[10:30am] Wireless resources and CDMA (related to lab #4)

- In wireless communications, we have multiple resources. For multiple users to communicate simultaneously, we can divide these resources.
 - Time (Time division duplex) used in 4G/5G cellular systems
 - Each user gets a time slot and uses the entire transmission band.
 - Only one user communicates at any given time.
 - Schedule each user, e.g. rotate among the different users
 - Frequency (Frequency division duplex) used in 4G/5G cellular systems
 - Each user gets own frequency band within the transmission band
 - All users transmit at the same time
 - Space (Multiple antennas to perform directional filtering) used in 4G/5G 0
 - Each user gets a cone (set of angles) 30° for frequencies below 7 GHz • (4G/5G) and 2-7° for millimeter wave frequencies (5G)
 - Can be used with either time or frequency division duplex to improve communication performance
 - Code (CDMA, code division multiple access)
 - Each user gets own initial value for linear shift feedback register (LSFR) to generate the psuedonoise (PN) sequence
 - Use pseudonoise sequences to get the codes (lab #4)
 - To transmit a bit of one, transmit the next 32 values of your PN sequence using your LSFR value
 - To transmit a bit of zero, transmit the logical complement of the the next 32 values of your PN sequence using your LSFR value
 - Basestation correlates with the known sequences
 - Other users appears to be noise
 - Allows multiple users to transmit at the same time and frequency
- See Appendix L

Student #1 Student #1 Student #2 i Student N Resource Allocation in 46156 erial View Vireless Communications Sectors) Dan Time Division Duplexing (Basestation) · Frequency Division Duplexing Uplink [Student #2] Student N] asestation Downlink -> Student #1 con use fregs all ever Tx band. GPS + 26 code Division Multiple Access · Angle - spatial - using lots of artennas · Each Student (user) would have a different code (different PN start)

[11:00 am] Conclusion to FIR and IIR filtering

- An LTI system is uniquely defined by its impulse response
- Complex exponentials are eigenfunctions of LTI systems
- System must be initially at rest (zero initial conditions) as a necessary condition for linearity to hold
- Linear phase requires even or odd symmetry about its midpoint.
 - Group delay is negative derivative of phase
 - Linear phase gives constant group delay
 - FIR filters can be easily designed to have linear phase by enforcing the symmetry
 - Phase is also important if multiple transmitters or receivers are present
 - Example: phase difference between two ears is used to identify different directions of sound
- Several common FIR filters
 - Averaging filter
 - Lowpass
 - Linear phase
 - Impulse response is a rectangular pulse
 - First order difference
 - Highpass
 - Linear phase
- IIR filters
 - For a given frequency selective specification, IIR filters can have lower order
 - Nonlinear phase, but may have approximately linear phase over a range
 - Can become unstable due to numerical errors
 - Implement as cascade of biquads

[11:20] Sampling and Aliasing

- Examples of aliasing
 - o <u>Helicopter video demonstrating aliasing</u>
 - Helicopter rotor is between 6 and 10 Hz, but has between 2-7 blades
 - Camera frame rate is between 24 and 30 hz
 - Motor appears still because frame rate is an integer multiple of rotor rate times the number of blades.
 - Water and sound experiment

• Sampling theorem: in order to sample a signal x(t) whose maximum frequency is f_{max} and reconstruct it perfectly, we must sample at a rate f_s such that

 $f_s > 2f_{\max}$

- We can place an analog lowpass filter before the sampler to enforce this condition
- Sampling can be modeled by multiplication of a signal with an impulse train

$$\underbrace{x(t)}_{\substack{\text{continuous}\\\text{signal}}} \cdot \underbrace{s_T(t)}_{\substack{\text{impulse train}\\\text{impulse train}}}$$

• In the frequency domain, this creates replicas of the spectrum shifted by multiples of the sampling rate

$$\mathcal{F}\{s_T(t)\} = \frac{1}{T} s_{1/T}(f)$$
$$\mathcal{F}\{x(t)s_T(t)\} = f_s \sum_{k=-\infty}^{\infty} X(f - kf_s)$$

• To reconstruct, we must apply an ideal lowpass filter to remove these replicas