Applied Programming

Numerical Differentiation and Digital Differentiators

More details in: U. Ascher and C. Grief, "A First Course in Numerical Methods", chapters 14, 15

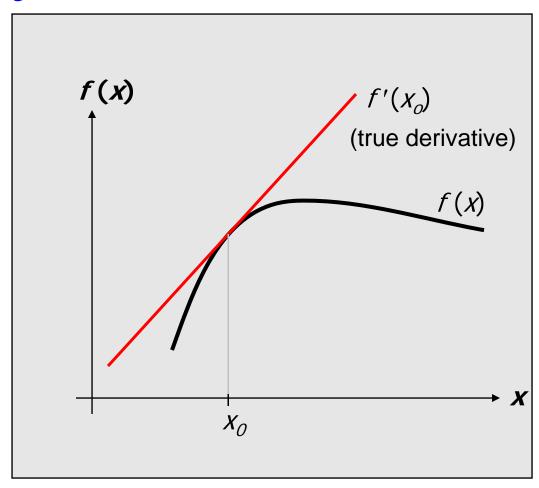
Why Numerical Differentiation?

Analytical differentiation is:

- Difficult:
 - -requires symbolic computations.
- Not feasible:
 - closed form of function not available.
- Not practical:
 - in many situations a numerical version of the derivative is preferred.
- Necessary:
 - Many sensors read position but we may want velocity (encoders)

Derivative: Geometric Interpretation

In the plane (2D), the derivative at a point x_o is the slope of the tangent to a curve f(x) at the point x_o



Derivative: Definition

Calculus:

The derivative of a function f(x) at a point x_i defined as

$$f(x_i) = \lim_{h \to 0} \frac{f(x_i + h) - f(x_i)}{h}$$

- In many practical situations we do not know the function f(x) explicitly
- Our general objective is to estimate the derivative of the function f(x) at a desired point x_i

Numerical Differentiation

Numerical differentiation is required for:

- I. Estimation of the derivative of a function available only at discrete set of points, e.g., $y_i = f(x_i)$, i=1...nExample: Compute velocity using position encoder measurements.
- II. Discretization of differential equations to simulate physical systems

Example: Flight simulators

Numerical Differentiation in Estimation (I)

Objective

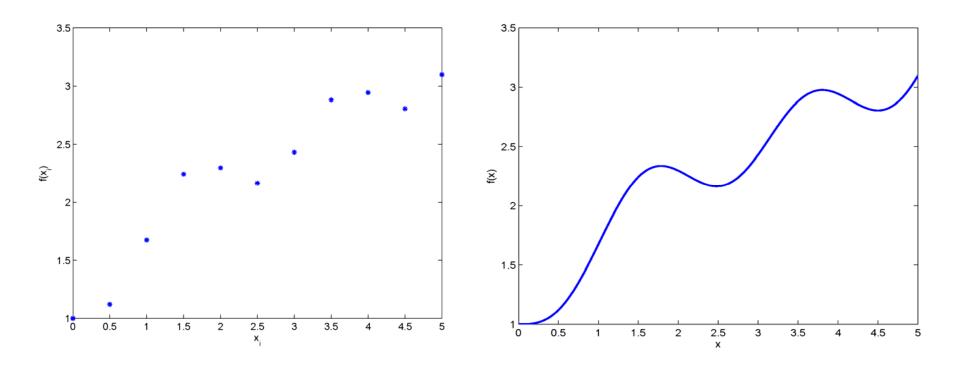
Find the derivative of a function *given* only samples of the function at discrete points, e.g., $y_i = f(x_i)$, i=1..n

- Sometimes we want to find the derivative at all points: f'(x)
- Sometimes we want the derivative at a given point: $f'(x^*)$

Warning: In most applications the samples of f(x) are noisy

Numerical Differentiation I

Problem: Find the derivative of a function *given* a finite number of points $\{(x_i, y_i)\}, y_i = f(x_i)$



Given data points

We are not given "the function"

Numerical Differentiation I

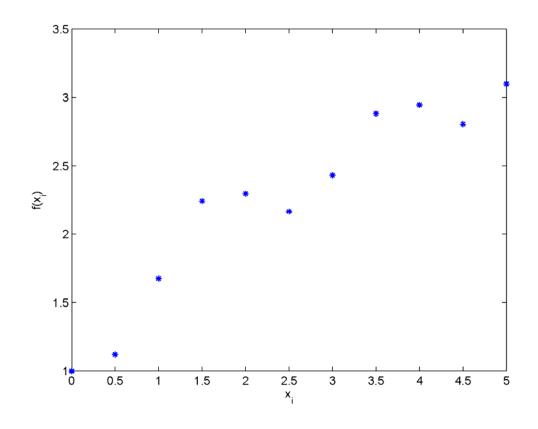
Given a finite set of points $\{(x_i, y_i)\}, y_i = f(x_i)$ **estimate the derivative** of the function

- One way to solve this problem is in two steps:
 - 1. Interpolate (or fit) a function to the data (e.g., find a "model" for the data)
 - 2. Differentiate (analytically) the closed form of the interpolated (fitted) function.

Example:

Interpolation/Differentiation

Example: Find the derivative of f(x) given the points shown



Proposed Solution:

- 1. Perform a cubic spline interpolation
- Take derivative of interpolated splines

Example: Interpolation/Differentiation

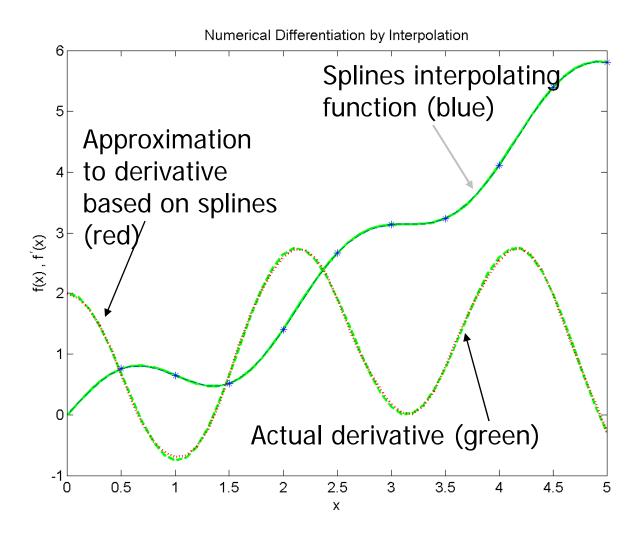
Results: (Plots on next slide)

- Spline Coefficients (d,c,b,a)
 Derivatives
 - 0 2,0000 -1.5488-0.1895-4.6464-0.37892.0000 1.5559 -2.51270.6489 0.7590 4.6678 -5.0253 0.6489 2.0461 -0.1788-0.69680.6498 6.1382 -0.3576-0.6968-1.27132.8903 0.6590 0.5125 -3.81385.7807 0.6590 -2.23721.4056 0.9834 2.5958 -6.7116 1.9668 2.5958 0.8653 -2.37241.9013 2.6698 2.5959 -4.7448 1.9013 2.2376 -1.07440.1779 3.1355 6.7127 -2.1489 0.1779 -0.6938 2.2819 0.7816 3.2355 -2.08144.5638 0.7816 -2.41081.2412 2.5432 4.1101 -7.23242.4824 2.5432 0.1565 -2.37505.3907 1.9763 $3d_i(x-x_i)^2 + 2c_i(x-x_i) + b_i$

Advantage: We can find the *derivative at any* point in the range (not only at data points)

[note: derivative is a quadratic spline]

Example: Interpolation/Differentiation



Numerical Differentiation I

Given a finite set of points $\{(x_i, y_i)\}$, $y_i = f(x_i)$ estimate the derivative of the function

- If we only need to compute the derivative at the data points other alternatives could be more efficient.
- To derive algorithms for numerical differentiation we can start from the definition of derivative.

Numerical Differentiation at Points

Using the definition from calculus

$$f'(x_i) = \lim_{oldsymbol{h} o 0} rac{f(x_i + oldsymbol{h}) - f(x_i)}{oldsymbol{h}}$$

- We can approximate derivative at a point x_i by taking the difference between to adjacent values h units apart and dividing by h
 - h is a very small non-zero value

Numerical Differentiation

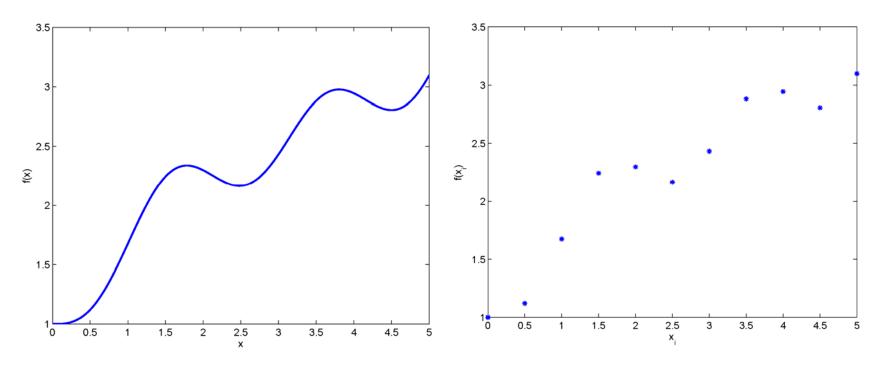
The proposed approximation has clear practical limitations

$$f(x_i) = rac{f(x_i + h) - f(x_i)}{h}$$
 $h o 0$ (for best accuracy)

- 1. Machine precision
 - ⇒ h may not be made arbitrarily small
- 2. May not have control over h
 - ⇒ h may be fixed (by the problem) and too large for accurate approximation

Problem to be Solved

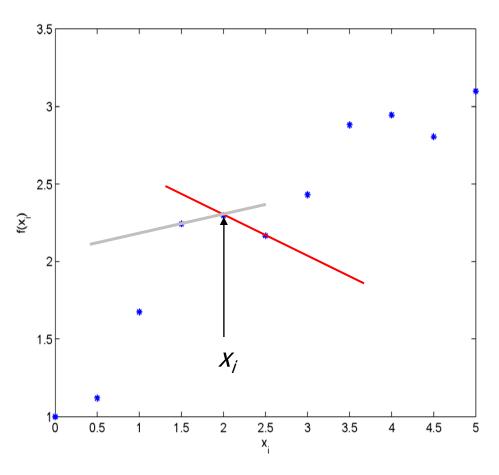
How do we estimate the derivative at x_i from just a set of points $\{(x_i, y_i)\}, y_i = f(x_i)$ [for a given h]?



We are not given this

... but only some points

Numerical Differentiation at Pts



Calculus: Use two points (the current point and one neighbor) to estimate the derivative at the current (*i* th) point.

We could choose:

- a. x_i and x_{i+1} , (right derivative, or forward)
- b. x_{i-1} and x_i (left derivative, or backward)

These choices lead to the two point *forward* and *backward* difference approximations, respectively.

$$f(x_i) = \frac{f(x_i + h) - f(x_i)}{h}$$

Forward Difference

Let:

x_i – current point

x_{i+1} - next point

 $h = delta = x_{i+1} - x_i$

SO:

$$f'(x_i) = rac{f(x_{i+1}) - f(x_i)}{x_{i+1} - x_i}$$

Backward Difference

Let:

x_i – current point

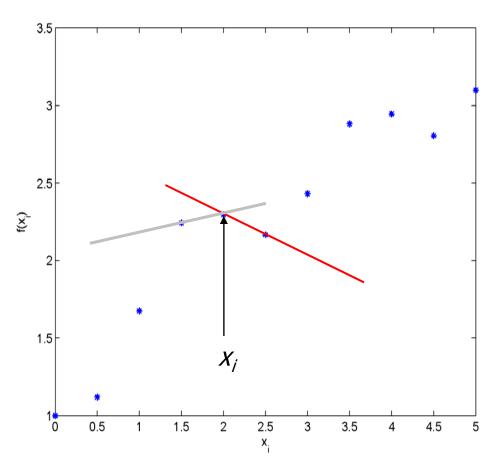
x_{i-1} - previous point

 $h = delta = x_i - x_{i-1}$

SO:

$$f'(x_i) = \frac{f(x_{i+1}) - f(x_i)}{x_{i+1} - x_i}$$
 $f'(x_i) = \frac{f(x_i) - f(x_{i-1})}{x_i - x_{i-1}}$

Numerical Differentiation at Pts



In mathematical notation we have:

Forward Difference

$$f'(x_i) = \frac{f(x_{i+1}) - f(x_i)}{x_{i+1} - x_i}$$

Backward Difference

$$f'(x_i) = \frac{f(x_i) - f(x_{i-1})}{x_i - x_{i-1}}$$

Are any of these "better"?

Which one is Better

- Depends on what we mean by better!
 - depends on the application
 - In principle they coincide when h=0
- Two common criteria:
 - The size of the (truncation) error as a function of the increment h
 - ii. The *causality* of the formula (required in most DSP applications)

Truncation Error Analysis

 To estimate the error (as h changes) we perform a Taylor series expansion in n terms with remainder around the point x_i:

$$f(x_{i} + \Delta x) = f(x_{i}) + \Delta x f^{(1)}(x_{i}) + \frac{(\Delta x)^{2}}{2!} f^{(2)}(x_{i}) + \dots + \frac{(\Delta x)^{n}}{n!} f^{(n)}(x_{i}) + \frac{(\Delta x)^{[n+1]}}{(n+1)!} f^{(n+1)}(\xi),$$

$$x_{i} < \xi < x_{i} + \Delta x$$

$$Remainder$$

(The number of terms *n* in the expansion depends on the differentiation formula analyzed)

Error: Two Point Formulas

Expand the function f around the point x_i in a Taylor series up to second order

$$f(x_i \pm \frac{h}{h}) = f(x_i) \pm \frac{h}{h} f^{(1)}(x_i) + \frac{h^2}{2!} f^{(2)}(\zeta)$$

Then it follows that

FW:
$$f^{(1)}(x_i) = \frac{f(x_i + h) - f(x_i)}{h} - \frac{f^{(2)}(\zeta)}{2} (h)$$

BW:
$$f^{(1)}(x_i) = \frac{f(x_i) - f(x_i - h)}{h} + \frac{f^{(2)}(\zeta)}{2} \left(\frac{h}{h}\right)$$

*** Both have O(h) error ***

Functions as "Signals"

Mathematical notation:

$$\{(x_k, y_k)\}_{k=0}^n, \quad y_k = f(x_k)$$

Signals notation: $h_k = x_k - x_{k-1}$

$$\{(x_k, y_k)\}_{k=0}^n, \quad f(kh_k) = y_k$$

(h_k is the "sampling period")

Special case: Uniform sampling $h = x_k - x_{k-1}$

$$f(kh_k) = y_k \Rightarrow f(kh) = y_k$$

 $\Rightarrow f[k] = y[k]$

It is sufficient to specify k, the index of the kth sample

Causality

- Assume that the function to be differentiated is a time signal (i.e., the independent variable, is time)
- Assume that the signal is sampled uniformly at increments of h units apart

$$f(\mathbf{x}) \stackrel{h}{\to} f(\mathbf{k} \mathbf{h}), \quad k \text{ indep. var}$$

• Using the notation introduced, we can write $f(x_k) = f(k h) = f[k]$

The notation f[i] (with square brackets) is common in DSP (where the "sampling period" h is implicit)

 Consider the problem of estimating the derivative of a signal at the current time k

Forward Difference Approximation

Using a signal interpretation we have:

$$f'(x_k) = \frac{f(x_{k+1}) - f(x_k)}{x_{k+1} - x_k} = \frac{f[k+1] - f[k]}{h}$$

To compute derivative at "time" kh , we need

- Value of signal at future "time", (k+1)k
- Value of signal at current "time", kh
 Computation of derivative at current time requires value of the signal in the future!

(since we cannot use future values, the forward difference is a **non-causal** algorithm for differentiation)

Backward Difference Approximation

Using a signal interpretation we have:

$$f'(x_k) = \frac{f(x_k) - f(x_{k-1})}{x_k - x_{k-1}} = \frac{f[k] - f[k-1]}{h}$$

To compute derivative at "time" kh , we need

- Value of signal at current "time", kh
- Value of signal at past "time", (k-1)h

Computation of derivative at current time requires present and past values only

(since we only use present and past values, the backward difference is a **causal** algorithm for differentiation)

Digital Differentiators

 From a signal processing perspective the differentiation operations can be implemented as a digital filter and is given by the difference equation

$$D_{F_2}[n] = \frac{1}{h} f[n+1] - \frac{1}{h} f[n]$$
Not causal ③
$$D_{B_2}[n] = \frac{1}{h} f[n] - \frac{1}{h} f[n-1]$$
Causal ③
$$f[n] \longrightarrow D_{B_2}[n]$$
Filter
$$D_{B_2}[n]$$

The *output* of this filter is the derivative of the input and is obtained as a *weighted sum* of the input samples. These filters are called **Finite**Impulse Response (FIR) filters

Summary

- Differentiation formulas based only on the value of the function at given points are called finite differences
- Generally, a small h gives better approximation to the derivative
 - i.e., lower truncation error
- The simplest formulas for the first derivatives are the two point forward and backward differences (O(h))
- In *DSP* applications only the *backward difference* can be implemented due to
 causality.

Recall: Two Point Differences

(Two point) Forward Difference

$$f'(x_k) = \frac{f(x_{k+1}) - f(x_k)}{x_{k+1} - x_k} = \frac{f[k+1] - f[k]}{h}$$

(Two point) Backward Difference

$$f'(x_k) = \frac{f(x_k) - f(x_{k-1})}{x_k - x_{k-1}} = \frac{f[k] - f[k-1]}{h}$$

The forward and backward difference formulas both have error that is "O(h)"

• If *h* is fixed, what can we do to minimize the error?

Reducing The Truncation Error

- If we use *more information* about the function, we should be able to obtain *better results* (reduce the error)
- What information can we use?
 The information that we have available is the value of the function at neighboring points.
- If instead of using only two points if we use more neighboring points to approximate the derivatives we may be able to do better

Higher Order Formulas

- Higher order (multiple point) finite difference formulas can be obtained (derived) in two ways:
 - 1. By *numerical interpolation* (e.g., Lagrange polynomials) *followed by symbolic differentiation and sampling*.
 - 2. By *Taylor Series Expansions*. The advantage of this approach is that we *also* get the truncation error estimates

Higher Order Formulas

Taylor Series Approach:

- To derive 3 points formulas proceed as follows:
 - Write one Taylor Series Expansion of the function (with remainder of order h³)
 per neighboring point
 - Combine these expressions into one eliminating terms in h²
 - Rewrite to obtain desired formula.
- This approach can be used to derive differentiation formulas of arbitrary number of points.

Reminder: General Taylor Expansion

• To estimate the error we perform a *Taylor* series expansion in n terms with remainder around the point x_i :

$$f(x_{i} + \Delta x) = f(x_{i}) + \Delta x f^{(1)}(x_{i}) + \frac{(\Delta x)^{2}}{2!} f^{(2)}(x_{i}) + \dots + \frac{(\Delta x)^{n}}{n!} f^{(n)}(x_{i}) + \frac{(\Delta x)^{[n+1]}}{(n+1)!} f^{(n+1)}(\xi),$$

$$x_{i} < \xi < x_{i} + \Delta x$$

In this case n will be 3

3 pts Forward Differences

• 3 pts Forward Difference (D_{F3})

$$f'(\mathbf{x}_k) pprox rac{-3f(\mathbf{x}_k) + 4f(x_{k+1}) - f(x_{k+2})}{x_{k+2} - x_k}$$
 $D_{F_3}[k] = rac{-3f[k] + 4f[k+1] - f[k+2]}{2h}$
 $= rac{1}{h} \left(-rac{3}{2}f[k] + 2f[k+1] - rac{1}{2}f[k+2]
ight)$

• The error is $O(h^2)$

The sum of the "weights" must add up to zero: $-\frac{3}{2} + 2 - \frac{1}{2} = 0$

3 pts Backward Differences

• 3 pts Backward Difference (D_{B3})

$$f'(m{x_k}) pprox rac{3f(m{x_k}) - 4f(m{x_{k-1}}) + f(m{x_{k-2}})}{m{x_k - x_{k-2}}} \ D_{B_3}[k] = rac{3f[k] - 4f[k-1] + f[k-2]}{2h} \ = rac{1}{h} \left(rac{3}{2}f[k] - 2f[k-1] + rac{1}{2}f[k-2]
ight)$$

• The error is $O(h^2)$

The sum of the "weights" must add up to zero (true for all finite difference formulas)

3 pts Central Differences

•3 pts Central Difference (D_{C3})

$$f'(\mathbf{x}_{k}) \approx \frac{f(x_{k+1}) - f(x_{k-1})}{x_{k+1} - x_{k-1}}$$

$$D_{C_3}[k] = \frac{f[k+1] - f[k-1]}{2h}$$

$$= \frac{1}{h} \left(\frac{1}{2} f[k+1] - \frac{1}{2} f[k-1] \right)$$

• The error is $O(h^2)$

Note: This is a **3 points** difference approximation that only **requires 2 function evaluations!**

Error Approximations

High Order Finite Difference Approximations

- Three-point finite difference formulas are of order 2.
 - The error is proportional to h^2
- N-point finite difference formulas are of order N-1, O(h^{N-1})
 - The error is proportional to h^{N-1}
- Error can be controlled without changing the sampling period "h".

Central Difference Advantage

Engineering Significance:

If there is no causality constraint, the central difference is the best because:

- For a given number of data points used in the approximation formula it leads to smaller errors.
- It only requires two data points for evaluation.
- This formulas is widely used in image processing

Digital Differentiators



 The 2nd order backward difference is described by the following difference equation

$$D_{B_3}[n] = rac{3}{2h}f[n] - rac{2}{h}f[n-1] + rac{1}{2h}f[n-2]$$

Compare to the 1st order backward difference

$$D_{B_2}[n] = \frac{1}{h} f[n] - \frac{1}{h} f[n-1]$$

Both can be implemented as FIR filters

Choice of Approximation

- When causality is an issue (digital signal processing) backward approximations are our only choice.
- When causality is not required central differences are more accurate

Caution:

Differentiation exacerbates errors (e.g., amplifies noise). To use it effectively you need to first filter the noise

Applied Programming

Numerical Integration

and

Digital Integrators

Why Numerical Integration?

Analytical integration is:

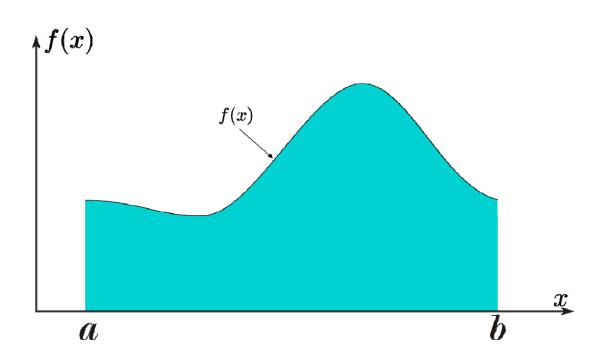
- Difficult:
 - -requires symbolic computations.
- Not possible:
 - closed form solution is not known.
- Not practical:
 - in many situations the function is not know.
- Applications:
 - Digital Integrators are often used in control systems (e.g. PI control)

Integral: Geometric Interpretation

Area under a "curve" between

$$a \le x \le b$$

$$\int_a^b f(x)dx$$



From Calculus:

Limit of Riemann Sum as $\Delta x \rightarrow 0$

$$\int_{a}^{b} f(x)dx = \lim_{\Delta x \to 0} \sum_{k=0}^{n-1} f(x_k) \Delta x$$

Use this idea to develop integration algorithms

Numerical Integration

Objective

Find the integral of a function in a given interval using only values of the function at a finite number of point, e.g., $y_i = f(x_i)$, i=1..n

- Sometimes we are given a closed form expression for the function to integrate and are free to pick the points
- Normally we are only given pairs $(x_i, f(x_i))$ that describe the *values of the function* at given points

Numerical Integration

Engineering applications of numerical integration:

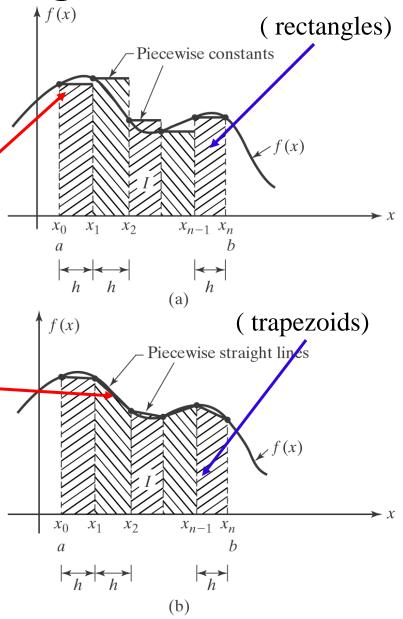
- I. Estimation of the integral of a function available only at a discrete set of points, e.g., $y_i = f(x_i)$, i=1..nExample: Compute velocity from acceleration measurements.
- II. Simulation of dynamic systems described by differential equations

 Example: Flight simulators, general numerical simulators control and signal processing systems, etc. (e.g., Simulink)
- III. Control Systems often use integrators in tracing systems to reduce errors.

Numerical Integration

General approach:

- 1. Discretize independent variable $(\Delta x = x_{i+1} x_i = h)$
- 2. Interpolate function between points x_i and x_{i+1} with a polynomial:
 - 1. Constant (0th order)
 - 2. Line (1st order)
 - 3. Parabola, ...
- 3. Integrate interpolated function, e.g., find area under polynomial between x_i and x_{i+1} and add all.



Signal Processing Perspective

 Discretization is uniform (equidistant pts) and is caused by sampling a input signal

$$f(x) \stackrel{t:=nh}{\longrightarrow} f(kh) = f[k]$$

 Integral is the output of a digital filter (digital integrator)



The integration algorithm defines the digital filter

Forward Rectangular Integrator

- 1. Discretization: Uniform (equidistant pts)
- 2. Interpolation: Constant (0th order poly)
- Derive algorithm $Integrator\ Algorithm \\ I_F[k] = I_F[k-1] + hf[k-1], \\ I_F[-1] = 0$ Digital Filter $I_F[k] = I_F[k]$

The *output* of this filter is the **integral** of the input. Note that the *output depends on past outputs* (it is recursive). These are called **Infinite Impulse Response (IIR) filters**

• Example 1: Integrator 1 (F. Rect.)

$$I_{F}[k] = I_{F}[k-1] + hy[k-1], \quad I_{F}[-1] = 0$$

By direct evaluation:

$$I_{F}[0] = I_{F}[-1] + hy[-1] = hy[-1]$$
 $I_{F}[1] = I_{F}[0] + hy[0] = h(y[-1] + y[0])$
 $I_{F}[2] = I_{F}[1] + hy[1] = h(y[-1] + y[0] + y[1])$
 \vdots
 $I_{F}[n] = h \sum_{k=1}^{n} y[k-1]$

A digital integrator is just an "accumulator"

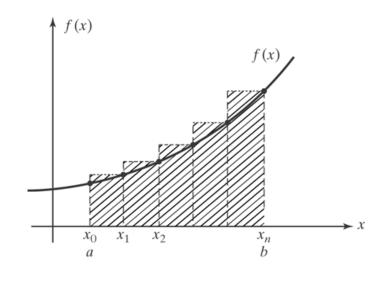
Backward Rectangular Integrator

- 1. Discretization: Uniform (equidistant pts)
- 2. Interpolation: Constant (0th order poly)
- Derive algorithm

Integrator Algorithm

$$I_B[k] = I_B[k-1] + hf[k],$$
 $I_B[-1] = 0$





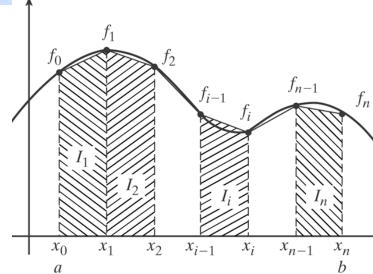
Note that both integration algorithms (forward and backward rectangular) are causal.

Trapezoidal Rule

- 1. Discretization: Uniform (equidistant pts)
- 2. Interpolation: Linear (1st order poly)
- Integrator Algorithm:

$$I_T[k] = I_T[k-1] + \overbrace{\frac{h}{2}(f[k-1]+f[k])}^{I_k}$$
 $I_T[-1] = 0$

As before h is constant and $I_T[k]$ is the approx. integral up to the k^{th} segment



Numerical Integration and Filters

 The recursive integration formulas can be implemented in digital (IIR) filters

(IIR = Infinite Impulse Response)

$$I_F[k] = I_F[k-1] + hy[k-1], \quad I_F[-1] = 0$$
 $I_B[k] = I_B[k-1] + hy[k], \quad I_B[-1] = 0$
 $I_T[k] = I_T[k-1] + \frac{h}{2}(y[k-1] + y[k]), I_T[-1] = 0$

- To "start" the filters we need to set their initial conditions (value of I[k] at k=-1)
- The output of the filter at "time" k is the integral of the function up to that time

Filters and Numerical Integration

What is the area under f(x) from x_o to x_n $\underbrace{f[k]}_{\text{Digital Filter}}$ Digital $I_B[k]$ $I_{B}[k]$

- Given an algorithm (described in recursive form for filtering) the integral of the function over the interval of integration requires the "solution of the difference equation"
- The simplest way to solve these equations is by direct evaluation

Example 1: Integrator 1 (Forward Rect.)

$$I_{F}[k] = I_{F}[k-1] + hf[k-1], \quad I_{F}[-1] = 0$$

By direct evaluation:

$$I_{F}[0] = I_{F}[-1] + hf[-1] = hf[-1]$$
 $I_{F}[1] = I_{F}[0] + hf[0] = h(f[-1] + f[0])$
 $I_{F}[2] = I_{F}[1] + hf[1] = h(f[-1] + f[0] + f[1])$
 \vdots
 $I_{F}[n] = h \sum_{k=1}^{n} f[k-1]$

A digital integrator is just an "accumulator"

• Example 2: Integrator 2 (Backward Rect)

$$I_B[k] = I_B[k-1] + hf[k], \quad I_B[-1] = 0$$

By direct evaluation:

$$I_B[0] = I_B[-1] + hf[0] = hf[0]$$
 $I_B[1] = I_B[0] + hf[1] = h(f[0] + f[1])$
 $I_B[2] = I_B[1] + hf[2] = h(f[0] + f[1] + f[2])$
 \vdots
 $I[n] = h \sum_{k=1}^{n} f[k]$

• Example 3: Trapezoidal rule

$$I[k] = I[k-1] + \frac{h}{2}(f[k-1] + f[k]), I[-1] = 0$$

• By direct evaluation:

$$I[0] = I[-1] + \frac{h}{2} (f[-1] + f[0]) = \frac{h}{2} (f[-1] + f[0])$$

$$I[1] = I[0] + \frac{h}{2} (f[0] + f[1]) = \frac{h}{2} (f[-1] + 2f[0] + f[1])$$

$$\vdots$$

$$I[n] = h \left(\frac{f[-1] + f[n]}{2} + \sum_{k=0}^{n-1} f[k] \right)$$

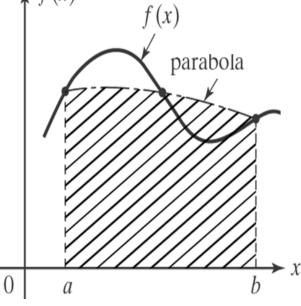
Other Integrators: Simpson (1/3) Rule

- 1. Discretization: Uniform (equidistant pts)
- 2. Interpolation: Quadratic (2st order poly)
- Recursive formula

$$I_S[2k] = I_S[k] + \frac{2h}{6} (y[2k-2] + 4y[2k-1] + y[2k])$$
 $I_S[0] = 0$

$$\uparrow^{f(x)} \qquad \uparrow^{f(x)}$$

•Numerical integral up to $2k^{th}$ segment is $I_S[2k]$



Simpson (1/3) Rule

$$I_S[2k] = I_S[k] + rac{2h}{6} \left(y[2k-2] + 4y[2k-1] + y[2k]
ight)$$
 $I_S[0] = 0$

- Simpson's (1/3) rule is often used in the derivation of a number of numerical Ordinary Differential Equation solvers
- This integrator is not used in DSP
 applications because it is not described
 by a standard difference equation

Choice of Integration Algorithm

If the purpose of numerical integration is:

- To design a *filter or controller*, then the rectangular or trapezoidal rules are often sufficient.
- To integrate a function accurately, then more advanced algorithms such as Romberg Integration or quadrature methods are necessary (beyond scope of this course).

This course will focus on integration algorithms for DSP and control applications.

Automated Controls

- Engineers build systems that "control themselves"
 - Home heating systems
 - Car cruse controls
 - XYZ cutting systems
 - Etc
- All these systems share the same basic requirement:
 - Have a "goal"
 - Measure the "actual"
 - Generate a "correction"

Simplest Control

- Bang-bang
 - -Either "Full ON"
 - -Or "Full OFF"
 - Home furnace is a good example.



www.zoro.com

- Easy to build
 - Not "smooth"

PID Control

- Much "smoother" than bang-bang
 - Many small robots are driven by independent DC motors (sometimes stepper motors).
- To steer these robots the velocity of these motors must be precisely coordinated and controlled.
- This is commonly achieved using a PID (Proportional-Integral-Derivative) controller.

PID Control

OVERSHOOT

-#FS

INPUT STEP

> /SLEW-RATE LIMITING

AΤ t = 0

MPLITUDE

ERROR BAND

FINAL VALUE

TIME

ERROR

BAND

 Proportional Integral Derivative (PID) Control is a popular method of controlling systems using closed-loop feedback. FINAL ENTRY INTO

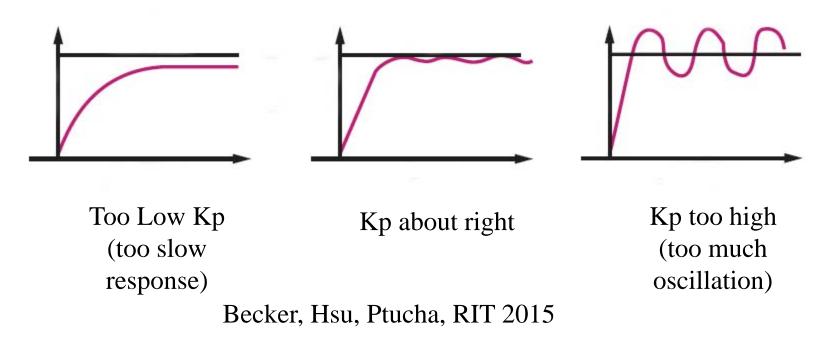
 We would like to get our system to our destination value as soon as possible and then minimize oscillations.

 Lets take a look at Proportional, Integral, and nup.//www.societyonobots.com/programming_riD.snunl Derivative controllers independently, then we'll put them all together.

Becker, Hsu, Ptucha, RIT 2015

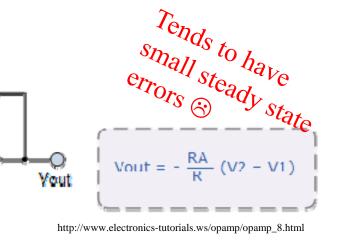
Proportional Control

- When using proportional control alone, the equilibrium state is below Vdes (for all values of Kp).
 - This is because as Vdest approaches Vact, the difference becomes zero and the drive is off.
 - Increasing gain will decrease error but then cause oscillations



Analog Proportional Control

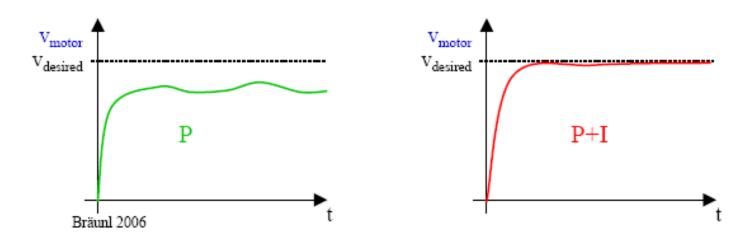
- Operational Amplifier (op amp)
 - Lots of configurations
 - -Setup for "difference" RA



- 2 inputs
 - -V1 what I have
 - -V2 what I want
 - Vout error (drive signal)
- In a digital world, this is subtraction

Integral Control

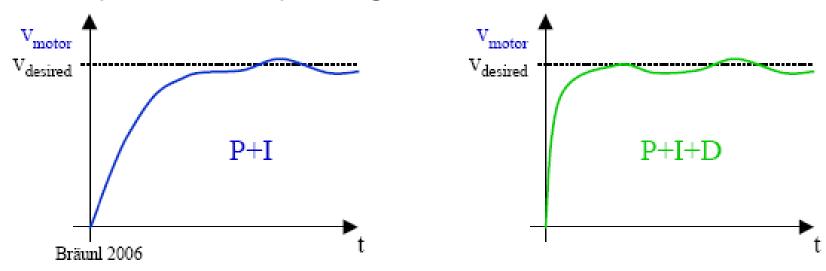
- The primary purpose of the Integral Controller is to minimize any steady-state error introduced by the Proportional Controller.
 - Problem: P-Controller may reach equilibrium without reaching the target value (steady-state error)
 - Solution: Introduce an integral mechanism to eliminate steady-state error.



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Derivative Control

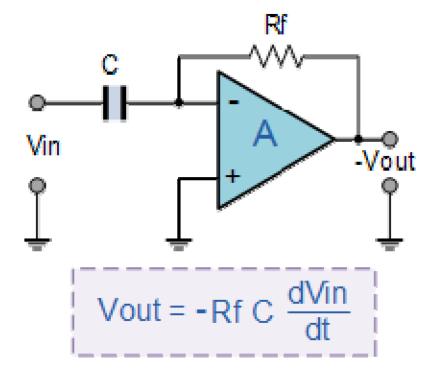
- Adding a Derivative Controller will allow us to bring our process into control faster
 - P-Controller responds slowly to change in input
 - P-Controller with high gain tends to oscillate
- Solution: Add a derivative term for response/dampening



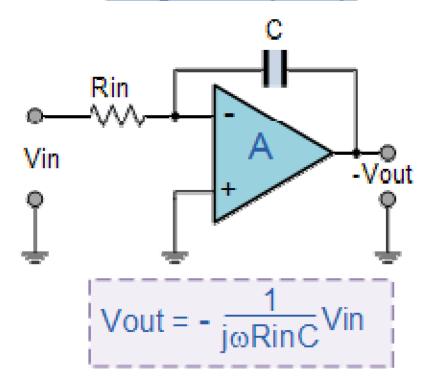
Becker, Hsu, Ptucha, RIT 2015

Analog

Differentiator Op-amp



Integrator Op-amp



- Yes it is possible to do "math" in a pure analog space!
 - But keeping the voltage in the rails is hard

PID Control

 A PID controller takes an error signal, and generates a control output, c(t), that is a weighted sum of the error, e(t),

is a weighted sum of the error,
$$e(t)$$
, $c(t) = K_p e(t) + K_i \int_0^t e(\tau) d\tau + K_d \frac{de(t)}{dt}$

• The "weights" (gains) K_p , K_i and K_d must be chosen to achieve a desired control action. They are often found by "trial and error", using "tuning algorithms" or using digital control techniques (Control Systems Theory)

Digital PID from Analog PID

How can we implement the PID control

$$c(t) = K_p e(t) + K_i \int_0^t e(\tau) d\tau + K_d \frac{de(t)}{dt}$$
 in a computer (e.g., microcontroller) ?

- Need to "discretize" the above equations:
 - Perform differentiation and integration numerically
 - The increment h represents the sampling period (how often we read the sensors and update the control)

$$c(t) = K_p e(t) + K_i \int_0^t e(\tau) d\tau + K_d \frac{de(t)}{dt}$$

First, need to select a sampling period
 h. This depends on the performance objectives, hardware used, etc. (faster is NOT better !!)

$$t = kh, k = ..., -1, 0, 1, ...$$

• Discretize term by term (h becomes implicit) c[k] = P[k] + I[k] + D[k]

$$c(t) = K_p e(t) + K_i \int_0^t e(\tau) d\tau + K_d \frac{de(t)}{dt}$$

Proportional Term: P[k]

$$P(kh) = K_p e(kh)$$

$$P[k] = K_p e[k]$$

The proportional gain for the digital PID and analog PID are the same

$$c(t) = K_p e(t) + K_i \int_0^t e(\tau) d\tau + K_d \frac{de(t)}{dt}$$

Integral term: I/k | Use trapezoid rule

$$I(kh) pprox K_i \left(\int_0^{(k-1)h} e(au) d au + \int_{(k-1)h}^{kh} e(au) d au
ight)$$
 $pprox K_i \left(I[k-1] + rac{h}{2} \left(e[k] + e[k-1]
ight)
ight)$

The integral term cannot be simplified further (the "output" depends on previous outputs)

$$c(t) = K_p e(t) + K_i \int_0^t e(\tau) d\tau + K_d \frac{de(t)}{dt}$$

Derivative term: D[k] Use backward difference

$$D(hk) pprox rac{K_d}{h} \left(e(hk) - e(h(k-1))
ight)$$

$$\left|D[k] = rac{K_d}{h}\left(e[k] - e[k-1]
ight)
ight|$$

The derivative gain for the digital PID and analog PID are different (scaled by sampling period)

Digital PID Control

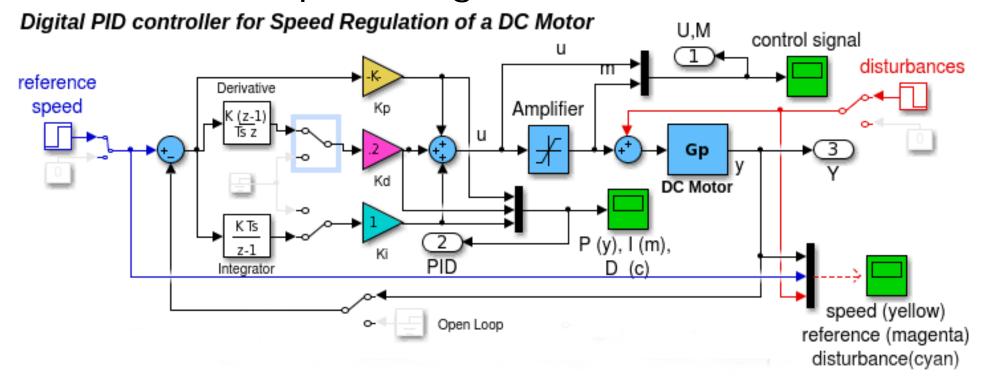
- The digital PID control law can be computed term by term.
- It is possible to write the PID controller using transfer functions (more in DSP)

$$C(z) = K_p E(z) + rac{K_i h}{2} \left(rac{z+1}{z-1}
ight) E(x) + rac{K_d}{h} \left(rac{z+1}{z}
ight) E(z)$$

The integrator is an IIR filter and the differentiator an FIR filter

Digital PID Control - Simulation

Demo: Speed Regulation of DC Motor



Simulink Block Diagram of a Digital PID Control System

