1 Bowed String Modeling

The modeling of string instruments was covered earlier in this course. However, a discussion of the bowing mechanism was delayed because of its inherent non-linearity. As we’ll see here, there is much similarity between model approximations for the clarinet reed and the bowed string mechanism.

1.1 The String Model

- As previously discussed, wave propagation along a string of finite length can be efficiently and accurately modeled with digital waveguide techniques.
- The “nut” is assumed ideally rigid, resulting in a reflection coefficient of -1 for displacement or velocity wave components.
- The bridge is assumed to have a finite impedance, so that some energy is transferred from the string to the instrument body.
- Losses along the string cause the string energy to decay (in the absence of a driving mechanism). The losses can be commuted and implemented at discrete positions along the distributed string model.
- The effects of dispersion resulting from string stiffness are less important in bowed string instruments because a driven system is forced to vibrate periodically. However, the natural inharmonicity of the strings will “color” the resulting harmonic sound spectrum.

1.2 The Bowing Mechanism

- Under normal, ideal bowing conditions, the bow and string interaction is referred to as a stick-slip mechanism: During the greater part of each vibration, the string is “stuck” to the bow and is carried with it in its motion. Then the string suddenly detaches itself and moves rapidly backward until it is caught again by the moving bow.
- The beginning and end of the slipping are triggered by the arrival of the propagating bend or “kink”.
- The string’s vertical motion at any one point is given by a sawtooth pattern.
- The round-trip time depends only on the string length and the wave velocity.
- This mechanism allows the player to add energy to the string and to sustain its vibrations.
- Bowing near the string end requires greater force and produces a louder, brighter sound than bowing farther from the end.
- The amplitude of vibration can be increased either by increasing the bow speed or by bowing closer to the bridge.
1.3 The Bow-String Interaction

- An approximation to the friction force exerted by the bow on the string is shown in Fig. 1 (Friedlander, 1953; Keller, 1953). This force is dependent on $v_\Delta = v_b - v_s$, the difference between the bow and string velocities.
- The bow and string are stuck together for $v_\Delta = 0$ (the point of infinite slope in the figure). In this case, the friction force is based primarily on static friction.
- For $|v_\Delta| > 0$, the string is “slipping” and the friction force is based roughly on kinetic friction, which is significantly less than the static friction (especially when rosin is applied to the bow).
- The maximum friction force is roughly proportional to the normal force between the bow and string.
- At all times, the force applied by the bow on the string must balance the reactive force of the string.
- The reactive force can be expressed in terms of the string wave impedance and traveling-wave components of velocity as $f_s = R_s[v_s^+ - v_s^-] = R_s[v_s - 2v_s^-]$, where $R_s$ is the string wave impedance.
- A graphical solution can be found by plotting this expression together with the friction force expression to determine a resulting outgoing traveling-wave component.

1.4 The Bow-String Interaction: A Scattering Approach

- The various components of the model are combined in the general system block diagram of Fig. 2.
- The bowing mechanism effectively divides the string into two parts and is implemented as a nonlinear two-port junction (in contrast to wind instrument reed mechanisms which are one-port junctions).
- As mentioned above, the applied bow force $f_b$ must at all times balance with the reactive force of the string ($f_s = R_s[v_s^+ - v_s^-]$).
The bow friction curve relates the bow force and the differential velocity in terms of a friction coefficient, $f_b(v_\Delta) = R_b(v_\Delta) \cdot v_\Delta$.

Smith [1986] recasts the relationship in terms of a differential velocity of known, incoming junction velocities ($v^+_\Delta$) to allow an expression of the form:

$$R_b(v_\Delta) \cdot v_\Delta = R_s [v^+_\Delta - v^-_\Delta]$$

where $v^+_\Delta = v_b - [v^+_{s,l} + v^+_{s,r}]$.

Traveling-wave components entering the bow-string junction from either side have + superscripts.

Ignoring possible non-zero phase in the bow-hair dynamics, this relationship can be solved simultaneously with the friction curve and represented in terms of a memoryless reflection coefficient as:

$$v^-_{s,r} = v^+_s + \hat{\rho}(v^+_\Delta) \cdot v^+_\Delta$$
$$v^-_{s,l} = v^+_s + \hat{\rho}(v^+_\Delta) \cdot v^+_\Delta$$

where

$$\hat{\rho}(v^+_\Delta) = r(v_\Delta(v^+_\Delta)) \frac{1}{1 + r(v_\Delta(v^+_\Delta))}$$

and $r(v_\Delta) = 0.25R_b(v_\Delta)/R_s$.

An example piece-wise linear reflection coefficient table is shown in Fig. 3.

- When the bow and string are stuck together, the velocity reflection coefficient is 1 and the $v^-_{s,r}$ and $v^-_{s,l}$ are computed so that the physical string velocity is equal to $v_b$.
- A velocity reflection coefficient of 0 corresponds to the absence of the bow discontinuity altogether.
- A complete digital waveguide implementation for the bowed string system, using a reflection coefficient table, is diagrammed in Fig. 4.

1.5 The Body Model

Methods were previously discussed for modeling the string instrument body.

A digital filter representation of a string instrument body must be of high order given the complicated nature of its spectral “signature”.

Commuted synthesis offers an efficient means for making use of a measured or synthesized body impulse response in plucked or struck string synthesis.

Commuted synthesis, however, is possible only with linear systems.

Julius Smith developed a linear system approach to bowed-string synthesis which is excited by a periodic excitation function, thus allowing a commuted-synthesis approach.
1.6 Bowing Control

- In general, the quality of physical modeling synthesis algorithms is highly dependent on physically realistic parameter control. This is especially evident with bowed-string models.


2 Coupled Mode Synthesis

*Coupled Mode Synthesis* is a technique that was first reported by Scott van Duyne in the Proceedings of the 1997 International Computer Music Conference [Van Duyne, 1997]. It provides an efficient technique for modal synthesis that allows easy control over decay rates and natural coupling effects.

2.1 Traditional Modal Synthesis

- Modal synthesis is a technique that can be used to efficiently simulate the sounds of objects that exhibit a relatively small number of exponentially decaying sinusoids, such as impulsively struck bars, plates, or blocks.

- These exponentially decaying sinusoids correspond to vibrating modes of the objects in question.

- In the traditional modal synthesis approach, each of these modes is modeled with a second-order digital filter of the form

\[
\frac{1}{1 - 2r_k \cos(2\pi f_k T_s) z^{-1} + r_k^2 z^{-2}}.
\]

(1)

where the \( f_k \) are the mode frequencies in Hz, the \( r_k \) are the pole radii controlling the decay rates, and \( T_s \) is the sample period.

- These resonant filters are typically connected in parallel, tuned and calibrated from measured recordings, and then excited with an appropriate “dry hit” (and/or residual) signal, as illustrated in Fig. [4]
2.2 A Reformulation

- For reasons to be discussed in the next section, Van Duyne reformulates the second-order filter of Eq. 1 with \( r_k = 1 \) as
  \[
  \frac{1}{1 + z^{-1}H_k(z)},
  \]
  where \( H_k \) is a first order allpass filter of the form
  \[
  H_k(z) = \frac{a_k + z^{-1}}{1 + a_k z^{-1}}
  \]
  and \( a_k = -\cos(2\pi f_k T_s) \).

- With \( r_k = 1 \), the resonator does not decay but rather behaves like an oscillator with frequency \( f_k \).

- Eq. 2 has an additional zero at \(-a_k\) that is not present in Eq. 1, though this has little affect on the resonating behavior of the filter.

- A block diagram of the resulting filter structure is shown in Fig. 6.

2.3 Coupling the Modes

- To “couple” many modes of vibration, we can create a bank of parallel digital resonators and provide a feedback path from their summed output back to each of their inputs that is controlled by a “coupling filter”.

- The filter structure of Fig. 4 however, is problematic in this context because of its direct feedforward path. If the “coupling filter” also has a direct feedforward path, the coupling structure will have a delay-free loop and not be computable.

- Van Duyne thus makes use of a modified filter structure as shown in Fig. 7.
Figure 7: The modified CMS resonator filter structure.

- The filter of Fig. 7 has a transfer function given by
  \[ \frac{z^{-1}H_k(z)}{1 + z^{-1}H_k(z)} \]  
  \hspace{1cm} (4)

- In terms of a magnitude response, the extra allpass filter \( H_k \) and unit delay \( z^{-1} \) in the numerator will have no affect. They will contribute an additional phase component but this will only modify the initial phase of the oscillations.

- With this new resonator structure, the complete coupled mode synthesis block diagram is as shown in Fig. 8.

Figure 8: The coupled mode filter structure.

- Because the individual modal resonators have no attenuation, the “coupling filter” completely controls the modal decay rates.

- The “coupling filter” can be considered akin to a lumped bridge impedance. If the “bridge” is fairly rigid, only a small amount of energy from the modes will be fed back to their inputs (the “coupling filter” will typically have a magnitude response gain on the order of 0.001 or less).

- The “coupling filter” also introduces attenuation because its output is out of phase with the loop signal (the loop signal is subtracted at the loop input summer, while the input is added).

- The [cms.m] Matlab script provides an example implementation of the CMS filter structure.

- The CMS structure has a physical representation as illustrated in Fig. 9 whereby a group of mass-spring resonators are coupled via a common base which itself losses energy via a dashpot.
2.4 Coupling Filter Calibration

- One approach to deriving the “coupling filter” is to determine a “single mode” loop filter, \( L(z) \), whose magnitude response at the various modal frequencies is equal to the respective “per sample” attenuation rates.
- The coupling filter can then be computed as

\[
\frac{2(1 - L(z))}{1 + N + (1 - N)L(z)},
\]

where \( N \) is the number of modes connected to the coupling filter.

2.5 CMS Algorithm Control

- Because the modal decay rates are controlled by a single “coupling filter”, it is easy to make overall sound color modifications.
- For example, Van Duyne suggests that the “coupling filter” is naturally lowpass and that the lowpass cutoff can be modified to produce sounds that range from “wood-like” to “metallic” (in general, metal objects will have more high frequency energy).

2.6 Statistical Mode Modeling

- For sounds composed of many densely packed modes, a modeling of each mode is impractical.
- In such a case, one should choose only the most perceptually important modes to model as above. The remaining densely packed modes can then be statistically modeled using exponentially enveloped noise that is filtered by a lowpass filter whose bandwidth is decreasing over time.
- The time-varying lowpass filter causes the higher frequency energy in the noise to decay faster, a natural property in most struck percussive objects.
- A further refinement can include a strike-dependent lowpass filtering to simulate mallet brightness.
- A block diagram of the full CMS algorithm with statistical mode modeling is shown in Fig. 10.
Figure 10: The CMS algorithm with statistical mode modeling.

References


