Two Approaches to Physical Modeling

Music 420 Lecture Elementary Finite Different Schemes

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Outline

- White Box and Black Box Physical Modeling
- Ordinary Differential Equations
 - Equivalent Circuits
 - Reference Directions
 - Examples
- Difference Equations (Finite Difference Schemes)
 - Backward Euler (BE)
 - Forward Euler (FE)
 - Trapezoidal Rule for Numerical Integration

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- Bilinear Transform (BLT)
- Digital Filter Design Formulation

- 1. "White Box" Modeling:
 - (a) Find the describing *differential equations* from basic physical principles
 - (b) *Digitize* the differential equations to obtain *difference equations* implemented in software
- 2. "Black Box" Modeling:
 - (a) Measure the *system response* to a representative set of input signals
 - (b) Fit a *computational model* to the measured input-output set
 - (c) In the Linear, Time-Invariant (LTI) case, a Multi-Input, Multi-Output (MIMO) digital filter will suffice

This class *blends* white- and black-box approaches:

- 1. LTI sections become fast, accurate digital filters
- 2. *Nonlinear* or *rapidly time-varying* subsystems normally get a white-box approach (reeds, hammers, bows, ...)

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Equivalent Circuit for a Force-Driven Mass

Ordinary Differential Equations

Ordinary Differential Equations (ODEs) typically result from Newton's laws of motion:

f(t) = m a(t) (Force = Mass times Acceleration)

Acceleration a(t) relates to velocity v(t) and position x(t) by differentiation with respect to time $t{:}$

$$a(t) \stackrel{\Delta}{=} \dot{v}(t) \stackrel{\Delta}{=} \frac{d \dot{x}(t)}{dt} \stackrel{\Delta}{=} \ddot{x}(t) \stackrel{\Delta}{=} \frac{d^2 x(t)}{dt^2}$$

Physical Diagram:

$$a(t), v(t), x(t) \rightarrow \\ \begin{array}{c} x = 0 \\ f(t) \rightarrow m \\ \hline \end{array} \\ \hline \end{array}$$

Force f(t) driving mass m along frictionless surface



- Mass m is an *inductor* L = m Henrys
- Driving force f(t) is a voltage source
- Mass velocity v(t) is the *loop current*

The ODE is obtained from the equivalent circuit by summing all "voltages" around the current loop to zero to obtain

$$-f(t) + m\dot{v} = 0$$

The minus sign for f(t) occurs because the current arrow entered the minus side of the "voltage source"

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- "Reference directions" (\pm) on the voltage source and circuit elements may be chosen arbitrarily—just keep track and be consistent
- When f(t) is positive, "current" is pushed from its + to its terminal, *i.e.*, v(t) will be positive if the rest of the circuit is just a wire or a resistor
- The "force drop" across the mass m is positive when v(t) increases in the direction going from its + to terminal. This can be interpreted as the inertial *reaction force* of the mass that opposed the external *applied force* (Newton's first law of motion)

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Force driving an ideal mass and dashpot





$$0 = -f(t) + f_m + f_\mu
0 = -f(t) + m \dot{v}(t) + \mu v(t)$$

Force f(t) driving mass m along surface with friction force $\mu v(t)$:

$$\begin{split} f(t) \;&=\; m\,\ddot{x}(t) + \mu\,v(t) \\ &=\; m\,\ddot{x}(t) + \mu\,\dot{x}(t) \end{split}$$

- Note that the friction force is positive to the *left* in this figure, *i.e.*, it is a *reaction force*
- The inertial reaction force of the mass points to the left as well (not shown, but equal to -f(t))



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An ideal spring described by Hooke's law

$$f(t) \;=\; k\, x(t) \;=\; k \int_0^t v(\tau)\, d\tau \;\; \longleftrightarrow \;\; \frac{V(s)}{s}$$

where k denotes the *spring constant*, x(t) denotes the *compressive* spring displacement from rest at time t, and f(t) is the force required for displacement x(t)

If the force on a mass is due to a spring then, as discussed later, we may write the ODE as

$$k x(t) + m \ddot{x}(t) = 0$$

(Spring Force + Mass Inertial Force = 0)

Physical diagram:





- Driving force $f_{ext}(t)$ is to the right on the mass
- Driving force + mass inertial force + spring force = 0
- Mass velocity = spring velocity
- This is a *series* combination of the spring and mass

If two physical elements are connected so that they share a *common velocity*, then they are said to be formally connected *in series* The "series" nature of the connection becomes more clear when the *equivalent circuit* is considered:

$$v_{m}(t) = v_{k}(t)$$

$$f_{ext}(t)$$

$$f_{m}(t)$$

$$f_{m}(t$$

- The driving force is applied to the mass such that a positive force results in a positive mass displacement and positive spring displacement (compression)
- The common mass and spring velocity appear as a single current running through the inductor and capacitor that model the mass and spring, respectively

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Mass-Spring-Dashpot ODE

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If the mass is sliding with *friction*, then a simple ODE model is given by

 $k x(t) + \mu \dot{x}(t) + m \ddot{x}(t) = 0$

(Spring + Friction + Inertial Forces = 0)

Physical diagram:



We will use such ODEs to model mass, spring, and dashpot elements, and their equivalent circuits

Difference Equations (Finite Difference Schemes)

- There are many methods for converting ODEs to difference equations
- For white-box modeling, we'll use a very simple, order-preserving methods which *replaces each derivative or integral with a first-order finite difference:*

$$\dot{x}(t) \stackrel{\Delta}{=} \frac{d}{dt} x(t) \stackrel{\Delta}{=} \lim_{\delta \to 0} \frac{x(t) - x(t - \delta)}{\delta}$$
$$\approx \frac{x(nT) - x[(n - 1)T]}{T} \stackrel{\Delta}{=} \hat{x}(t)$$

for sufficiently small T (the sampling interval)

- This is formally known as the *Backward Euler* (BE), or *backward difference* method for differentiation approximation
- In addition to BE, we'll look at Forward Euler (FE), BiLinear Transform (BLT), and a few others
- For a more advanced treatment of finite difference schemes, see **Numerical Sound Synthesis** by Stefan Bilbao (2009, Wiley)

Backward Euler Finite-Difference Equation for a Force-Driven Mass

• Newton's f = ma can be written in terms of force f and velocity v or momentum p = mv as

$$f(t) = m \dot{v}(t) = \dot{p}(t)$$

• The backward-difference substitution gives

$$f(nT) \approx m \frac{v(nT) - v[(n-1)T]}{T} \stackrel{\Delta}{=} m \hat{v}(nT)$$

for $n = 0, 1, 2, \ldots$ Or, in a lighter notation,

$$f_n \approx m \frac{v_n - v_{n-1}}{T} \stackrel{\Delta}{=} m \hat{v}_n, \quad n = 0, 1, 2, \dots$$

with $v_{-1} \stackrel{\Delta}{=} 0$

- \bullet We often use a "hat" to denote approximation: $\hat{v}\approx v$
- In this case, \hat{v}_n is more accurately written as $\hat{v}_{n-1/2}$
- Solving for v_n yields a *difference equation* (finite difference scheme):

$$\hat{v}_n = \hat{v}_{n-1} + \frac{T}{m} f_n, \quad n = 0, 1, 2, \dots$$

with $\hat{v}_{-1} \stackrel{\Delta}{=} 0$

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Summary of Backward Euler

$$v_n = v_{n-1} + T \hat{v}_n$$

$$\iff \boxed{\hat{v}_n = \frac{v_n - v_{n-1}}{T}}$$

$$\uparrow = \uparrow$$

$$V(z) = z^{-1}V(z) + T \hat{V}(z)$$

$$\Rightarrow \qquad \boxed{\hat{V}(z) = \frac{1 - z^{-1}}{T}V(z)}$$

Expressing BE as a *conformal map* from s to z:

$$s \leftarrow \frac{1 - z^{-1}}{T}$$

The ideal differentiator H(s) = s, which is a first-order continuous-time LTI filter, is mapped to a first-order discrete-time LTI filter $H(z) = (1 - z^{-1})/T$.

Accuracy of Backward Euler

Suppose we take the backward-difference approximation $f_n = (m/T)(v_n - v_{n-1})$, and expand v_{n-1} in Taylor series about v_n . This yields:

$$f_n = \frac{m}{T} \left(v_n - \left(v_n - T \frac{dv}{dt} \Big|_{nT} + \mathcal{O}(T^2) \right) \right)$$
$$= m \frac{dv}{dt} \Big|_{nT} + \mathcal{O}(T)$$

- We say that the backward difference approximation has an error of *order* T, written $\mathcal{O}(T)$
- The order of the error tells us how fast the error approaches zero as the sampling rate $f_s = 1/T$ approaches infinity
- Backward Euler maps infinite frequency $s = \infty$ to z = 0 (maximally damped), while trapezoidal rule (bilinear transform) maps $s = \infty$ to z = -1 (no damping introduced)

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Delay-Free Loops

Backward-Euler numerical integrator:

$$v_n = v_{n-1} + T \dot{v}_n$$

Corresponding BE digital mass model:

$$\hat{v}_n = \hat{v}_{n-1} + \frac{T}{m} f_n$$

where \hat{v}_n is the *n*th sample of the estimated velocity, f_n is the driving force at sample *n*, *m* is the mass, and *T* is the sampling interval

• Note that a *delay-free loop* appears if f_n depends on v_n (e.g., due to friction):

$$\hat{v}_n = \hat{v}_{n-1} + \frac{T}{m} f_n(\hat{v}_n)$$

- In such a case, the difference equation is not *computable* in this form
- Non-computable finite-differences schemes such as this are said to be *implicit*
- We can address this by using a *forward-difference* ("Forward Euler")in place of a backward difference

Forward-Euler (FE)

The *backward difference* was based on the usual left-sided limit in the definition of the time derivative:

$$\dot{x}(t) = \lim_{\delta \to 0} \frac{x(t) - x(t - \delta)}{\delta} \approx \frac{x_n - x_{n-1}}{T}$$

The *forward difference* comes from the right-sided limit:

$$\dot{x}(t) = \lim_{\delta \to 0} \frac{x(t+\delta) - x(t)}{\delta} \approx \frac{x_{n+1} - x_n}{T}$$

- As $T \to 0$, the forward and backward difference approximations approach the same limit, because x(t) is assumed continuous and differentiable at t
- The forward difference gives an *explicit finite* difference scheme for the force-driven-mass problem above, even if the driving force f_n depends on current velocity v_n :

$$\hat{v}_{n+1} = \hat{v}_n + \frac{T}{m} f_n, \quad n = 0, 1, 2, \dots$$

with $v_0 \stackrel{\Delta}{=} 0$

• We obtain the same finite-difference scheme by introducing an *ad hoc delay* in the driving force of the Backward Euler scheme to get $\hat{v}_n = \hat{v}_{n-1} + (T/m) f_{n-1}$

$$U_{n-1} + (I/m)J_{n-1}$$

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Trapezoidal Rule for Numerical Integration

The velocity v(t) can be written as

$$v(t) = v(0) + \left(\int_0^t \dot{v}(\tau)d\tau\right)$$

In particular,

$$\begin{aligned} v(nT) &= v(0) + \int_0^{(n-1)T} \dot{v}(\tau) d\tau + \int_{(n-1)T}^{nT} \dot{v}(\tau) d\tau \\ &= v[(n-1)T] + \int_{(n-1)T}^{nT} \dot{v}(\tau) d\tau \\ &\approx v[(n-1)T] + T \frac{\dot{v}[(n-1)T] + \dot{v}(nT)}{2} \end{aligned}$$

- This approximation replaces a one-sample integral by the area under the *trapezoid* having vertices $(n-1,0), (n-1,\dot{v}_{n-1}), (n,0), (n,\dot{v}_n)$
- \bullet In other words, $\dot{v}(t)$ is approximated by a straight line between time n-1 and n
- This is a *first-order* approximation of $\dot{v}(t)$ in contrast to the *zero-order* approximation used by forward and backward Euler schemes

Centered Finite Difference

Backward Euler $[s \leftarrow (1-z^{-1})/T)]$ has a 1/2 sample delay at all frequencies, while Forward Euler $[s \leftarrow (z-1)/T)]$ has a 1/2 sample advance. We can eliminate this time-skew using a centered finite difference:

$$\hat{v}(nT) = \frac{v_{n+1} - v_{n-1}}{2T}$$

$$\Rightarrow \quad f_n \approx \frac{m}{2T}(v_{n+1} - v_{n-1})$$

$$\Rightarrow \quad \hat{v}_{n+1} = \hat{v}_{n-1} + \frac{2T}{m}f_n$$

- No time delay or advance
- Compare the Leapfrog integrator
- s to z mapping is

$$s = \frac{z - z^{-1}}{2T} \rightarrow \frac{e^{j\omega T} - e^{-j\omega T}}{2T} = j \frac{\sin(\omega T)}{T} \approx j\omega$$

at low frequencies, but note how it reaches a maximum at $\omega T=\pi/2$ and comes back down to 0 at $\omega T=\pi$

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- We will see that the commonly used *bilinear transform* is *equivalent*
- Model is *exact* if driving force is *piecewise linear*, having a constant slope over each sampling interval
- (Backward Euler is similarly exact for a *piecewise-constant* driving force)

Bilinear Transform as Compensated BE/FE

In Newton's law $f = m\dot{v}$, look at the Backward Euler (BE) approximation of the time-derivative:

$$f(t) = m \dot{v} \approx m \frac{v(t) - v(t - T)}{T}$$

We see there is a 1/2 sample delay in the first-order difference on the right. This misaligns the force f(t) and subsequent velocity by half a sample. A very simple delay compensation is to use a *two-point average* on the left:

$$\frac{f(n) + f(n-1)}{2} \approx m \frac{v(n) - v(n-1)}{T}$$

The extra attenuation at high frequencies due to the two-point average actually *helps*. Taking the z transform:

$$\frac{1+z^{-1}}{2}F(z) \approx m\frac{1-z^{-1}}{T}V(z)$$

$$F(z) \approx m \left(\frac{2}{T} \frac{1-z^{-1}}{1+z^{-1}}\right) V(z)$$

which is the *bilinear transform* of F(s) = ms V(s):

$s\mapsto$	2	1	_	z^{-1}
	\overline{T}	1	+	z^{-1}

Frequency Warping is the Only Error

We have

$$F(z) \approx m \left(\frac{2}{T}\frac{1-z^{-1}}{1+z^{-1}}\right)V(z)$$

using the *bilinear transform* (*trapezoidal integration* in the time domain)

Let's look along the unit circle in the z plane:

$$\frac{F(e^{j\omega T})}{V(e^{j\omega T})} \approx m \left(\frac{2}{T} \frac{1 - e^{-j\omega T}}{1 + e^{-j\omega T}}\right) = m j \left(\frac{2}{T} \tan\left(\frac{\omega T}{2}\right)\right)$$

Since the exact formula is $F(e^{j\omega T})/V(e^{j\omega T})=m\,j\omega,$ we can push all of the error into a frequency warping:

$$\omega_d \stackrel{\Delta}{=} \frac{2}{T} \tan\left(\frac{\omega_a T}{2}\right)$$

- Frequency-warping is the *only error* over the unit circle when using the bilinear transform
- What started out as different gain errors on the left and right became the correct gains at warped frequency locations
- Frequency-warping implications should also be considered in the time domain

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Filter Design Approach

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We've been talking about the *white-box* approach in which every first-order element (mass, spring, ...) is *explicitly modeled* by a first-order finite-difference scheme. This is especially needed for elements that are *time varying* or pushed into *nonlinear* regimes of operation.

When a system is linear and time-invariant (LTI), there is no need for such fine-grained modeling, and we can take a a *black-box* approach, in which we need only model the *frequency response* from the input(s) to output(s) of the system using a *digital filter*.

Filter Design Approach to Ideal Integrators and Differentiators

Consider the following simple cases:

- Integrator: H(s) = 1/s (e.g., force-driven mass with a velocity output)
- Differentiator: H(s) = s(e.g., force-driven spring with a velocity output)

The *digital filter design formulation* typically minimizes *frequency-response* error with respect to the filter coefficients

Ideal Frequency Responses

• Ideal Digital Integrator

$$H(e^{j\omega T}) = \frac{1}{j\omega}, \quad \omega \in [-\pi/T, \pi/T]$$

• Ideal Digital Differentiator:

$$H(e^{j\omega T}) = j\omega, \quad \omega \in [-\pi/T, \pi/T]$$

- Exact match is not possible in finite order
- Minimize $\| H(e^{j\omega T}) \hat{H}(e^{j\omega T}) \|$ where \hat{H} is the digital filter frequency response and $\| E \|$ denotes some norm of E
- This is a *digital filter design* formulation



- Discontinuity at $z = -1 \Rightarrow$ no exact solution (polynomial approximation over the unit circle)
- Need *oversampling* and a *don't-care band* at high frequencies (e.g., 20 kHz to 22.05 kHz)
- The frequency response can be arbitrary between the upper limit of human hearing (20kHz) and $f_s/2$
- A small increment in oversampling factor yields a large decrease in required filter order for a given spec

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Explicit and Implicit Finite Difference Schemes

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Explicit:

$$y_{n+1} = x_n + f(y_n)$$

Implicit:

$$y_{n+1} = x_n + f(y_{n+1})$$

- A finite difference scheme is said to be *explicit* when it can be computed forward in time using quantities from previous time steps
- We will associate explicit finite difference schemes with *causal digital filters*
- In *implicit* finite-difference schemes, the output of the time-update $(y_{n+1} \text{ above})$ depends on itself, so a causal recursive computation is not specified
- Implicit schemes are generally solved using
 - iterative methods (such as Newton's method) in nonlinear cases, and
 - matrix-inverse methods for linear problems
- Implicit schemes are typically used offline (not in real time)

Semi-Implicit Finite Difference Schemes

- *Implicit* schemes can often be converted to *explicit* schemes (*e.g.*, for real-time usage) by limiting the number of iterations used to solve the implicit scheme
- These are called *semi-implicit finite-difference schemes*
- Iterative convergence is generally improved by working at a very high sampling rate, and by initializing each iteration to the solution for the previous sample
- See the 2009 CCRMA/EE thesis by David Yeh¹ for semi-implicit schemes for real-time computational modeling of nonlinear analog guitar effects (such as overdrive distortion)
- Convex optimization methods can be used to develop powerful new semi-implicit finite-difference schemes: http://www.stanford.edu/~boyd/cvxbook/

¹http://ccrma.stanford.edu/~dtyeh

Recall the mass m sliding on friction μ :

$$f(t) \xrightarrow{x = 0} m$$

$$\mu v(t) \xrightarrow{w} m$$

ODE:

$$f(t) = m \ddot{x}(t) + \mu v(t)$$
$$= m \ddot{x}(t) + \mu \dot{x}(t)$$

Take the Laplace Transform of both sides and apply the *differentiation theorem* (three times):

$$F(s) = m \left[s^2 X(s) - s x(0) - \dot{x}(0) \right] + \mu \left[s X(s) - x(0) \right]$$

= $m s^2 X(s) + \mu s X(s)$

assuming zero initial conditions $x(0) = \dot{x}(0) = 0$.

Force-to-Velocity Transfer Function (often called the "admittance" or "mobility"):

$$H(s) \stackrel{\Delta}{=} \frac{V(s)}{F(s)} = \frac{sX(s)}{F(s)} = \boxed{\frac{1}{ms + \mu}}$$

Bilinear Transform

The *bilinear transform* is a one-to-one mapping from the s plane to the z plane:

$$s = c \frac{1 - z^{-1}}{1 + z^{-1}}, \quad c > 0, \quad c = \frac{2}{T} \quad (\text{typically})$$

$$\Rightarrow z = \frac{1 + s/c}{1 - s/c}$$

Starting with a *continuous-time* transfer function $H_a(s)$, we obtain the *discrete-time* transfer function

$$H_d(z) \stackrel{\Delta}{=} H_a\left(c\frac{1-z^{-1}}{1+z^{-1}}\right)$$

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Properties of the Bilinear Transform

The bilinear transform maps an *s*-plane transfer function $H_a(s)$ to a *z*-plane transfer function:

$$H_d(z) \stackrel{\Delta}{=} H_a\left(c\frac{1-z^{-1}}{1+z^{-1}}\right)$$

We can observe the following *properties* of the bilinear transform:

- Analog dc (s = 0) maps to digital dc (z = 1)
- Infinite analog frequency ($s = \infty$) maps to the maximum digital frequency (z = -1)
- The entire $j\omega$ axis in the *s* plane (where $s \triangleq \sigma + j\omega$) is mapped exactly *once* around the unit circle in the *z* plane (rather than summing around it infinitely many times, or "aliasing" as it does in ordinary sampling)
- *Stability is preserved* (when *c* is real and positive)
- Order of the transfer function is preserved
- Choose c to map any particular finite frequency (such as a resonance frequency) from the $j\omega_a$ axis in the s plane to a particular desired location on the unit circle $e^{j\omega_d}$ in the z plane. Other frequencies are "warped".

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Bilinear Transform of Force-Driven Mass

We have, from $f = m\dot{v} \leftrightarrow F(s) = ms V(s)$,

$$V(s) = \frac{1}{ms}F(s)$$

Setting $s=(2/T)(1-z^{-1})/(1+z^{-1})$ according to the bilinear transform yields

$$V_d(z) = \frac{T}{2m} \frac{1+z^{-1}}{1-z^{-1}} F_d(z)$$

where we defined

$$F_d(z) = F\left(\frac{2}{T}\frac{1-z^{-1}}{1+z^{-1}}\right)$$
$$V_d(z) = V\left(\frac{2}{T}\frac{1-z^{-1}}{1+z^{-1}}\right)$$

The resulting finite-difference scheme is then

$$v_d(n) - v_d(n-1) = \frac{T}{2m} [f_d(n) + f_d(n-1)]$$

i.e.,

$$v_d(n) = v_d(n-1) + \frac{T}{2m} [f_d(n) + f_d(n-1)]$$

We see that this is the same as the backward Euler scheme plus a new term $(T/2m)f_d(n-1)$.

Hybrid Euler-Bilinear Mapping

We can easily interpolate between Backward Euler and Bilinear Transform:

	$1 + \alpha$	$1 - z^{-1}$
$s \rightarrow -$	Т	$\overline{1+\alphaz^{-1}}$

- $\alpha = 0$ gives Backward Euler (high-frequency modes artificially damped)
- $\alpha = 1$ gives Bilinear Transform (high-frequency modes artificially squeezed in frequency)
- \bullet Intermediate α allows optimization of another consideration, such as decay time
- Low-frequency response approximately invariant, dc maps to dc in every case

Example: Leaky Integrator

$$H_{a}(s) = \frac{1}{s+\epsilon} \longrightarrow H_{d}(z) = \frac{1}{\frac{1+\alpha}{T}\frac{1-z^{-1}}{1+\alpha z^{-1}}+\epsilon}$$
$$= g\frac{1+\alpha z^{-1}}{1-pz^{-1}}, \quad \boxed{p = \frac{1-\alpha\frac{\epsilon T}{1+\alpha}}{1+\frac{\epsilon T}{1+\alpha}}}, \quad g = \frac{T}{1+\alpha+\epsilon T}$$

For the Trapezoid Rule (bilinear transform),

$$f_n = m \left. \frac{dv}{dt} \right|_{nT} + \mathcal{O}(T^2)$$

so it is second-order accurate in ${\cal T}$ We will come back to this below

Backward Difference Conformal Map

We saw that the *backwards difference* substitution can be seen as a *conformal map* taking the s plane to the z plane:

	$s \rightarrow$	1	$-z^{-1}$
S			T

Look at the image of the $j\omega$ axis under this mapping:



The continuous-time frequency axis, $s = j\omega$, is not mapped to the discrete-time frequency axis (unit circle):

- dc (s = 0) mapped to dc (z = 1)
- infinite frequency mapped to (z = 0)

This means *artificial damping* will be introduced for high-frequency system resonances

Laplace Analysis of Trapezoidal Rule

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The z transform of the trapezoid rule yields

$$F(z) = \frac{2m}{T} \frac{1 - z^{-1}}{1 + z^{-1}} V(z)$$

Since F(s) = ms V(s), the s to z mapping has become

$$s \to \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}}$$

which is of course the standard bilinear transform:

- $s = j\omega$ axis maps to the |z| = 1 unit circle where it belongs
- dc maps to dc
- Infinite frequency maps to half the sampling rate
- Frequency axis is *warped*, especially at high frequencies
- Stability preserved precisely

Trapezoidal Rule Frequency Mapping

Let's look at the s to z mapping,

$$s = \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}}$$

on the unit circle, where $s = j\omega_a$ and $z = e^{j\omega_d T}$:

$$j\omega_a = \frac{2}{T} \frac{1 - e^{-j\omega_d T}}{1 + e^{-j\omega_d T}} = j\frac{2}{T} \tan(\omega_d T/2)$$

or

 $\frac{\omega_a T}{2} = \tan\left(\frac{\omega_d T}{2}\right)$

• Near dc ($\omega_d = 0$), we have

$$\omega_a = \frac{2}{T} \tan(\omega_d T/2) = \omega_d + \mathcal{O}(T^3)$$

where, since $\tan(\theta)$ is odd, there are no even-order terms in its series expansion

In general, the trapezoid rule is a second-order accurate approximation to a derivative, in the limit of small T (i.e., near dc). Here, it is third-order accurate along the unit circle at dc.

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Why Don't We Always Use the Bilinear Transform?

- Backward Euler (BE) is still sometimes needed:
 - Damps out unwanted high-frequency oscillations (warped)
 - Avoids oscillations at half the sampling rate from a real exponential
 - * TR warps high-frequency poles toward half the sampling rate:

$$s = g' \cdot (1 - z^{-1}) / (1 + z^{-1})$$

toward $z = -1 \leftrightarrow (-1)^n$

* BE warps high-frequency poles toward z = 0 so it *never* introduces alternating-sign oscillations:

$$s = g \cdot (1 - z^{-1})$$

- * Alternating-sign oscillations due to BLT can be problematic in nonlinear circuits such as those containing diodes (see Kurt Werner thesis for a real-world example)
- Recall also that Forward Euler (FE) can break a *delay-free loop*, and pairs well with BE in series

For

$$f(t) = m a(t) = m \dot{v}(t) = \dot{p}(t) = \lim_{T \to 0} \frac{p(t) - p(t - T)}{T} \approx \frac{p(t) - p(t - T)}{T}$$

• Backward Euler (BE)

$$f_n = \frac{1}{T} \left(p_n - p_{n-1} \right)$$

- is $\mathcal{O}(T)$ (first-order accurate in T)
- Bilinear Transform, or Trapezoid Rule (TR)

$$f_n = \frac{2}{T} \left(p_n - p_{n-1} \right) - f_{n-1},$$

is $\mathcal{O}(T^2)$ (second-order accurate in T)

• A continuum of transforms

$$s = \frac{1+\alpha}{T} \frac{1-z^{-1}}{1+\alpha z^{-1}}$$

exists between BE and TR and can be optimized for the application at hand (see Kurt Werner thesis and Germain and Werner DAFx-15 paper for details—Germain thesis coming soon)

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Physical Model Formulations

Reminder of the various kinds of physical model representations we are considering:

- Ordinary Differential Equations (ODE)
- Partial Differential Equations (PDE)
- Difference Equations (DE)
- Finite Difference Schemes (FDS)
- (Physical) State Space Models
- Transfer Functions (between physical signals)
- Modal Representations (Parallel Second-Order Filters)
- Equivalent Circuits
- Impedance Networks
- Wave Digital Filters (WDF)
- Digital Waveguide (DW) Networks

We are mainly concerned with *real-time computational physical models*

State-Space Models

The *state space* formulation replaces an Nth-order ODE by a *vector* first-order ODE.

Review of discrete-time case:

$$\underline{x}(n+1) = \mathbf{A} \underline{x}(n) + \mathbf{B} \underline{u}(n)$$
$$y(n) = \mathbf{C} \underline{x}(n) + \mathbf{D} \underline{u}(n)$$

where

- $\underline{x}(n) \in \mathbb{R}^N = \textit{state vector} \text{ at time } n$
- $\underline{u}(n) = p \times 1$ vector of inputs
- $y(n) = q \times 1$ output vector
- $\mathbf{A} = N \times N$ state transition matrix
- $\mathbf{B} = N \times p$ input coefficient matrix
- $\mathbf{C} = q \times N$ output coefficient matrix
- $\mathbf{D} = q \times p$ direct path coefficient matrix

The state-space representation is especially powerful for

- *multi-input, multi-output* (MIMO) linear systems
- *time-varying* linear systems (every matrix can have a time subscript *n*)

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Continuous-Time State Space Models:

In continuous time, we obtain a first-order vector ODE in which a vector of *state time-derivatives* is driven by linear combinations of state variables:

$$\underline{\dot{x}}(t) = \mathbf{A} \underline{x}(t) + \mathbf{B} \underline{u}(t)$$
$$\underline{y}(t) = \mathbf{C} \underline{x}(t) + \mathbf{D} \underline{u}(t)$$

State-Space Advantages:

- State-space models are used extensively in advanced modeling applications
- Extensive support in Matlab, with many numerically excellent associated tools and techniques (such as the singular value decomposition, to name one)
- Analytically powerful for theory work
- Example: Solution of $\underline{\dot{x}}(t)=\mathbf{A}\,\underline{x}(t)$ is $\underline{x}(t)=e^{At}\,\underline{x}(0),$ where the matrix exponential is defined as

$$e^{\mathbf{A}t} \stackrel{\Delta}{=} I + \mathbf{A}t + \frac{1}{2}\mathbf{A}^2t^2 + \frac{1}{3!}\mathbf{A}^3t^3 + \cdots$$

• We won't do much with state-space modeling in this class, but you should know it exists and that it should be considered for larger, more complex systems than we will be dealing with

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Digitizing State Space Models (Simplistically)

Starting with a continuous-time state-space model

$$\begin{split} \dot{\underline{x}}(t) &= \mathbf{A}\,\underline{x}(t) + \mathbf{B}\,\underline{u}(t) \\ \underline{y}(t) &= \mathbf{C}\,\underline{x}(t) + \mathbf{D}\,\underline{u}(t) \\ \longleftrightarrow \quad s\underline{X}(s) - \underline{x}(0) &= \mathbf{A}\,\underline{X}(s) + \mathbf{B}\,\underline{U}(s) \\ \underline{Y}(s) &= \mathbf{C}\,\underline{X}(s) + \mathbf{D}\,\underline{U}(s) \end{split}$$

we can, e.g., apply Backward Euler, Trapezoidal Rule (Bilinear Transform), or anything in between:

$$s = g \frac{1 - z^{-1}}{1 + \alpha z^{-1}}, \quad \alpha \in [0, 1]$$

to get, letting $g=(1+\alpha)/T$ and defining $\underline{x}_n=\underline{x}(nT)$,

$$\frac{\underline{x}_n - \underline{x}_{n-1}}{T} = \mathbf{A} \left[\frac{\underline{x}_n + \alpha \underline{x}_{n-1}}{1 + \alpha} \right] + \mathbf{B} \left[\frac{\underline{u}_n + \alpha \underline{u}_{n-1}}{1 + \alpha} \right]$$
$$\underline{y}_n = \mathbf{C} \underline{x}_n + \mathbf{D} \underline{u}_n$$

for zero initial conditions $\underline{x}(0) = \underline{0} \Rightarrow$

$$\underline{x}_{n+1} = \left(I - \mathbf{A} \frac{T}{1+\alpha}\right)^{-1} \left(I + \mathbf{A} \frac{\alpha T}{1+\alpha}\right) \underline{x}_n + \left(I - \mathbf{A} \frac{T}{1+\alpha}\right)^{-1} \mathbf{B} T \left(\frac{z+\alpha}{1+\alpha}\right) u_n$$

where $z u_n \stackrel{\Delta}{=} u_{n+1}$

More sophisticated methods will digitize in a manner that conserves energy and/or momentum

Recommended Related Courses at Stanford

- Math 226
- AA 214 A/B/C
- ME 300 A/B/C
- ME 335 A/B/C