTI System Model**

Recall that h[n], or the system response, is the system output to the input  $\delta[n]$ . We then decompose our input signal, x[n], into a series of shifted  $\delta$  functions.

$$x[n] = x[0]\delta[n] + x[1]\delta[n-1] + \dots + x[k-1]\delta[n-k+1] + \dots$$

Since the system is linear and time invariant, the output is going to be,

$$y[n] = x[0]h[n] + x[1]h[n-1] + \dots + x[k-1]h[n-k+1] + \dots$$

If x[n] and h[n] are finite, we can construct a vector-matrix model of LTI systems. Let x[n] have length P and h[n] have length Q + 1.

**Important.** When constructing the vector representations of x[n] and h[n], add zeros the end so that the length of both vectors are at least P + Q.

The vector-matrix model is as follows. Note the matrix structure.

Succinctly, we can represent this as,

 $\vec{y} = \boldsymbol{H}\vec{x}$ 

#### DFT + LTI systems

Rather beautifully, DFT basis vectors are eigenvectors of H. We will have P + Q DFT vectors, since that is the dimensionality of our model.

$$u_k[n] = \frac{1}{\sqrt{P+Q}} e^{i\left(\frac{2\pi}{P+Q}k\right)n}$$

Let  $\vec{u}_k$  be the vectorized version of  $u_k[n]$  and let H[k] be the  $k^{th}$  DFT coefficient of h[n]. (Recall that  $H[k] = \langle \vec{h}, \vec{u}_k \rangle$ ).

$$\boldsymbol{H}\vec{u}_{k} = \underbrace{\left(\sqrt{P+Q} \times H[k]\right)}_{\text{eigenvalue}} \vec{u}_{k}$$

This is really useful. (1) can be really difficult to work with, so we simply decompose it into the DFT domain. Let *Y* be the DFT coefficient of  $\vec{y}$ , *H* be the DFT of  $\vec{h}$  and *X* be the DFT of  $\vec{x}$ . Then,

$$\begin{bmatrix} Y[0] \\ Y[1] \\ \vdots \\ \vdots \\ \vdots \\ Y[P+Q-1] \end{bmatrix} = \sqrt{P+Q} \begin{bmatrix} H[0] & 0 & 0 & \cdots & 0 & 0 & 0 & \cdots & 0 \\ 0 & H[1] & 0 & \cdots & 0 & 0 & 0 & \cdots & 0 \\ 0 & 0 & H[2] & \cdots & 0 & 0 & 0 & 0 & \cdots & 0 \\ \vdots & 0 \\ \vdots & 0 \\ \vdots & 0 \\ \vdots & 0 \\ 0 & 0 & \cdots & 0 & 0 & 0 & 0 & H[P+Q-1] \end{bmatrix} \begin{bmatrix} X[0] \\ X[1] \\ \vdots \\ \vdots \\ \vdots \\ X[P+Q-1] \end{bmatrix}$$

$$(2)$$

In other words,

$$Y[k] = \sqrt{P + Q}H[k]X[k]$$

#### **REMEMBER TO ZERO PAD WHEN VECTORIZING.**

### Questions

#### 1. Fast Convolution Using FFT

We have a system with finite impulse response h(k), k = 0, ..., M.

(a) How do you compute the response of this system to an input u(k), k = 0, ..., L - 1?

- (b) How many operations (multiplications and additions) does this take?
- (c) Can you think of a faster way to do this when M + L is large? (Hint: computing the DFT using the Fast Fourier Transform (FFT) algorithm takes approximately  $N \log N$  operations where N is the length of the vector being transformed.)

#### 2. DFT and Bode Plots

Consider the circuit

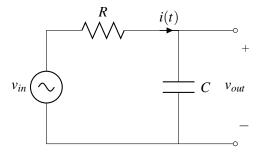


Figure 1: A first order RC Low Pass Filter

- (a) Write a differential equation describing the evolution of  $v_{out}$ .
- (b) Draw a Bode plot for this circuit.
- (c) Discretize the differential equation assuming a zero order hold on the input  $v_{in}$ . (Hint: The differential equation  $a\dot{x} = -x + u$  can be discretized as  $x(k+1) = e^{-\frac{1}{a}T}x(k) + (1 e^{-\frac{1}{a}T})u(k)$  where T is the sampling period.)
- (d) What is the impulse response of this LTI system?
- (e) In Python, compute the DFT of the impulse response. You can assume that R = 1, C = 1/10, N = 1001 (number of samples) and T = 1/10 (sampling period).
- (f) Plot the amplitude and phase of the first N/2 DFT coefficients. How do they compare with the Bode plots?

#### **Contributors:**

- Siddharth Iyer.
- John Maidens.