Distributed Modeling in Discrete Time

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Outline

- Ideal Vibrating String
- Finite Difference Approximation (FDA)
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- Sampled Traveling Waves versus Finite Differences
- Lossy 1D Wave Equation
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Example One-Dimensional Waveguides

- Any elastic medium displaced along 1D
- Air column of a clarinet or organ pipe
 - Air-pressure deviation $p \leftrightarrow$ string displacement y

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- Longitudinal volume velocity $u \leftrightarrow \mathsf{transverse}\ \mathsf{string}\ \mathsf{velocity}\ v$
- Vibrating strings
 - Really need at least three coupled 1D waveguides:
 - * Horizontally polarized transverse waves
 - * Vertical polarized transverse waves
 - * Longitudinal waves
 - (Typically 1 or 2 WG per string used in practice)
 - Bowed strings also require *torsional waves*
 - (Typical: one waveguide per string [plane of the bow])
 - $-\ensuremath{\mathsf{Piano}}$ requires up to three coupled strings per key
 - * Two-stage decay
 - * Aftersound
 - (Typical: 1 or 2 waveguides per string)

Let's first review the finite difference approximation applied to the ideal string (for comparison purposes):



Wave Equation



Newton's second law

 $\mathsf{Force} = \mathsf{Mass} \times \mathsf{Acceleration}$

Assumptions

- Lossless
- Linear
- Flexible (no "Stiffness")
- $\bullet \ {\rm Slope} \ y'(t,x) \ll 1$

Finite Difference Approximation (FDA)

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and

$$\begin{split} \dot{y}(t,x) &\approx \frac{y(t,x) - y(t-1,x)}{T} \\ y'(t,x) &\approx \frac{y(t,x) - y(t,x-X)}{X} \end{split}$$

(t, m) = (t, T, m)

- T = temporal sampling interval
- X = spatial sampling interval
- \bullet Exact in limit as sampling intervals \rightarrow zero
- Half a sample delay at each frequency. Fix: $\dot{y}(t,x)\approx [y(t+T,x)-y(t-T,x)]/(2T)$

Zero-phase second-order difference:

$$\begin{split} \ddot{y}(t,x) &\approx \frac{y(t+T,x) - 2y(t,x) + y(t-T,x)}{T^2} \\ y''(t,x) &\approx \frac{y(t,x+X) - 2y(t,x) + y(t,x-X)}{X^2} \end{split}$$

• All odd-order derivative approximations suffer a half-sample delay error

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• All even order cases can be compensated as above

FDA of 1D Wave Equation

Substituting finite difference approximation (FDA) into the wave equation $Ky'' = \epsilon i y$ gives

$$\begin{split} & K \frac{y(t,x+X) - 2y(t,x) + y(t,x-X)}{X^2} \\ & \epsilon \frac{y(t+T,x) - 2y(t,x) + y(t-T,x)}{T^2} \end{split}$$

 \Rightarrow Time Update:

$$\begin{array}{ll} y(t+T,x) &=& \frac{KT^2}{\epsilon X^2} \left[y(t,x+X) - 2y(t,x) + y(t,x-X) \right] \\ &\quad + 2y(t,x) - y(t-T,x) \end{array}$$

Let $c \stackrel{\Delta}{=} \sqrt{K/\epsilon}$ (speed of sound along the string). In practice, we typically normalize such that

• $T = 1 \Rightarrow t = nT = n$

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- $X = cT = 1 \Rightarrow x = mX = m$, and $\boxed{y(n+1,m) = y(n,m+1) + y(n,m-1) - y(n-1,m)}$
- Recursive *difference equation* in two variables (time and space)
- Time-update recursion for time n + 1 requires *all* values of string displacement (i.e., all m), for the two preceding time steps (times n and n 1)
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Traveling-Wave Solution

One-dimensional lossless wave equation:

$$Ky'' = \epsilon \ddot{y}$$

Plug in traveling wave to the right:

$$y(t,x) = y_r(t-x/c)$$

$$\Rightarrow \quad y'(t,x) = -\frac{1}{c}\dot{y}(t,x)$$

$$y''(t,x) = -\frac{1}{c^2}\ddot{y}(t,x)$$

- Since c [△] √K/ϵ, the wave equation is satisfied for any shape traveling to the right at speed c (but remember slope ≪ 1)
- Similarly, any $\mathit{left-going}$ traveling wave at speed $c, \ y_l(t+x/c),$ statisfies the wave equation

- Recursion typically started by assuming zero past displacement: $y(n,m) = 0, n = -1, 0, \forall m.$
- Higher order wave equations yield more terms of the form $y(n-l,m-k) \Leftrightarrow$ frequency-dependent *losses* and/or *dispersion* characteristics are introduced into the FDA:
- Linear differential equations with constant coefficients give rise to some linear, time-invariant discrete-time system via the FDA
 - Linear, time-invariant, "filtered waveguide" case:

$$\sum_{k=0}^{\infty} \alpha_k \frac{\partial^k y(t,x)}{\partial t^k} = \sum_{l=0}^{\infty} \beta_l \frac{\partial^l y(t,x)}{\partial x^l}$$

- More general linear, time-invariant case

$$\sum_{k=0}^{\infty}\sum_{l=0}^{\infty}\alpha_{k,l}\frac{\partial^k\partial^l y(t,x)}{\partial t^k\partial x^l} = \sum_{m=0}^{\infty}\sum_{n=0}^{\infty}\beta_{m,n}\frac{\partial^m\partial^n y(t,x)}{\partial t^m\partial x^n}$$

- Nonlinear example:

$$\frac{\partial y(t,x)}{\partial t} = \left(\frac{\partial y(t,x)}{\partial x}\right)^2$$

- Time-varying example:

$$\frac{\partial y(t,x)}{\partial t} = t^2 \frac{\partial y(t,x)}{\partial x}$$

• General solution to lossless, 1D, second-order wave equation:

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$$y(t, x) = y_r(t - x/c) + y_l(t + x/c)$$

- $y_l(\cdot)$ and $y_r(\cdot)$ are arbitrary twice-differentiable functions (slope $\ll 1)$
- Important point: Function of two variables *y*(*t*, *x*) is replaced by two functions of a single (time) variable ⇒ reduced complexity.

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• Published by d'Alembert in 1747

Laplace-Domain Analysis

- $\bullet \ e^{st}$ is an eigenfunction under differentiation
- Plug it in:

$$y(t,x) = e^{st + vx}$$

• By differentiation theorem

• Wave equation becomes

$$Kv^2y = \epsilon s^2y$$
$$\implies \frac{s^2}{v^2} = \frac{K}{\epsilon} = c^2$$
$$\implies v = \pm \frac{s}{c}$$

Thus

$$y(t,x) = e^{s(t \pm x/c)}$$

is a solution for all $\boldsymbol{s}.$

Interpretation: left- and right-going exponentially enveloped complex sinusoids

General eigensolution:

 $y(t,x) = e^{s(t\pm x/c)}, \quad s \text{ arbitrary, complex}$

By superposition,

$$y(t,x) = \sum_{i} A^{+}(s_{i})e^{s_{i}(t-x/c)} + A^{-}(s_{i})e^{s_{i}(t+x/c)}$$

is also a solution for all $A^+(s_i)$ and $A^-(s_i)$.

Alternate derivation of D'Alembert's solution:

- Specialize general eigensolution to $s \stackrel{\Delta}{=} j\omega$
- Extend summation to an integral over ω \Rightarrow Inverse Fourier transform gives

$$y(t,x) = y_r\left(t - \frac{x}{c}\right) + y_l\left(t + \frac{x}{c}\right)$$

where $y_r(\cdot)$ and $y_l(\cdot)$ are arbitrary continuous functions

Infinitely long string plucked simultaneously at three points marked 'p'

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- Initial displacement = sum of two identical triangular pulses
- At time t_0 , traveling waves centers are separated by $2ct_0$ meters
- String is not moving where the traveling waves overlap at same slope.

Sampled Traveling Waves in a String

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For discrete-time simulation, we must *sample* the traveling waves

- Sampling interval $\stackrel{\Delta}{=} T$ seconds
- Sampling rate $\stackrel{\Delta}{=} f_s \operatorname{Hz} = 1/T$
- Spatial sampling rate $\stackrel{\Delta}{=} X \text{ m/s} \stackrel{\Delta}{=} cT$ \Rightarrow systolic grid

For a vibrating string with length L and fundamental frequency f_0 ,

$$c = f_0 \cdot 2L$$
 $\left(\frac{\text{periods}}{\text{sec}} \cdot \frac{\text{meters}}{\text{period}} = \frac{\text{meters}}{\text{sec}}\right)$

so that

$$X = cT = (f_0 2L)/f_s = L[f_0/(f_s/2)]$$

Thus, the number of *spatial samples* along the string is

$$L/X = (f_s/2)/f_0$$

or

Number of spatial samples = Number of string harmonics

Examples:

- Spatial sampling interval for (1/2) CD-quality digital model of Les Paul electric guitar (strings \approx 26 inches long)
 - $-~X = L f_0/(f_s/2) = L82.4/22050 \approx 2.5~\mathrm{mm}$ for low E string
 - $-~X\approx 10~{\rm mm}$ for high E string (two octaves higher and the same length)
 - Low E string: $(f_s/2)/f_0 = 22050/82.4 = 268$ harmonics (spatial samples)
 - High E string: 67 harmonics (spatial samples)
- Number of harmonics = number of oscillators required in *additive synthesis*
- Number of harmonics = number of two-pole filters required in *subtractive, modal,* or *source-filter decomposition synthesis*

Examples (continued):

- Sound propagation in air:
 - Speed of sound $c\approx 331$ meters per second
 - $-\;X=331/44100=7.5\;\mathrm{mm}$
 - Spatial sampling rate = $\nu_s = 1/X = 133~\mathrm{samples/m}$
 - Sound speed in air is *comparable* to that of transverse waves on a guitar string (faster than some strings, slower than others)
 - Sound travels much faster in most solids than in air
 - Longitudinal waves in strings travel faster than transverse waves

Sampled Traveling Waves in any Digital Waveguide

 $\begin{array}{rcl} x & \rightarrow & x_m & = & mX \\ t & \rightarrow & t_n & = & nT \end{array}$

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 \Rightarrow

$$\begin{array}{lll} y(t_n, x_m) &=& y_r(t_n - x_m/c) + y_l(t_n + x_m/c) \\ &=& y_r(nT - mX/c) + y_l(nT + mX/c) \\ &=& y_r\left[(n - m)T\right] + y_l\left[(n + m)T\right] \\ &=& y^+(n - m) + y^-(n + m) \end{array}$$

where we defined

$$y^+(n) \stackrel{\Delta}{=} y_r(nT)$$
 $y^-(n) \stackrel{\Delta}{=} y_l(nT)$

- "+" superscript \implies right-going
- "-" superscript \implies *left-going*
- $y_r\left[(n-m)T\right]=y^+(n-m)=$ output of m-sample delay line with input $y^+(n)$
- $y_l \left[(n+m)T \right] \stackrel{\scriptscriptstyle \Delta}{=} y^-(n+m) = \textit{input}$ to an m-sample delay line whose output is $y^-(n)$

Lossless digital waveguide with observation points at x=0 and x=3X=3cT

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• Recall:

y

$$\begin{array}{rcl} y(t,x) &=& y^+ \left(\frac{t-x/c}{T}\right) + y^- \left(\frac{t+x/c}{T}\right) \\ &\downarrow \\ (nT,mX) &=& y^+(n-m) + y^-(n+m) \end{array}$$

- Position $x_m = mX = mcT$ is eliminated from the simulation
- Position x_m remains laid out from left to right
- Left- and right-going traveling waves must be summed to produce a physical output

$$y(t_n, x_m) = y^+(n-m) + y^-(n+m)$$

• Similar to ladder and lattice digital filters

Important point: Discrete time simulation is *exact* at the sampling instants, to within the numerical precision of the samples themselves. To avoid *aliasing* associated with sampling,

- Require all initial waveshapes be *bandlimited* to $(-f_s/2, f_s/2)$
- Require all external driving signals be similarly bandlimited
- Avoid nonlinearities or keep them "weak"
- Avoid time variation or keep it slow
- Use plenty of lowpass filtering with rapid high-frequency roll-off in severely nonlinear and/or time-varying cases
- Prefer "feed-forward" over "feed-back" around nonlinearities when possible

Relation of Sampled D'Alembert Traveling Waves to the Finite Difference Approximation

Recall FDA result [based on $\dot{x}(n) \approx x(n) - x(n-1)$]:

$$y(n+1,m) = y(n,m+1) + y(n,m-1) - y(n-1,m)$$

Traveling-wave decomposition (exact in lossless case at sampling instants):

$$y(n,m) = y^+(n-m) + y^-(n+m)$$

Substituting into FDA gives

$$\begin{array}{rcl} y(n+1,m) &=& y(n,m+1)+y(n,m-1)-y(n-1,m) \\ &=& y^+(n-m-1)+y^-(n+m+1) \\ && +y^+(n-m+1)+y^-(n+m-1) \\ && -y^+(n-m-1)-y^-(n+m-1) \\ &=& y^-(n+m+1)+y^+(n-m+1) \\ &=& y^+[(n+1)-m]+y^-[(n+1)+m] \\ &\stackrel{\Delta}{=}& y(n+1,m) \end{array}$$

- FDA recursion is also *exact* in the lossless case (!)
- Recall that FDA introduced artificial damping in mass-spring systems
- The last identity above can be rewritten as

$$y(n+1,m) \stackrel{\Delta}{=} y^{+}[(n+1)-m] + y^{-}[(n+1)+m]$$

= $y^{+}[n-(m-1)] + y^{-}[n+(m+1)]$

The Lossy 1D Wave Equation



The ideal vibrating string.

Sources of loss in a vibrating string:

- 1. Yielding terminations
- 2. Drag due to air viscosity
- 3. Internal bending friction

Simplest case: Add a term proportional to velocity:

$$Ky'' = \epsilon \ddot{y} \underbrace{+\mu \dot{y}}_{\text{new}}$$

More generally,

$$Ky'' = \epsilon \ddot{y} + \sum_{\substack{m=0\\m \text{ odd}}}^{M-1} \mu_m \frac{\partial^m y(t,x)}{\partial t^m}$$

where μ_m may be determined *indirectly* by *measuring* linear damping versus frequency

• Displacement at time n + 1 and position m is the superposition of left- and right-going components from positions m - 1 and m + 1 at time n

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- The physical wave variable can be computed for the next time step as the sum of incoming traveling wave components from the left and right
- Lossless nature of the computation is clear

$$y(t,x) = e^{-(\mu/2\epsilon)x/c}y_r(t-x/c) + e^{(\mu/2\epsilon)x/c}y_l(t+x/c)$$

Assumptions:

- Small displacements $(y' \ll 1)$
- Small losses ($\mu \ll \epsilon \omega$)
- $c \stackrel{\Delta}{=} \sqrt{K/\epsilon}$ = as before (wave velocity in lossless case)

Components decay exponentially in direction of travel

Sampling with t = nT, x = mX, and X = cT gives

$$y(t_n, x_m) = g^{-m}y^+(n-m) + g^my^-(n+m)$$

where $q \stackrel{\Delta}{=} e^{-\mu T/2\epsilon}$



- \bullet Order ∞ distributed system reduced to finite order
- Loss factor $g = e^{-\mu T/2\epsilon}$ summarizes distributed loss in one sample of propagation
- Discrete-time simulation exact at sampling points
- Initial conditions and excitations must be bandlimited
- Bandlimited interpolation reconstructs continuous case

Loss Consolidation

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- Loss terms are simply constant gains $g \leq 1$
- Linear, time-invariant elements commute
- Applicable to undriven and unobserved string sections
- Simulation becomes more accurate at the outputs (fewer round-off errors)
- Number of multiplies greatly reduced in practice

Frequency-Dependent Losses

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- Losses in nature tend to *increase* with frequency
 - Air absorption
 - Internal friction
- Simplest string wave equation giving higher damping at high frequencies

$$Ky'' = \epsilon \ddot{y} + \mu_1 \dot{y} + \underbrace{+\mu_3 \frac{\partial^3 y(t,x)}{\partial t^3}}_{\text{new}}$$

- Used in Chaigne-Askenfelt piano string PDE
- Damping asymptotically proportional to ω^2
- Waves propagate with frequency-dependent attenuation (zero-phase filtering)
- Loss consolidation remains valid (by commutativity)



The Dispersive One-Dimensional Wave Equation

Stiffness introduces a restoring force proportional to the fourth spatial derivative:

$$\epsilon \ddot{y} = K y'' \underbrace{-\kappa y''''}_{\text{new}}$$

where

- $\kappa = \frac{Q\pi a^4}{4}$ (moment constant)
- $\bullet a = string radius$
- Q = Young's modulus (stress/strain) (spring constant for solids)
- Stiffness is a *linear* phenomenon
 - Imagine a "bundle" or "cable" of ideal string fibers
 - Stiffness is due to the *longitudinal* springiness

Limiting cases

- Reverts to ideal flexible string at very low frequencies $(Ky'' \gg \kappa y'''')$
- Becomes ideal bar at very high frequencies $(Ky'' \ll \kappa y'''')$

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Digital Simulation of Stiff Strings

- Allpass filters implement a frequency-dependent delay
- \bullet For stiff strings, we must generalize X=cT to

$$X = c(\omega)T \Rightarrow T(\omega) = X/c(\omega) = c_0T_0/c(\omega)$$

where $T_0 = T(0) = \text{zero-stiffness sampling interval}$

• Thus, replace unit delay z^{-1} by

 $z^{-1} \rightarrow z^{-c_0/c(\omega)} \stackrel{\Delta}{=} H_a(z)$ (frequency-dependent delay)

- Each delay element becomes an allpass filter
- In general, $H_a(z)$ is *irrational*
- We approximate $H_a(z)$ in practice using some finite-order fractional delay digital filter



• Phase velocity increases with frequency

$$c(\omega) \stackrel{\Delta}{=} c_0 \left(1 + \frac{\kappa \omega^2}{2Kc_0^2} \right)$$

where $c_0 = \sqrt{K/\epsilon} =$ zero-stiffness phase velocity

- Note ideal-string (LF) and ideal-bar (HF) limits
- Traveling-wave components see a frequency-dependent sound speed
- High-frequency components "run out ahead" of low-frequency components ("HF precursors")
- Traveling waves "disperse" as they travel ("dispersive transmission line")
- String overtones are "stretched" and "inharmonic"
- Higher overtones are progressively sharper (Period(ω) = 2 × Length / $c(\omega$))
- Piano strings are audibly stiff

Reference: L. Cremer: Physics of the Violin

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General Allpass Filters

• General, order *L*, allpass filter:

$$H_{a}(z) \stackrel{\Delta}{=} z^{-L} \frac{A(z^{-1})}{A(z)}$$

= $\frac{a_{L} + a_{L-1}z^{-1} + \dots + a_{1}z^{-(L-1)} + z^{-L}}{1 + a_{1}z^{-1} + a_{2}z^{-2} + \dots + a_{L}z^{-L}}$

• General order *L*, monic, minimum-phase polynomial:

 $A(z) \stackrel{\Delta}{=} 1 + a_1 z^{-1} + a_2 z^{-2} + \dots + a_L z^{-L}$

where $A(z_i) = 0 \Rightarrow |z_i| < 1$ (roots inside unit circle)

- Numerator polynomial = *reverse* of denominator
- First-order case:

$$H_a(z) \stackrel{\Delta}{=} \frac{a_1 z^{-1} + 1}{1 + a_1 z^{-1}}$$

- Each pole p_i gain-compensated by a zero at $z_i = 1/p_i$
- There are papers in the literature describing methods for designing allpass filters with a prescribed *group delay* (see reader for refs)
- For piano strings L is on the order of 10

Allpass filters are *linear and time invariant* which means they *commute* with other linear and time invariant elements



- At least one sample of pure delay must normally be "pulled out" of ideal desired allpass along each rail
- Ideal allpass design minimizes phase-delay error $P_c(\omega)$
- Minimizing $|| P_c(\omega) c_0/c(\omega) ||_{\infty}$ approximately minimizes *tuning error* for modes of freely vibrating string (main audible effect)
- Minimizing group delay error optimizes decay times

- \bullet Online draft of the book^1 containing this material
- \bullet Derivation of the wave equation for vibrating $\mathsf{strings}^2$

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 $^{^{\}rm 1}{\rm http://ccrma.stanford.edu/~jos/waveguide/}^{\rm 2}{\rm http://ccrma.stanford.edu/~jos/waveguide/String_Wave_Equation.html$