Lecture 11

Discrete Time Signals

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Discrete time signals

- Sampling a continuous time signal at frequency fs converts the signal to discrete time.
- How frequently we need to sample is governed by the Sampling Theorem.



Basic discrete signals



Discrete sinusoidal signal



Compare this with continuous time signal equation:

$$x(t) = A\cos(\omega_0 t + \phi)$$
 sampling $x[n] = A\cos(\Omega_0 n + \phi)$

- The discrete time signal is sampled at f_S , where $T_S = 1/f_S$ is the sampling period (i.e. time step between successive samples).
- Note that Ω₀ in discrete time domain is angle increment of this sinusoidal signal between samples. Its unit is radians/sample (not rad/sec as in continuous time case.

Discrete sinusoidal signal



Operations on discrete signals

Sum of two signals:

$$s[n] = x[n] + y[n]$$

Product of two signals:

 $p[n] = x[n] \cdot y[n]$

Amplification of a signal:

 $y[n] = \alpha \, . \, x[n]$

Delaying a signal by k samples:

$$y[n] = x[n-k]$$

Discrete signal and impulses

We can represent a causal discrete signal x[n] in terms of sum of weighted delayed impulses:



 $x[n] = x[0] \,\delta[n] + x[1] \,\delta[n-1] + x[2] \,\delta[n-2] + x[3] \,\delta[n-3] + \dots$

$$x[n] = \sum_{k=0}^{\infty} x[k] \,\delta[n-k]$$

Energy of a discrete signal

The energy of a discrete signal can be computed easily – simply sum the square of each sample values:



 Instantaneous energy of the signal at sample *i* over a window of K samples is:

$$E\{x_i[n]\} = \sum_{k=0}^{K-1} |x[i+k]|^2$$

An alternative representation of discrete signals

- Instead of representing discrete signals in terms of impulse functions with various delay, we can transform the discrete signal into another domain (or mathematical representation).
- Let us assume that we use a transformation that maps an impulse function with delay k such that:

$$\delta[n-k] \stackrel{\mathsf{Z}}{\to} z^{-k}$$

Then the discrete signal x[n] is transformed to another function in terms of the variable z:

$$x[n] = x[0] \,\delta[n] + x[1] \,\delta[n-1] + x[2] \,\delta[n-2] + x[3] \,\delta[n-3] + \dots$$

$$x[n] \xrightarrow{\mathsf{Z}} X[z] = x[0] z^{0} + x[1] z^{-1} + x[2] z^{-1} + x[3] z^{-3} + \dots$$
$$X[z] = \sum_{k=0}^{\infty} x[k] z^{-k}$$

 X[z] is the z-transform of the signal x[n]. For now, you only need to remember that z^{-k} represents k sample period delay.

Example – Gyroscope signal

- Assume pitch angle p(t) changes from 0 to
 10 shortly after t = 0
- After sampling, we get p[n] as shown where it takes two sample periods to reach final value of 10

• Gyroscope measure angular velocity
$$\frac{dp}{dt}$$

In discrete time domain, we get

$$\frac{\Delta p}{\Delta t} = \dot{p}[n] = \frac{p[n] - p[n-1]}{\Delta t}$$

- For discrete signals, we compute differentiation using differences
- This graph shows the sample values of the gyroscope reading



Derive pitch angle from gyro data

- How do we obtain an estimate of the pitch angle p̂[n] from the gyro reading p̂[n]?
- We perform integration:

 $\hat{p}[\mathbf{n}] = \hat{p}[\mathbf{n} - 1] + \dot{p}[n]\Delta t$

 So integration in discrete time domain is perform with summation in a recursive equation



Problem of Drift in Gyroscope

- All transducers that measure physical quantities (such as angular velocity) has errors
- In gyroscope, the problematic error is the DC offset. That means even if the gyro is NOT rotating, the IMU returns a small value *\epsilon*
- The result of such offset after integration is to yield a pitch angle estimate that increases or decrease linearly with time as shown here.

$$\hat{p}[n] = \hat{p}[n-1] + \varepsilon$$
 for $n > 0$



Problem of noise in accelerometer data

- Let us assume that the Pybench board is instantaneously rotated to 10 degrees
- The accelerometer's measurement will reflect this, but superimpose on this is a lot of noise
- Partly is due to unwanted forces acting on the mass inside the IMU



Three Big Ideas (1)

1. Discrete sinusoidal signal is of the form:

$$x(t) = A\cos(\omega_0 t + \phi)$$
 sampling $x[n] = A\cos(\Omega_0 n + \phi)$
 Ω_0 is angle increment between samples
ln radian/sample

2. Any discrete signal can be expressed as **weighted sum** of unit impulses, delayed and scaled.



Three Big Ideas (2)

3. If we map the delayed impulse (delay function) as:

$$\delta[n-k] \rightarrow z^{-k}$$

We transforms discrete time signals to an new domain, called z-domain.

This transform is known as z-transform, and is useful to model discrete signals.

$$x[n] \xrightarrow{z-\text{transform}} X[z] = x[0] z^0 + x[1] z^{-1} + x[2] z^{-1} + x[3] z^{-3} + \dots$$