# Haskore Music Tutorial 

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## 1 Introduction

Haskore is a collection of Haskell modules designed for expressing musical structures in the high-level, declarative style of functional programming. In Haskore, musical objects consist of primitive notions such as notes and rests, operations to transform musical objects such as transpose and tempo-scaling, and operations to combine musical objects to form more complex ones, such as concurrent and sequential composition. From these simple roots, much richer musical ideas can easily be developed.

Haskore is a means for describing music-in particular Western Music-rather than sound. It is not a vehicle for synthesizing sound produced by musical instruments, for example, although it does capture the way certain (real or imagined) instruments permit control of dynamics and articulation.

Haskore also defines a notion of literal performance through which observationally equivalent musical objects can be determined. From this basis many useful properties can be proved, such as commutative, associative, and distributive properties of various operators. An algebra of music thus surfaces.

In fact a key aspect of Haskore is that objects represent both abstract musical ideas and their concrete implementations. This means that when we prove some property about an object, that property is true about the music in the abstract and about its implementation. Similarly, transformations that preserve musical meaning also preserve the behavior of their implementations. For this reason Haskell is often called an executable specification language; i.e. programs serve the role of mathematical specifications that are directly executable.

Building on the results of the functional programming community's Haskell effort has several important advantages: First, and most obvious, we can avoid the difficulties involved in new programming language design, and at the same time take advantage of the many years
of effort that went into the design of Haskell. Second, the resulting system is both extensible (the user is free to add new features in substantive, creative ways) and modifiable (if the user doesn't like our approach to a particular musical idea, she is free to change it).

In the remainder of this paper I assume that the reader is familar with the basics of functional programming and Haskell in particular. If not, I encourage reading at least A Gentle Introduction to Haskell [HF92] before proceeding. I also assume some familiarity with equational reasoning; an excellent introductory text on this is [BW88].

### 1.1 Acknowledgements

Many students have contributed to Haskore over the years, doing for credit what I didn't have the spare time to do! I am indebted to them all: Amar Chaudhary, Syam Gadde, Bo Whong, and John Garvin, in particular. Thanks also to Alastair Reid for implementing the first Midi-file writer, to Stefan Ratschan for porting Haskore to GHC, and to Matt Zamec for help with the Csound compatibility module. I would also like to express sincere thanks to my friend and talented New Haven composer, Tom Makucevich, for being Haskore's most faithful user.

## 2 The Architecture of Haskore

Figure 1 shows the overall structure of Haskore. Note the independence of high level structures from the "music platform" on which Haskore runs. Originally, the goal was for Haskore compositions to run equally well as conventional midi-files [IMA90], NeXT MusicKit score files [JB91], and csound score files [Ver86], and for Haskore compositions to be displayed and printed in traditional notation using the CMN (Common Music Notation) subsystem. In reality, only one of these platforms is currently supported: midi. It is probably the most popular platform for users, and its use with Haskore is described in detail in this tutorial. ${ }^{1}$

In any case, the independence of abstract musical ideas from the concrete rendering platform is accomplished by having abstract notions of musical object, player, instrument, and performance. All of this resides in the box labeled "Haskore" in the diagram above.

At the module level, Haskore is organized as follows:

```
> module HaskoreLoader (module HaskoreLoader, module Basics,
> module Performance,
> -- module Players,
> module HaskToMidi, module TestHaskore, module ReadMidi)
> where
```

$>$

[^0]

Figure 1: Overall System Diagram

```
> import Basics -- described in Section 3
> import Performance -- described in Section 4
> -- import Players -- described in Section 5
> import HaskToMidi -- described in Section 6
> import TestHaskore
> import ReadMidi
```

This document was written in the literate programming style, and thus the ${ }^{\mathrm{LAT}} \mathrm{E} \mathrm{X}$ manuscript file from which it was generated is an executable Haskell program. It can be compiled under LATEX in two ways: a basic mode provides all of the functionality that most users will need, and an extended mode in which various pieces of lower-level code are provided and documented as well (see file header for details). This version was compiled in extended mode. The document can be retrieved via WWW from: http://haske11. org/haskore (consult the README file for details). It is also delivered with the standard joint Nottingham/Yale Hugs release.

The Haskore code conforms to Haskell 1.4, and has been tested under the June, 1998 release of Hugs 1.4. Unfortunately Hugs does not yet support mutually recursive modules, so all references to the module Players in this document are commented out, which in effect makes it part of module Performance (with which it is mutually recursive).

A final word before beginning: As various musical ideas are presented in this Haskore tutorial, I urge the reader to question the design decisions that are made. There is no supreme theory of music that dictates my decisions, and what I present is actually one of several versions that I have developed over the years (this version is much richer than the one described in [HMGW96]; it is the "Haskore in practice" version alluded to in Section 6 of that paper). I believe that this version is suitable for many practical purposes, but the reader may wish to modify it to better satisfy her intuitions and/or application.

## 3 The Basics

```
> module Basics where
> import Ix
> import Ratio
> infixr 5 :+:, :=:
```

Perhaps the most basic musical idea is that of a pitch, which consists of a pitch class (i.e. one of 12 semi-tones) and an octave:

```
> type Pitch = (PitchClass, Octave)
> data PitchClass = Cf | C | Cs | Df | D | Ds | Ef | E | Es | Ff | F | Fs
```

```
> | Gf | G | Gs | Af | A | As | Bf | B | Bs
> deriving (Eq,Ord,Ix,Show,Read)
> type Octave = Int
```

So a Pitch is a pair consisting of a pitch class and an octave. Octaves are just integers, but we define a datatype for pitch classes, since distinguishing enharmonics (such as G\# and Ab) may be important (especially for notation). By convention, $\mathrm{A} 440=(\mathrm{A}, 4)$.

Musical objects are captured by the Music datatype: ${ }^{2}$

```
> data Music = Note Pitch Dur [NoteAttribute] -- a note \ atomic
> | Rest Dur -- a rest / objects
> | Music :+: Music -- sequential composition
> | Music :=: Music -- paralle1 composition
> | Tempo (Ratio Int) Music -- scale the tempo
> | Trans Int Music -- transposition
> | Instr IName Music -- instrument labe1
> | Player PName Music -- player labe1
> | Phrase [PhraseAttribute] Music -- phrase attributes
> deriving (Show, Eq)
> type Dur = Ratio Int -- in whole notes
> type IName = String
> type PName = String
```

Here a Note is its pitch paired with its duration (in number of whole notes), along with a list of NoteAttributes (defined later). A Rest also has a duration, but of course no pitch or other attributes.

Note that durations are represented as rational numbers; specifically, as ratios of two Haskore Int values. Previous versions of Haskore used floating-point numbers, but rational numbers are more precise (as long as the Int values do not exceed the maximum allowable).

From these two atomic constructors we can build more complex musical objects using the other constructors, as follows:

- $m 1$ :+: $m 2$ is the sequential composition of $m 1$ and $m 2$; i.e. $m 1$ and $m 2$ are played in sequence.
- $m 1:=: m 2$ is the parallel composition of $m 1$ and $m 2$; i.e. $m 1$ and $m 2$ are played simultaneously.

[^1]- Tempo a $m$ scales the rate at which $m$ is played (i.e. its tempo) by a factor of a.
- Trans i m transposes mby interval i (in semitones).
- Instr iname $m$ declares that $m$ is to be performed using instrument iname.
- Player pname m declares that m is to be performed by player pname.
- Phrase pas m declares that m is to be played using the phrase attributes (described later) in the list pas.

It is convenient to represent these ideas in Haskell as a recursive datatype because we wish to not only construct musical objects, but also take them apart, analyze their structure, print them in a structure-preserving way, interpret them for performance purposes, etc.

### 3.1 Convenient Auxiliary Functions

In anticipation of the need to translate between different number types, we define the following coercion functions:

```
> rtof :: Ratio Int -> Float
> rtof r = float (numerator r) / float (denominator r)
>
> float :: Int -> Float
> float = fromInteger . toInteger
```

Treating pitches simply as integers is also useful in many settings, so let's also define some functions for converting between Pitch values and AbsPitch values (integers):

```
> type AbsPitch = Int
>
> absPitch :: Pitch -> AbsPitch
> absPitch (pc,oct) = 12*oct + pitchClass pc
>
> pitch :: AbsPitch -> Pitch
> pitch ap = ([C,Cs,D,Ds,E,F,Fs,G,Gs,A,As,B] !! mod ap 12,
> quot ap 12)
>
> pitchClass :: PitchClass -> Int
> pitchClass pc = case pc of
> Cf -> -1; C -> 0; Cs -> 1 -- or should Cf be 11?
> Df -> 1; D -> 2; Ds -> 3
```

```
> Ef -> 3; E -> 4; Es -> 5
> Ff -> 4; F -> 5; Fs -> 6
> Gf -> 6; G -> 7; Gs -> 8
> Af -> 8; A -> 9; As -> 10
> Bf -> 10; B -> 11; Bs -> 12 -- or should Bs be 0?
```

We can also define a function trans, which transposes pitches (analogous to Trans, which transposes values of type Music):

```
> trans :: Int -> Pitch -> Pitch
> trans i p = pitch (absPitch p + i)
```

Finally, for convenience, let's create simple names for familiar notes, durations, and rests, as shown in Figure 2. Despite the large number of them, these names are sufficiently "unusual" that name clashes are unlikely.

Exercise 1 Show that abspitch . pitch = id, and, up to enharmonic equivalences, pitch . abspitch = id.

Exercise 2 Show that trans $i(t r a n s ~ j p)=t r a n s(i+j) p$.

### 3.2 Some Simple Examples

With this modest beginning, we can already express quite a few musical relationships simply and effectively.

Lines and Chords. Two common ideas in music are the construction of notes in a horizontal fashion (a line or melody), and in a vertical fashion (a chord):

```
> line, chord :: [Music] -> Music
> line = foldr1 (:+:)
> chord = foldr1 (:=:)
```

From the notes in the C major triad in register 4, I can now construct a C major arpeggio and chord as well:

```
> cMaj = [ n 4 qn [] | n <- [c,e,g] ] -- octave 4, quarter notes
>
> cMajArp = line cMaj
> cMajChd = chord cMaj
```

```
cf,c,cs,df,d,ds,ef,e,es,ff,f,fs,gf,g,gs,af,a,as,bf,b,bs ::
        Octave -> Dur -> [NoteAttribute] -> Music
    cf o = Note (Cf,o); c o = Note (C,o); cs o = Note (Cs,o)
    df o = Note (Df,o); d o = Note (D,o); ds o = Note (Ds,o)
    ef o = Note (Ef,o); e o = Note (E,o); es o = Note (Es,o)
> ff o = Note (Ff,o); f o = Note (F,o); fs o = Note (Fs,o)
> gf o = Note (Gf,o); g o = Note (G,o); gs o = Note (Gs,o)
> af o = Note (Af,o); a o = Note (A,o); as o = Note (As,o)
> bf o = Note (Bf,o); b o = Note (B,o); bs o = Note (Bs,o)
>
> bn, wn, hn, qn, en, sn, tn, sfn :: Dur
dwn, dhn, dqn, den, dsn, dtn :: Dur
ddhn, ddqn, dden :: Dur
bnr, wnr, hnr, qnr, enr, snr, tnr :: Music
dwnr, dhnr, dqnr, denr, dsnr, dtnr :: Music
ddhnr, ddqnr, ddenr :: Music
bn = 2 ; bnr = Rest bn -- brevis rest
wn = 1 ; wnr = Rest wn -- whole note rest
hn = 1%2 ; hnr = Rest hn -- half note rest
qn = 1%4 ; qnr = Rest qn -- quarter note rest
en = 1%8 ; enr = Rest en -- eight note rest
sn = 1%16 ; snr = Rest sn -- sixteenth note rest
tn = 1%32 ; tnr = Rest tn -- thirty-second note rest
sfn = 1%64 ; sfnr = Rest sfn -- sixty-fourth note rest
dwn = 3%2 ; dwnr = Rest dwn -- dotted whole note rest
dhn = 3%4 ; dhnr = Rest dhn -- dotted half note rest
dqn = 3%8 ; dqnr = Rest dqn -- dotted quarter note rest
den = 3%16 ; denr = Rest den -- dotted eighth note rest
dsn = 3%32 ; dsnr = Rest dsn -- dotted sixteenth note rest
dtn = 3%64 ; dtnr = Rest dtn -- dotted thirty-second note rest
ddhn = 7%8 ; ddhnr = Rest ddhn -- double-dotted half note rest
ddqn = 7%16 ; ddqnr = Rest ddqn -- double-dotted quarter note rest
dden = 7%32 ; ddenr = Rest dden -- double-dotted eighth note rest
```

Figure 2: Convenient note names and pitch conversion functions.

Delay and Repeat. Suppose now that we wish to describe a melody $m$ accompanied by an identical voice a perfect 5th higher. In Haskore we simply write " $m:=$ : Trans 7 m." Similarly, a canon-like structure involving $m$ can be expressed as " $m:=$ : delay d m," where:

```
> delay :: Dur -> Music -> Music
> delay d m = Rest d :+: m
```

Of course, Haskell's non-strict semantics also allows us to define infinite musical objects. For example, a musical object may be repeated ad nauseum using this simple function:

```
> repeatM :: Music -> Music
> repeatM m = m :+: repeatM m
```

Thus an infinite ostinato can be expressed in this way, and then used in different contexts that extract only the portion that's actually needed.

Inversion and Retrograde. The notions of inversion, retrograde, retrograde inversion, etc. used in 12-tone theory are also easily captured in Haskore. First let's define a transformation from a line created by 1 ine to a list:

```
> lineToList :: Music -> [Music]
> lineToList n@(Rest 0) = []
> lineToList (n :+: ns) = n : lineToList ns
>
> retro, invert, retroInvert, invertRetro :: Music -> Music
> retro = line . reverse . lineToList
> invert m = line (map inv 1)
> where 1@(Note r _ _: _) = lineToList m
> inv (Note p d nas) = Note (pitch (2*(absPitch r) - absPitch p)) d nas
> inv (Rest d) = Rest d
> retroInvert = retro . invert
> invertRetro = invert . retro
```

Exercise 3 Show that "retro . retro," "invert . invert," and "retroInvert . invertRetro" are the identity on values created by line.


Figure 3: Nested Polyrhythms
Polyrhythms. For some rhythmical ideas, consider first a simple triplet of eighth notes; it can be expressed as "Tempo (3\%2) m," where $m$ is a line of three eighth notes. In fact Tempo can be used to create quite complex rhythmical patterns. For example, consider the "nested polyrhythms" shown in Figure 3. They can be expressed quite naturally in Haskore as follows (note the use of the where clause in pr2 to capture recurring phrases):

```
> pr1, pr2 :: Pitch -> Music
> pr1 p = Tempo (5%6) (Tempo (4%3) (mkLn 1 p qn :+:
> Tempo (3%2) (mkLn 3 p en :+:
> mkLn 2 p sn :+:
> mkLn 1 p qn ) :+:
> mkLn 1 p qn) :+:
> Tempo (3%2) (mkLn 6 p en))
>
> pr2 p = Tempo (7%6) (m1 :+:
> Tempo (5%4) (mkLn 5 p en) :+:
> m1 :+:
> mkLn 2 p en)
> where m1 = Tempo (5%4) (Tempo (3%2) m2 :+: m2)
> m2 = mkLn 3 p en
>
> mkLn n p d = line (take n (repeat (Note p d [])))
```

To play polyrhythms pr1 and pr2 in parallel using middle C and middle G, respectively, we would do the following (middle C is in the 5th octave):

```
> pr12 :: Music
> pr12 = pr1 (C,5) :=: pr2 (G,5)
```

Symbolic Meter Changes We can implement a notion of "symbolic meter changes" of the form "oldnote $=$ newnote" (quarter note $=$ dotted eighth, for example) by defining a function:

```
> (=:=) :: Dur -> Dur -> Music -> Music
> old =:= new = Tempo (new/old)
```

Of course, using the new function is not much longer than using Tempo directly, but it may have nemonic value.

Determining Duration It is sometimes desirable to compute the duration in beats of a musical object; we can do so as follows:

```
dur :: Music -> Dur
>
> dur (Note _ d _) = d
> dur (Rest d) = d
> dur (m1 :+: m2) = dur m1 + dur m2
> dur (m1 :=: m2) = dur m1 'max` dur m2
> dur (Tempo a m) = dur m / a
> dur (Trans _ m) = dur m
> dur (Instr _ m) = dur m
> dur (Player _ m) = dur m
> dur (Phrase _ m) = dur m
```

Super-retrograde. Using dur we can define a function revM that reverses any Music value (and is thus considerably more useful than retro defined earlier). Note the tricky treatment of ( $:=:$ ).

```
> revM :: Music -> Music
> revM n@(Note _ _ _) = n
> revM r@(Rest _) = r
> revM (Tempo a m) = Tempo a (revM m)
> revM (Trans i m) = Trans i (revM m)
> revM (Instr i m) = Instr i (revM m)
> revM (Phrase pas m) = Phrase pas (revM m)
> revM (m1 :+: m2) = revM m2 :+: revM m1
> revM (m1 :=: m2) =
> let d1 = dur m1
```

```
> d2 = dur m2
> in if d1>d2 then revM m1 :=: (Rest (d1-d2) :+: revM m2)
> else (Rest (d2-d1) :+: revM m1) :=: revM m2
```

Truncating Parallel Composition Note that the duration of $m 1:=: m 2$ is the maximum of the durations of $\{m 1$ and $m 2$ (and thus if one is infinite, so is the result). Sometimes we would rather have the result be of duration equal to the shorter of the two. This is not as easy as it sounds, since it may require interrupting the longer one in the middle of a note (or notes).

We will define a "truncating parallel composition" operator (/=), but first we will define an auxiliary function cut such that cut $d m$ is the musical object $m$ "cut short" to have at most duration d:

```
> cut :: Dur -> Music -> Music
> cut newDur m | newDur <= 0 = Rest 0
> cut newDur (Note x oldDur y) = Note x (min oldDur newDur) y
> cut newDur (Rest oldDur) = Rest (min oldDur newDur)
> cut newDur (m1 :=: m2) = cut newDur m1 :=: cut newDur m2
> cut newDur (m1 :+: m2) = let m1' = cut newDur m1
>
>
> cut newDur (Tempo a m) = Tempo a (cut (newDur * a) m)
> cut newDur (Trans a m) = Trans a (cut newDur m)
> cut newDur (Instr a m) = Instr a (cut newDur m)
> cut newDur (Player a m) = Player a (cut newDur m)
> cut newDur (Phrase a m) = Phrase a (cut newDur m)
```

Note that cut is equipped to handle a Music object of infinite length.
With cut, the definition of (/=:) is now straightforward:

```
> (/=:) :: Music -> Music -> Music
> m1 /=: m2 = cut (min (dur m1) (dur m2)) (m1 :=: m2)
```

Unfortunately, whereas cut can handle infinite-duration music values, (/=:) cannot.

Exercise 4 Define a version of (/=:) that shortens correctly when either or both of its arguments are infinite in duration.

Trills A trill is an ornament that alternates rapidly between two (usually adjacent) pitches. We will define two versions of a trill function, both of which take the starting note and an interval for the trill note as arguments (the interval is usually one or two, but can actually be anything). One version will additionally have an argument that specifies how long each trill note should be, whereas the other will have an argument that specifies how many trills should occur. In both cases the total duration will be the same as the duration of the original note.

Here is the first trill function:

```
> tril1 :: Int -> Dur -> Music -> Music
> tril1 i sDur (Note p tDur x) =
> if sDur >= tDur then Note p tDur x
> else Note p sDur x
> :+: tril1 (negate i) sDur (Note (trans i p) (tDur-sDur) x)
> tri11 i d (Tempo a m) = Tempo a (tri11 i (d*a) m)
> tri11 i d (Trans a m) = Trans a (tri11 i d m)
> tril1 i d (Instr a m) = Instr a (tri11 i d m)
> trill i d (Player a m) = Player a (trill i d m)
> tril1 i d (Phrase a m) = Phrase a (trill i d m)
> trill _ _ _ = error "Trill input must be a single note."
```

It is simple to define a version of this function that starts on the trill note rather than the start note:

```
> trill' :: Int -> Dur -> Music -> Music
> trill' i sDur m =
> trill (negate i) sDur (Trans i m)
```

The second way to define a trill is in terms of the number of subdivided notes to be included in the trill. We can use the first trill function to define this new one:

```
> trilln :: Int -> Int -> Music -> Music
> trilln i nTimes m =
> trill i (dur m / (nTimes%1)) m
```

This, too, can be made to start on the other note.

```
> tril1n' :: Int -> Int -> Music -> Music
> trilln' i nTimes m =
> trilln (negate i) nTimes (Trans i m)
```

```
> data PercussionSound =
    AcousticBassDrum -- Midi Key 35
    | BassDrum1 -- Midi Key 36
    | SideStick -- ...
    | AcousticSnare | HandClap | ElectricSnare | LowFloorTom
    | ClosedHiHat | HighFloorTom | Peda1HiHat | LowTom
    | OpenHiHat | LowMidTom | HiMidTom | CrashCymbal1
    | HighTom | RideCymba11 | ChineseCymba1 | RideBe11
    | Tambourine | SplashCymba1 | Cowbe11 | CrashCymbal2
    | Vibras7ap | RideCymbal2 | HiBongo | LowBongo
    | MuteHiConga | OpenHiConga | LowConga | HighTimbale
    | LowTimbale | HighAgogo | LowAgogo | Cabasa
    | Maracas | ShortWhistle | LongWhistle | ShortGuiro
    | LongGuiro | Claves | HiWoodBlock | LowWoodBlock
    | MuteCuica | OpenCuica | MuteTriang7e
    | OpenTriangle -- Midi Key }8
deriving (Show,Eq,Ord,Ix,Enum)
```

Figure 4: General Midi Percussion Names

Finally, a ro 11 can be implemented as a trill whose interval is zero. This feature is particularly useful for percussion.

```
> rol1 :: Dur -> Music -> Music
> rolln :: Int -> Music -> Music
>
> roll dur m = trill 0 dur m
> rolln nTimes m = trilln 0 nTimes m
```

Percussion Percussion is a difficult notion to represent in the abstract, since in a way, a percussion instrument is just another instrument, so why should it be treated differently? On the other hand, even common practice notation treats it specially, even though it has much in common with non-percussive notation. The midi standard is equally ambiguous about the treatment of percussion: on one hand, percussion sounds are chosen by specifying an octave and pitch, just like any other instrument, on the other hand these notes have no tonal meaning whatsoever: they are just a conveneient way to select from a large number of percussion sounds. Indeed, part of the General Midi Standard is a set of names for commonly used percussion sounds.

Since Midi is such a popular platform, we can at least define some handy functions for using the General Midi Standard. We start by defining the datatype shown in Figure 4, which borrows its constructor names from the General Midi standard. The comments reflecting the
"Midi Key" numbers will be explained later, but basically a Midi Key is the equivalent of an absolute pitch in Haskore terminology. So all we need is a way to convert these percussion sound names into a Music object; i.e. a Note:

```
> perc :: PercussionSound -> Dur -> [NoteAttribute] -> Music
> perc ds dur na = Note (pitch (fromEnum ds + 35)) dur na
```

For example, here are eight bars of a simple rock or "funk groove" that uses perc and rol1:

```
> funkGroove
    = 1et p1 = perc LowTom qn []
        p2 = perc AcousticSnare en []
        in Tempo 3 (Instr "Drums" (cut 8 (repeatM
            ( (p1 :+: qnr :+: p2 :+: qnr :+: p2 :+:
                p1 :+: p1 :+: qnr :+: p2 :+: enr)
                :=: roll en (perc ClosedHiHat 2 []) )
        )))
```

Exercise 5 Find a simple piece of music written by your favorite composer, and transcribe it into Haskore. In doing so, look for repeating patterns, transposed phrases, etc. and reflect this in your code, thus revealing deeper structural aspects of the music than that found in common practice notation.

Appendix C shows the first 28 bars of Chick Corea’s "Children's Song No. 6" encoded in Haskore.

### 3.3 Phrasing and Articulation

Recall that the Note constructor contained a field of NoteAttributes. These are values that are attached to notes for the purpose of notation or musical interpretation. Likewise, the Phrase constructor permits one to annotate an entire musical object with PhraseAttributes. These two attribute datatypes cover a wide range of attributions found in common practice notation, and are shown in Figure 5. Beware that use of them requires the use of a player that knows how to interpret them! Players will be described in more detail in Section 5.

Note that some of the attributes are parameterized with a numeric value. This is used by a player to control the degree to which an articulation is to be applied. For example, we would expect Legato 1.2 to create more of a legato feel than Legato 1.1. The following constants represent default values for some of the parameterized attributes:

```
data NoteAttribute = Volume Float -- by convention: 0=min, 100=max
    | Fingering Int
    | Dynamics String
        deriving (Eq, Show)
data PhraseAttribute = Dyn Dynamic
                            | Art Articulation
                                Orn Ornament
    deriving (Eq, Show)
data Dynamic = Accent Float | Crescendo Float | Diminuendo Float
    | PPP | PP | P | MP | SF | MF | NF | FF | FFF | Loudness Float
    | Ritardando Float | Accelerando Float
    deriving (Eq, Show)
data Articulation = Staccato Float | Legato Float | Slurred Float
    | Tenuto | Marcato | Pedal | Fermata | FermataDown | Breath
    | DownBow | UpBow | Harmonic | Pizzicato | LeftPizz
    | BartokPizz | Swe11 | Wedge | Thumb | Stopped
    deriving (Eq, Show)
data Ornament = Tri11 | Mordent | InvMordent | Doub7eMordent
    | Turn | TrilledTurn | ShortTril1
    | Arpeggio | ArpeggioUp | ArpeggioDown
    | Instruction String | Head NoteHead
    deriving (Eq, Show)
data NoteHead = DiamondHead | SquareHead | XHead | TriangleHead
    | TremoloHead | SlashHead | ArtHarmonic | NoHead
    deriving (Eq, Show)
```

Figure 5: Note and Phrase Attributes.

```
> legato, staccato :: Articulation
> accent, bigAccent :: Dynamic
>
> legato = Legato 1.1
> staccato = Staccato 0.5
> accent = Accent 1.2
> bigAccent = Accent 1.5
```

To understand exactly how a player interprets an attribute requires knowing how players are defined. Haskore defines only a few simple players, so in fact many of the attributes in Figure 5 are to allow the user to give appropriate interpretations of them by her particular player. But before looking at the structure of players we will need to look at the notion of a performance (these two ideas are tightly linked, which is why the Players and Performance modules are mutually recursive).

## 4 Interpretation and Performance

```
> module Performance (module Performance, module Basics) -- module Players
> where
>
> import Basics
> -- import Players
```

Now that we have defined the structure of musical objects, let us turn to the issue of performance, which we define as a temporally ordered sequence of musical events:

```
> type Performance = [Event]
>
> data Event = Event {eTime :: Time, eInst :: IName, ePitch :: AbsPitch,
> eDur :: DurT, eVol :: Volume, pFields :: [Float]}
> deriving (Eq,Ord,Show)
>
> type Time = Float
> type DurT = Float
> type Volume = Float
```

An event is the lowest of our music representations not yet committed to Midi, csound, or the MusicKit. An event Event $\{\mathrm{eTime}=\mathrm{s}$, eInst $=\mathrm{i}$, ePitch = p, eDur = d, eVol
$=v\}$ captures the fact that at start time $s$, instrument $i$ sounds pitch $p$ with volume $v$ for $a$ duration $d$ (where now duration is measured in seconds, rather than beats).

To generate a complete performance of, i.e. give an interpretation to, a musical object, we must know the time to begin the performance, and the proper volume, key and tempo. We must also know what players to use; that is, we need a mapping from the PNames in an abstract musical object to the actual players to be used. (We don't yet need a mapping from abstract INames to instruments, since this is handled in the translation from a performance into, say, Midi, such as defined in Section 6.)

We can thus model a performer as a function perform which maps all of this information and a musical object into a performance:

```
> perform :: PMap -> Context -> Music -> Performance
>
> type PMap = PName -> Player
> data Context = Context {cTime :: Time, cPlayer :: Player, cInst :: IName,
> cDur :: DurT, cKey :: Key, cVol :: Volume}
> deriving Show
> type Key = AbsPitch
perform pmap c@Context {cTime = t, cPlayer = pl, cDur = dt, cKey = k} m =
    case m of
        Note p d nas -> playNote pl c p d nas
        Rest d -> []
        m1 :+: m2 -> perform pmap c m1 ++
                        perform pmap (c {cTime = t + dur m1 * dt}) m2
        m1 :=: m2 -> merge (perform pmap c m1) (perform pmap c m2)
        Tempo a m -> perform pmap (c {cDur = dt / rtof a}) m
        Trans p m -> perform pmap (c {cKey = k + p}) m
        Instr nm m -> perform pmap (c {cInst = nm}) m
        Player nm m -> perform pmap (c {cPlayer = pmap nm}) m
        Phrase pas m -> interpPhrase pl pmap c pas m
```

Some things to note:

1. The Context is the running "state" of the performance, and gets updated in several different ways. For example, the interpretation of the Tempo constructor involves scaling dt appropriately and updating the DurT field of the context.
2. Interpretation of notes and phrases is player dependent. Ultimately a single note is played by the playNote function, which takes the player as an argument. Similarly, phrase interpretation is also player dependent, reflected in the use of interpPhrase. Precisely how these two functions work is described in Section 5.
```
perform pmap c m = fst (perf pmap c m)
perf :: PMap -> Context -> Music -> (Performance, DurT)
>
> perf pmap c@Context {cTime = t, cPlayer = pl, cDur = dt, cKey = k} m =
> case m of
    Note p d nas -> (playNote pl c p d nas, rtof d *dt)
    Rest d -> ([], rtof d *dt)
    m1 :+: m2 -> let (pf1,d1) = perf pmap c m1
    (pf2,d2) = perf pmap (c {cTime = t+d1}) m2
    in (pf1++pf2, d1+d2)
    m1 :=: m2 -> let (pf1,d1) = perf pmap c m1
                                    (pf2,d2) = perf pmap c m2
                                    in (merge pf1 pf2, max d1 d2)
    Tempo a m -> perf pmap (c {cDur = dt / rtof a}) m
> Trans p m -> perf pmap (c {cKey = k + p}) m
> Instr nm m -> perf pmap (c {cInst = nm}) m
> Player nm m -> perf pmap (c {cPlayer = pmap nm}) m
> Phrase pas m -> interpPhrase pl pmap c pas m
```

Figure 6: The "real" perform function.
3. The DurT component of the context is the duration, in seconds, of one whole note. To make it easier to compute, we can define a "metronome" function that, given a standard metronome marking (in beats per minute) and the note type associated with one beat (quarter note, eighth note, etc.) generates the duration of one whole note:

```
> metro :: Float -> Dur -> DurT
> metro setting dur = 60 / (setting * rtof dur)
```

Thus, for example, metro 96 qn creates a tempo of 96 quarter notes per minute.
4. In the treatment of (:+:), note that the sub-sequences are appended together, with the start time of the second argument delayed by the duration of the first. The function dur (defined in Section 3.2) is used to compute this duration. Note that this results in a quadratic time complexity for perform. A more efficient solution is to have perform compute the duration directly, returning it as part of its result. This version of perform is shown in Figure 6.
5. In contrast, the sub-sequences derived from the arguments to ( $:=:$ ) are merged into a time-ordered stream. The definition of merge is given below.

```
> merge :: Performance -> Performance -> Performance
    merge a@(e1:es1) b@(e2:es2) =
        if e1 < e2 then e1 : merge es1 b
        else e2 : merge a es2
merge [] es2 = es2
merge es1 [] = es1
```

Note that merge compares entire events rather than just start times. This is to ensure that it is commutative, a desirable condition for some of the proofs used in Section 11. Here is a more efficient version that will work just as well in practice:

```
> merge a@(e1:es1) b@(e2:es2) =
    if eTime e1 < eTime e2 then e1 : merge es1 b
    e1se e2 : merge a es2
> merge [] es2 = es2
> merge es1 [] = es1
```


## 5 Players

```
module Players (module Players, module Music, module Performance)
    where
import Music
import Performance
```

In the last section we saw how a performance involved the notion of a player. The reason for this is the same as for real players and their instruments: many of the note and phrase attributes (see Section 3.3) are player and instrument dependent. For example, how should "legato" be interpreted in a performance? Or "diminuendo?" Different players interpret things in different ways, of course, but even more fundamental is the fact that a pianist, for example, realizes legato in a way fundamentally different from the way a violinist does, because of differences in their instruments. Similarly, diminuendo on a piano and a harpsichord are different concepts.

With a slight stretch of the imagination, we can even consider a "notator" of a score as a kind of player: exactly how the music is rendered on the written page may be a personal, stylized process. For example, how many, and which staves should be used to notate a particular instrument?

In any case, to handle these issues, Haskore has a notion of a player which "knows" about differences with respect to performance and notation. A Haskore player is a 4 -tuple consisting of a name and three functions: one for interpreting notes, one for phrases, and one for producing a properly notated score.

```
> data Player = MkPlayer { pName :: PName,
    playNote :: NoteFun,
    interpPhrase :: PhraseFun,
    notatePlayer :: NotateFun }
    deriving Show
> type NoteFun =
    Context -> Pitch -> Dur -> [NoteAttribute] -> Performance
> type PhraseFun =
> PMap -> Context -> [PhraseAttribute] -> Music -> (Performance,DurT)
> type NotateFun = ()
```

The last line above is because notation is currently not implemented. Note that both NoteFun and PhraseFun functions return a Performance (imported from module Perform).

```
defPlayer :: Player
defPlayer = MkPlayer { pName = "Default",
    playNote = defPlayNote defNasHandler,
    interpPhrase = defInterpPhrase defPasHandler,
    notatePlayer = defNotatePlayer () }
defP1ayNote :: (Context->NoteAttribute->Event->Event) -> NoteFun
defPlayNote nasHandler
    c@(Context cTime cPlayer cInst cDur cKey cVol) p d nas =
            [ foldr (nasHandler c)
                        (Event {eTime = cTime, eInst = cInst,
                                    ePitch = absPitch p + cKey,
                                    eDur = rtof d * cDur, eVol = cVol})
        nas ]
defNasHandler :: Context-> NoteAttribute -> Event -> Event
defNasHandler c (Volume v) ev = ev {eVol = (cVol c + v)/2}
defNasHandler _ _ ev = ev
defInterpPhrase :: (PhraseAttribute->Performance->Performance) -> PhraseFun
defInterpPhrase pasHandler pmap context pas m =
            1et (pf,dur) = perf pmap context m
            in (foldr pasHandler pf pas, dur)
defPasHandler :: PhraseAttribute -> Performance -> Performance
defPasHandler (Dyn (Accent x)) pf = map (\e -> e {eVol = x * eVol e}) pf
defPasHandler (Art (Staccato x)) pf = map (\e -> e {eDur = x * eDur e}) pf
defPasHandler (Art (Legato x)) pf = map (\e -> e {eDur = x * eDur e}) pf
defPasHandler _ pf = pf
defNotatePlayer :: () -> NotateFun
defNotatePlayer _ = ()
```

Figure 7: Definition of default Player defP1ayer.

### 5.1 Examples of Player Construction

A "default player" called defPlayer (not to be confused with "deaf player"!) is defined for use when none other is specified in the score; it also functions as a base from which other players can be derived. defPlayer responds only to the Volume note attribute and to the Accent, Staccato, and Legato phrase attributes. It is defined in Figure 7. Before reading this code, recall how players are invoked by the perform function defined in the last section; in particular, note the calls to playNote and interpPhase defined above. Then note:

1. defPlayNote is the only function (even in the definition of perform) that actually generates an event. It also modifies that event based on an interpretation of each note attribute by the function defHasHandler.
2. defNasHandler only recognizes the Volume attribute, which it uses to set the event volume accordingly.
3. defInterpPhrase calls (mutually recursively) perform to interpret a phrase, and then modifies the result based on an interpretation of each phrase attribute by the function defPasHandler.
4. defPasHandler only recognizes the Accent, Staccato, and Legato phrase attributes. For each of these it uses the numeric argument as a "scaling" factor of the volume (for Accent) and duration (for Staccato and Lagato). Thus (Phrase [Legato 1.1] m) effectively increases the duration of each note in $m$ by $10 \%$ (without changing the tempo).

It should be clear that much of the code in Figure 7 can be re-used in defining a new player. For example, to define a player weird that interprets note attributes just like defPlayer but behaves differently with respect to phrase attributes, we could write:

```
weird :: Player
weird = MkPlayer { pname = "Weirdo",
    playNote = defPlayNote defNasHandler,
    interpPhrase = defInterpPhrase myPasHandler
    notatePlayer = defNotatePlayer () }
```

and then supply a suitable definition of myPasHandler. That definition could also re-use code, in the following sense: suppose we wish to add an interpretation for Crescendo, but otherwise have myPasHandler behave just like defPasHandler.

```
myPasHandler :: PhraseAttribute -> Performance -> Performance
myPasHandler (Dyn (Crescendo x)) pf = ...
myPasHandler pa pf = defPasHandler pa pf
```

Exercise 6 Fill in the . . . in the definition of myPasHand7er according to the following strategy: Assume $0<\mathrm{x}<1$. Gradually scale the volume of each event by a factor of 1.0 through $1.0+\mathrm{x}$, using linear interpolation.

Exercise 7 Choose some of the other phrase attributes and provide interpretations of them, such as Diminuendo, Slurred, Tri11, etc. (The trill functions from section 3.2 may be useful here.)

Figure 8 defines a relatively sophisticated player called fancyPlayer that knows all that defPlayer knows, and much more. Note that Slurred is different from Legato in that it doesn't extend the duration of the last note(s). The behavior of (Ritardando $x$ ) can be explained as follows. We'd like to "stretch" the time of each event by a factor from 0 to $x$, linearly interpolated based on how far along the musical phrase the event occurs. I.e., given a start time $t_{0}$ for the first event in the phrase, total phrase duration $D$, and event time $t$, the new event time $t^{\prime}$ is given by:

$$
t^{\prime}=\left(1+\frac{t-t_{0}}{D} x\right)\left(t-t_{0}\right)+t_{0}
$$

Further, if $d$ is the duration of the event, then the end of the event $t+d$ gets stretched to a new time $t_{d}^{\prime}$ given by:

$$
t_{d}^{\prime}=\left(1+\frac{t+d-t_{0}}{D} x\right)\left(t+d-t_{0}\right)+t_{0}
$$

The difference $t_{d}^{\prime}-t^{\prime}$ gives us the new, stretched duration $d^{\prime}$, which after simplification is:

$$
d^{\prime}=\left(1+\frac{2\left(t-t_{0}\right)+d}{D} x\right) d
$$

Acce1erando behaves in exactly the same way, except that it shortens event times rather than lengthening them. And, a similar but simpler strategy explains the behaviors of Crescendo and Diminuendo.

## 6 Midi

Midi ("musical instrument digital interface") is a standard protocol adopted by most, if not all, manufacturers of electronic instruments. At its core is a protocol for communicating musical events (note on, note off, key press, etc.) as well as so-called meta events (select synthesizer patch, change volume, etc.). Beyond the logical protocol, the Midi standard also specifies electrical signal characteristics and cabling details. In addition, it specifies what is known as a standard Midi file which any Midi-compatible software package should be able to recognize.

Over the years musicians and manufacturers decided that they also wanted a standard way to refer to common or general instruments such as "acoustic grand piano," "electric piano," "violin," and "acoustic bass," as well as more exotic ones such as "chorus aahs," "voice oohs," "bird tweet," and "helicopter." A simple standard known as General Midi was developed to fill this role. It is nothing more than an agreed-upon list of instrument names along with a

```
fancyPlayer :: Player
fancyPlayer = MkPlayer { pName = "Fancy",
    playNote = defPlayNote defNasHandler,
    interpPhrase = fancyInterpPhrase,
    notatePlayer = defNotatePlayer () }
fancyInterpPhrase :: PhraseFun
fancyInterpPhrase pmap c [] m = perf pmap c m
fancyInterpPhrase pmap c@Context {cTime = t, cPlayer = pl, cInst = i,
                                    cDur = dt, cKey = k, cVol = v}
            (pa:pas) m =
let pfd@(pf,dur) = fancyInterpPhrase pmap c pas m
    loud x = fancyInterpPhrase pmap c (Dyn (Loudness x) : pas) m
    stretch x = let t0 = eTime (head pf); r = x/dur
                        upd (e@Event {eTime = t, eDur = d}) =
                        1et dt = t-t0
                        t' = (1+dt*r)*dt + t0
                    d' = (1+(2*dt+d)*r)*d
            in e {eTime = t', eDur = d'}
            in (map upd pf, (1+x)*dur)
    inflate x = let t0 = eTime (head pf); r = x/dur
                upd (e@Event {eTime = t, eVol = v}) =
                    e {eVol = (1+(t-t0)*r)*v}
            in (map upd pf, dur)
    in case pa of
    Dyn (Accent x) -> (map (\e-> e {eVol = x * eVol e}) pf, dur)
    Dyn PPP -> loud 40 ; Dyn PP -> loud 50 ; Dyn P -> loud 60
    Dyn MP -> loud 70 ; Dyn SF -> loud 80 ; Dyn MF -> loud 90
    Dyn NF -> loud 100 ; Dyn FF -> loud 110 ; Dyn FFF -> loud 120
    Dyn (Loudness x) -> fancyInterpPhrase pmap c {cVol = x} pas m
    Dyn (Crescendo x) -> inflate x ; Dyn (Diminuendo x) -> inflate (-x)
    Dyn (Ritardando x) -> stretch x ; Dyn (AcceTerando x) -> stretch (-x)
    Art (Staccato x) -> (map (\e-> e {eDur = x * eDur e}) pf, dur)
    Art (Legato x) -> (map (\e-> e {eDur = x * eDur e}) pf, dur)
    Art (Slurred x) ->
            let lastStartTime = foldr (\e t -> max (eTime e) t) 0 pf
                setDur e = if eTime e < lastStartTime
                        then e {eDur = x * eDur e}
                            else e
    in (map setDur pf, dur)
    Art _ -> pfd -- Remaining articulations:
        -- Tenuto | Marcato | Peda1 | Fermata | FermataDown
        -- | Breath | DownBow | UpBow | Harmonic | Pizzicato
        -- | LeftPizz | BartokPizz | Swel1 | Wedge | Thumb | Stopped
                            -> pfd -- Remaining ornamenations:
        -- Tril1 | Mordent | InvMordent | DoubleMordent | Turn
        -- | TrilledTurn | ShortTril1 | Arpeggio | ArpeggioUp
        -- | ArpeggioDown | Instruction String | Head NoteHead
    -- Design Bug: To do these right we need to keep the KEY SIGNATURE
    -- around so that we can determine, for example, what the trill note is.
    -- Alternatively, provide an argument to Trill to carry this info.
```

Figure 8: Definition of Player fancyP1ayer.
program patch number for each, a parameter in the Midi standard that is used to select a Midi instrument's sound.

Most "sound-blaster"-like sound cards on conventional PC's know about Midi, as well as General Midi. However, the sound generated by such modules, and the sound produced from the typically-scrawny speakers on most PC's, is often quite poor. It is best to use an outboard keyboard or tone generator, which are attached to a computer via a Midi interface and cables. It is possible to connect several Midi instruments to the same computer, with each assigned a different channel. Modern keyboards and tone generators are quite amazing little beasts. Not only is the sound quite good (when played on a good stereo system), but they are also usually multi-timbral, which means they are able to generate many different sounds simultaneously, as well as polyphonic, meaning that simultaneous instantiations of the same sound are possible.

If you decide to use the General Midi features of your sound-card, you need to know about another set of conventions known as "Basic Midi." The most important aspect of Basic Midi is that Channel 10 ( 9 in Haskore's 0-based numbering) is dedicated to percussion. A future release of Haskore should make these distinctions more concrete.

Haskore provides a way to specify a Midi channel number and General Midi instrument selection for each IName in a Haskore composition. It also provides a means to generate a Standard Midi File, which can then be played using any conventional Midi software. Finally, it provides a way for existing Midi files to be read and converted into a Music object in Haskore. In this section the top-level code needed by the user to invoke this functionality will be described, along with the gory details.

```
> module HaskToMidi (performToMidi, UserPatchMap,
> module GeneralMidi, module OutputMidi,
> module LoadMidi, module MidiFile)
> where
> import Basics
> import Performance
> import MidiFile
> import GeneralMidi
> import List(partition)
> import Char(toLower,toUpper)
> import OutputMidi
> import LoadMidi
```

Instead of converting a Haskore Performance directly into a Midi file, Haskore first converts it into a datatype that represents a Midi file, which is then written to a file in a separate pass. This separation of concerns makes the structure of the Midi file clearer, makes debugging easier, and provides a natural path for extending Haskore’s functionality with direct Midi capability.

Here is the basic structure of the key modules (*) and functions (=):


A UserPatchMap is a user-supplied table for mapping instrument names (IName's) to Midi channels and General Midi patch names. The patch names are by default General Midi names, although the user can also provide a PatchMap for mapping Patch Names to unconventional Midi Program Change numbers.

```
> type UserPatchMap = [(IName,GenMidiName,MidiChanne1)]
```

See Appendix A for an example of a useful user patch map.
Given a UserPatchMap, a performance is converted to a datatype representing a Standard Midi File using the performToMidi function. If the given UserPatchMap is invalid, it creates a new one by matching the instrument names with General Midi names and mapping the instruments to channels one by one.

```
> performToMidi :: Performance -> UserPatchMap -> MidiFile
> performToMidi pf pMap =
> let splitList = splitByInst pf
> insts = map fst splitList
> rightMap = if (allValid pMap insts)
                                    then pMap
                                    else (makeGMMap insts)
    in MidiFile mfType (Ticks division)
        (map (performToMEvs rightMap) (splitByInst pf))
```

A table of General Midi assignments called genMidiMap is imported from GeneralMidi in Appendix E. The Midi file datatype itself is imported from the module MidiFile, functions for writing it to files are found in the module OutputMidi, and functions for reading MIDI files come from the modules LoadMidi and ReadMidi. All these modules are described later in this section.

The following function is used to test whether or not every instrument in a list is found in a UserPatchMap.

```
> allValid :: UserPatchMap -> [IName] -> Bool
> allValid upm = and . map (lookupB upm)
>
> lookupB :: UserPatchMap -> IName -> Bool
> lookupB [] _ = False
> lookupB ((y,_,_):ys) x = x `partialMatch` y || lookupB ys x
```

If a Haskore user only uses General Midi instrument names as INames, we can define a function that automatically creates a UserPatchMap from these names. Note that, since there are only 15 Midi channels plus percussion, we can handle only 15 instruments. Perhaps in the future a function could be written to test whether or not two tracks can be combined with a Program Change (tracks can be combined if they don't overlap).

```
> makeGMMap :: [IName] -> UserPatchMap
> makeGMMap iList = makeGMMap' O iList
> where makeGMMap' _ [] = []
> makeGMMap' n - | n>=15 =
> error "Too many instruments; not enough MIDI channels."
> makeGMMap' n (i:is) =
> if map toLower i 'elem` percList
> then (i, "Acoustic Grand Piano", 9) : makeGMMap' n is
> else (i, gmMatch i genMidiMap, chanList !! n)
                                    : makeGMMap' (n+1) is
    gmMatch i ((gmInst,chan):gcs) =
    if i 'partialMatch' gmInst -- or use == ?
    then gmInst
    else gmMatch i gcs
    gmMatch i [] = error("instrument " ++ i ++ " not found")
    percList = ["percussion", "perc", "drums"]
    chanList = [0..8] ++ [10..15] -- 10th channe1 (#9) is for percussion
```


### 6.1 The Gory Details

Some preliminaries, otherwise known as constants:

```
> mfType = 1 :: MFType -- midi-file type 1 always used
> division = 96 :: Int -- time-code division: 96 ticks per quarter note
```

Since we are implementing Type 1 Midi Files, we can associate each instrument with a separate track. So first we partition the event list into separate lists for each instrument. (Again, due to the limited number of MIDI channels, we can handle no more than 15 instruments.)

```
> splitByInst :: Performance -> [(IName,Performance)]
> splitByInst [] = []
> splitByInst pf = (i,pf1) : sp1itByInst pf2
> where i = eInst (head pf)
> (pf1,pf2) = partition (\e -> eInst e == i) pf
```

The crux of the conversion process is performToMEvs, which converts a Performance into a stream of MEvents.

```
> performToMEvs :: UserPatchMap -> (IName,Performance) -> [MEvent]
> performToMEvs pMap (inm,perf) =
> let (midiChan,progNum) = unMap pMap inm
> setupInst = MidiEvent 0 (ProgChange midiChan progNum)
> setTempo = MetaEvent 0 (SetTempo defST)
> loop [] = []
> loop (e:es) = let (mev1,mev2) = mkMEvents midiChan e
in mev1 : insertMEvent mev2 (loop es)
    in setupInst : setTempo : loop perf
```

A source of incompatibilty between Haskore and Midi is that Haskore represents notes with an onset and a duration, while Midi represents them as two separate events, an note-on event and a note-off event. Thus MkMEvents turns a Haskore Event into two MEvents, a NoteOn and a NoteOff.

```
> mkMEvents :: MidiChanne1 -> Event -> (MEvent,MEvent)
> mkMEvents mChan (Event {eTime = t, ePitch = p, eDur = d, eVol = v})
> = (MidiEvent (toDelta t) (NoteOn mChan p v'),
> MidiEvent (toDelta (t+d)) (NoteOff mChan p v') )
> where v' = min 127 (round v)
```

The final critical function is insertMEvent, which inserts an MEvent into an already timeordered sequence of MEvents.

```
> insertMEvent :: MEvent -> [MEvent] -> [MEvent]
> insertMEvent mev1 [] = [mev1]
> insertMEvent mev1@(MidiEvent t1 _) mevs@(mev2@(MidiEvent t2 _):mevs') =
> if t1 <= t2 then mev1 : mevs
> else mev2 : insertMEvent mev1 mevs'
```

The following functions lookup IName's in UserPatchMaps to recover channel and program change numbers. Note that the function that does string matching ignores case, and allows substring matches. For example, "chur" matches "Church Organ". Note also that the first match succeeds, so using a substring should be done with care to be sure that the correct instrument is selected.

```
> unMap :: UserPatchMap -> IName -> (MidiChannel,ProgNum)
> unMap pMap iName = (channel, gmProgNum gmName)
> where (gmName, channe1) = lookup23 iName pMap
>
> gmProgNum :: GenMidiName -> ProgNum
> gmProgNum gmName = lookup2 gmName genMidiMap
>
> partialMatch :: String -> String -> Bool
> partialMatch "piano" "Acoustic Grand Piano" = True
> partialMatch s1 s2 =
> let s1' = map toLower s1; s2' = map toLower s2
> len = min (length s1) (length s2)
> in take len s1' == take len s2'
>
> lookup2 x ((y,z):ys) = if x 'partialMatch` y then z else lookup2 x ys
> lookup2 x [] = error ("Instrument " ++ x ++ " unknown")
>
> lookup23 x ((y,z,q):ys) = if x 'partialMatch` y then (z,q) else lookup23 x ys
> lookup23 x [] = error ("Instrument " ++ x ++ " unknown")
```


### 6.2 Midi-File Datatypes

```
> module MidiFile(
> MidiFile(..), Division(..), Track, MFType, MEvent(..), ElapsedTime,
> MPitch, Velocity, ControlNum, PBRange, ProgNum, Pressure,
> MidiChannel, ControlVal,
> MidiEvent(..),
```

> MTempo, SMPTEHours, SMPTEMins, SMPTESecs, SMPTEFrames, SMPTEBits, > MetaEvent(..),

```
data MidiFile = MidiFile MFType Division [Track] deriving (Show, Eq)
>
> data Division = Ticks Int | SMPTE Int Int
> deriving (Show,Eq)
>
type Track = [MEvent]
type MFType = Int
>
> data MEvent = MidiEvent ElapsedTime MidiEvent
> | MetaEvent ElapsedTime MetaEvent
    | NoEvent
    deriving (Show,Eq)
type ElapsedTime = Int
>
> -- Midi Events
>
type MPitch = Int
> type Velocity = Int
> type ControlNum = Int
> type PBRange = Int
> type ProgNum = Int
> type Pressure = Int
> type MidiChanne1 = Int
> type ControlVa1 = Int
>
> data MidiEvent = NoteOff MidiChanne1 MPitch Velocity
> | NoteOn MidiChanne1 MPitch Velocity
> | PolyAfter MidiChanne1 MPitch Pressure
> | ProgChange MidiChanne1 ProgNum
> | Control MidiChannel ControlNum ControlVal
> | PitchBend MidiChanne1 PBRange
```

```
> | MonoAfter MidiChannel Pressure
> deriving (Show, Eq)
>
> -- Meta Events
>
> type MTempo = Int
> type SMPTEHours = Int
> type SMPTEMins = Int
> type SMPTESecs = Int
> type SMPTEFrames = Int
> type SMPTEBits = Int
>
> data MetaEvent = SequenceNum Int
> | TextEvent String
> | Copyright String
> | TrackName String
> | InstrName String
> | Lyric String
> | Marker String
> | CuePoint String
> | MIDIPrefix MidiChannel
> | EndOfTrack
> | SetTempo MTempo
> | SMPTEOffset SMPTEHours SMPTEMins SMPTESecs SMPTEFrames SMPTEBits
> | TimeSig Int Int Int Int
> | KeySig KeyName Mode
> | SequencerSpecific [Int]
> | Unknown String
> deriving (Show, Eq)
```

$>$

The following enumerated type lists all the keys in order of their key signatures from flats to sharps. ( $\mathrm{Cf}=7$ flats, $\mathrm{Gf}=6$ flats $. . . \mathrm{F}=1$ flat, $\mathrm{C}=0$ flats/sharps, $\mathrm{G}=1$ sharp, ... $\mathrm{Cs}=7$ sharps.) Useful for transposition.

```
> data KeyName = KeyCf | KeyGf | KeyDf | KeyAf | KeyEf | KeyBf | KeyF
> | KeyC | KeyG | KeyD | KeyA | KeyE | KeyB | KeyFs | KeyCs
> deriving (Eq, Ord, Ix, Enum, Show)
```

The Key Signature specifies a mode, either major or minor.

```
> data Mode = Major | Minor
> deriving (Show, Eq)
```

Default duration of a whole note, in seconds; and the default SetTempo value, in microseconds per quarter note. Both express the default of 120 beats per minute.

```
> defDurT = 2 :: Float
> defST = truncate (1000000 / defDurT) :: Int
```


## 7 Outputting MIDI Files

The functions in this module allow MidiFiles to be made into Standard MIDI files (*.mid) that can be read and played by music programs such as Cakewalk.

```
> module OutputMidi (outputMidiFile, midiFileToString) where
> import MidiFile
> import IOExtensions (writeBinaryFile)
> import Monads (Output, runO, outO)
> import Bitops (bSplitAt, someBytes)
> import Ix
```

OutputMidiFile is the main function for writing MidiFile values to an actual file; its first argument is the filename:
> outputMidiFile :: String -> MidiFile -> IO ()
> outputMidiFile fn mf = writeBinaryFile fn (midiFileToString mf)

Exercise 8 Take as many examples as you like from the previous sections, create one or more UserPatchMaps, write the examples to a file, and play them using a conventional Midi player.

Appendix A defines some functions which should make the above exercise easier. Appendices B, C, and D contain more extensive examples.

Midi files are first converted to a monadic string computation using the function outMF, and then "executed" using runM :: MidiWriter a -> String.

```
> midiFileToString :: MidiFile -> String
> midiFileToString = runM . outMF
>
outMF :: MidiFile -> MidiWriter ()
outMF (MidiFile mft divisn trks) =
> do
> outChunk "MThd" (do
> out 2 mft -- format (type 0, 1 or 2)
> out 2 (length trks) -- length of tracks to come
> outputDivision divisn) -- time unit
> outputTracks trks
>
outputDivision :: Division -> MidiWriter ()
outputDivision (Ticks nticks) = out 2 nticks
outputDivision (SMPTE mode nticks) = do
out 1 (256-mode)
out 1 nticks
outputTracks :: [Track] -> MidiWriter ()
outputTracks trks = mapM_ outputTrack trks
>
> outputTrack :: Track -> MidiWriter ()
outputTrack trk = outChunk "MTrk" (mapM_ outputEvent (de1ta trk))
```

de1ta converts a track using absolute time to one using delta time, adding EndOfTrack if not already there.

```
delta :: Track -> Track
delta [] = []
delta trk | notEOT (last trk) = trk' ++ [MetaEvent 0 EndOfTrack]
> | otherwise = trk'
> where
> (t,trk') = mscan1 delta' O trk
> delta' :: Int -> -- current time
> MEvent -> -- event
> (Int, -- new time
> MEvent) -- event
> delta' t (MidiEvent dt e) = (dt , MidiEvent (dt-t) e)
> delta' t (MetaEvent dt e) = (dt , MetaEvent (dt-t) e)
> notEOT (MetaEvent _ EndOfTrack) = False
> notEOT _ = True
```

The following functions encode various MidiFile elements into the raw data of a standard MIDI file.

```
> outputEvent :: MEvent -> MidiWriter ()
> outputEvent (MidiEvent dt mevent) = do
> outVar dt
> outputMidiEvent mevent
> outputEvent (MetaEvent dt mevent) = do
> outVar dt
> outputMetaEvent mevent
> outputEvent _ = outStr ""
>
> outputMidiEvent :: MidiEvent -> MidiWriter ()
> outputMidiEvent (NoteOff c p v) = outChan 128 c [p,v]
> outputMidiEvent (NoteOn c p v) = outChan 144 c [p,v]
> outputMidiEvent (PolyAfter c p pr) = outChan 160 c [p,pr]
> outputMidiEvent (Control c cn cv) = outChan 176 c [cn,cv]
> outputMidiEvent (ProgChange c pn) = outChan 192 c [pn]
> outputMidiEvent (MonoAfter c pr) = outChan 208 c [pr]
> outputMidiEvent (PitchBend c pb) = outChan 224 c [lo,hi] -- smal1-endian!!
> where (hi,1o) = bSp1itAt 8 pb
>
> -- output a channe1 event
> outChan :: Int -> MidiChanne1 -> [Int] -> MidiWriter ()
> outChan code chan bytes = do
                                    out 1 (code+chan)
                                    mapM_ (out 1) bytes
outMeta :: Int -> [Int] -> MidiWriter ()
outMeta code bytes = do
out 1 }25
out 1 code
outVar (length bytes)
outList bytes
outMetaStr :: Int -> String -> MidiWriter ()
outMetaStr code bytes = do
out 1 }25
out 1 code
outVar (length bytes)
outStr bytes
> -- As with outChunk, there are other ways to do this - but
```

```
> -- it's not obvious which is best or if performance is a big issue.
> outMetaMW :: Int -> MidiWriter a -> MidiWriter a
outMetaMW code m = do
out 1 255
out 1 code
outVar (mLength m)
m
outputMetaEvent :: MetaEvent -> MidiWriter ()
outputMetaEvent (SequenceNum num) = outMetaMW 0 (out 2 num)
outputMetaEvent (TextEvent s) = outMetaStr 1 s
outputMetaEvent (Copyright s) = outMetaStr 2 s
> outputMetaEvent (TrackName s) = outMetaStr 3 s
> outputMetaEvent (InstrName s) = outMetaStr 4 s
> outputMetaEvent (Lyric s) = outMetaStr 5 s
> outputMetaEvent (Marker s) = outMetaStr 6 s
> outputMetaEvent (CuePoint s) = outMetaStr 7 s
> outputMetaEvent (MIDIPrefix c) = outMeta 32 [c]
> outputMetaEvent EndOfTrack = outMeta 47 []
>
> outputMetaEvent (SetTempo tp) = outMetaMW 81 (out 3 tp)
> outputMetaEvent (SMPTEOffset hr mn se fr ff)
> = outMeta 84 [hr,mn,se,fr,ff]
> outputMetaEvent (TimeSig n d c b) = outMeta 88 [n,d,c,b]
> outputMetaEvent (KeySig sf mi) = outMeta 89 [convert sf, fromMode mi]
>
> convert sf = if (k >= 0) then k
> e1se 255+k
> outputMetaEvent (SequencerSpecific codes)
> = outMeta 127 codes
> outputMetaEvent (Unknown s) = outMetaStr 21 s
```

The midiwriter accumulates a String. For all the usual reasons, the String is represented by ShowS.

```
> type MidiWriter a = Output Char a
>
> out :: Int -> Int -> MidiWriter ()
> outVar :: Int -> MidiWriter ()
> outList :: [Int] -> MidiWriter ()
> outStr :: String -> MidiWriter ()
>
> runM :: MidiWriter a -> String
```

```
> runM m = snd (runO m)
>
> mLength :: MidiWriter a -> Int
> mLength m = length (runM m)
>
> out 1 x = outO [toEnum x]
> out a x = mapM_ (out 1) (someBytes a x)
>
> outStr cs = outO cs
>
> outList xs = outStr (map toEnum xs)
```

Numbers of variable size are represented by sequences of 7-bit blocks tagged (in the top bit) with a bit indicating: (1) that more data follows; or (0) that this is the last block.

```
> outVar n = do
> outVarAux leftover
                out 1 data7
    where (1eftover, data7) = bSplitAt 7 n
        outVarAux 0 = return ()
        outVarAux x = do
            outVarAux leftover'
                            out 1 (128+data7') --make signal bit 1
                where (leftover',data7') = bSplitAt 7 x
    fromMode :: Mode -> Int
    fromMode Major = 0
> fromMode Minor = 1
>
> -- Note: here I've chosen to compute the track twice
> -- rather than store it. Other options are worth exploring.
>
> outChunk :: String -> MidiWriter a -> MidiWriter a
> outChunk tag m | length tag == 4 = do
                                    outStr tag
                                    out 4 (mLength m)
                                    m
```

Mapping scan (used in function delta):


```
    l loc
> mscan1 :: (a -> b -> (a,c)) -> a -> [b] -> (a,[c])
> mscan1 f 1 [] = (1,[])
> mscan1 f 1 (x:xs) = let (m, y ) = f 1 x
> (r,ys) = mscan1 f m xs
> in (r,y:ys)
```


## 8 Loading MIDI Files

The LoadMidi module loads and parses a MIDI File; it can convert it into a MidiFile data type object or simply print out the contents of the file.

```
> module LoadMidi (loadMidiFile, showMidiFile)
> where
> import MidiFile
> import IOExtensions (readBinaryFile)
> import Bitops (fromBytes, bshift1, bTrunc, bSplitAt)
> import Maybe (fromJust)
> import HaskoreUtils (unlinesS, rightS, concatS)
> import Monad
```

The main load function.

```
> loadMidiFile :: String -> IO MidiFile
> loadMidiFile filename =
> do
> contents <- readBinaryFile filename
> case runP parseMidiFile (contents, (AtBeginning,0),-1) of
        Just (mf,"",_,_) -> return mf
        Just (mf,_ ,_,_) -> do {print "Garbage left over." ; return mf}
        Nothing -> do print "Error reading midi file: unfamiliar format or fi
                                    return emptyMidiFile
```

```
>
> emptyMidiFile = MidiFile 0 (Ticks 0) [[]]
```

A MIDI file is made of "chunks," each of which is either a "header chunk" or a "track chunk." To be correct, it must consist of one header chunk followed by any number of track chunks, but for robustness's sake we ignore any non-header chunks that come before a header chunk. The header tells us the number of tracks to come, which is passed to getTracks.

```
> parseMidiFile :: MidiReader MidiFile
> parseMidiFile = do
> chunk <- getChunk
> case chunk of
> Header (format, nTracks, div) ->
> do
> tracks <- getTracks nTracks
> return (MidiFile format div tracks)
> _ -> parseMidiFile
```

Parse a number of track chunks. Like parseMidiFile, if a chunk is not a track chunk, it is just ignored.

```
> getTracks :: Int -> MidiReader [Track]
> getTracks 0 = return []
> getTracks n = do
> chunk <- getChunk
> tracks <- getTracks (n-1)
> case chunk of
> Track t -> return (t:tracks)
> _ -> return tracks
```

Parse a chunk, whether a header chunk, a track chunk, or otherwise. A chunk consists of a four-byte type code (a header is "MThd"; a track is "MTrk"), four bytes for the size of the coming data, and the data itself.

```
> getChunk :: MidiReader Chunk
getChunk = do
> ty <- getN 4
> size <- get4
> setSize size
```

```
> case ty of
> "MThd" -> do
    h <- header
                            return (Header h)
    "MTrk" -> do
                        t <- track
                            return (Track (unde7ta t))
    otherwise -> do
                                    g <- getN size
                                    return AlienChunk
data Chunk = Header (Int, Int, Division)
    | Track Track
    | AlienChunk
deriving Eq
```

Parse a Header Chunk. A header consists of a format ( 0,1 , or 2 ), the number of track chunks to come, and the smallest time division to be used in reading the rest of the file.

```
> header :: MidiReader (Int, Int, Division)
> header = do
> format <- get2
> nTracks <- get2
> div <- getDivision
> return (format, nTracks, div)
```

The division is implemented thus: the most significant bit is 0 if it's in ticks per quarter note; 1 if it's an SMPTE value.

```
> getDivision :: MidiReader Division
> getDivision = do
> x <- get1
> y <- get1
> if x < 128
> then return (Ticks (x*256+y))
> else return (SMPTE (256-x) y)
```

A track is a series of events. Parse a track, stopping when the size is zero.

```
> track :: MidiReader [MEvent]
> track = do
> size <- readSize
> case size of
> 0 -> return []
> _ -> do
> e <- fancyEvent
> es <- track
> return (e:es)
```

Each event is preceded by the delta time: the time in ticks between the last event and the current event. Parse a time and an event, ignoring System Exclusive messages.

```
> fancyEvent :: MidiReader MEvent
> fancyEvent = do
>> >
> case e of
> Midi midiEvent -> return (MidiEvent time midiEvent)
> Meta metaEvent -> return (MetaEvent time metaEvent)
> _ -> return (NoEvent)
```

Parse an event. Note that in the case of a regular Midi Event, the tag is the status, and we read the first byte of data before we call midiEvent. In the case of a MidiEvent with running status, we find out the status from the parser (it's been nice enough to keep track of it for us), and the tag that we've already gotten is the first byte of data.

```
event :: MidiReader MidiFileEvent
event = do
> tag <- get1
> case tag of
> 240 -> do
> size <- getVar
> contents <- getN size
> return (SysexStart
> return (SysexStart contents)
> 247 -> do
> size <- getVar
> contents <- getN size
> return (SysexCont contents)
> 255 -> do
> code <- get1
```

```
> size <- getVar
                                e <- metaEvent code size
                return (Meta e)
    x | x>127 -> do
            firstData <- get1
                    e <- midiEvent (decodeStatus tag) firstData
                                return (Midi e)
                                -> do -- running status
                            s <- readME
                            e <- midiEvent s tag
                            return (Midi e)
data MidiFileEvent = SysexStart String -- F0
    | SysexCont String -- F7
    | Midi MidiEvent
    | Meta MetaEvent
    deriving Show
```

Simpler versions of fancyTrack and fancyEvent, used in the Show functions.

```
plainTrack :: MidiReader [(ElapsedTime, MidiFileEvent)]
> plainTrack = oneOrMore plainEvent
>
> plainEvent :: MidiReader (ElapsedTime, MidiFileEvent)
> plainEvent = do
> time <- getVar
> e <- event
> return (time,e)
>
> data WhichMidiEvent = AtBeginning
| ItsaNoteOff
    | ItsaNoteOn
    | ItsaPolyAfter
    | ItsaControl
    | ItsaProgChange
    | ItsaMonoAfter
    | ItsaPitchBend
    deriving Show
type Status = (WhichMidiEvent, Int)
```

Find out the status (MidiEvent type and channel) given a byte of data.

```
decodeStatus :: Int -> Status
decodeStatus tag = (w, channe1)
    where w = case code of
        8 -> ItsaNoteOff
        9 -> ItsaNoteOn
        10 -> ItsaPolyAfter
        11 -> ItsaControl
        12 -> ItsaProgChange
        13 -> ItsaMonoAfter
        14 -> ItsaPitchBend
        _ -> error "invalid MidiEvent code"
        (code, channe1) = bSplitAt 4 tag
```

Parse a MIDI Event. Note that since getting the first byte is a little complex (there are issues with running status), it has already been handled for us by event.

```
midiEvent :: Status -> Int -> MidiReader MidiEvent
midiEvent s@(wME, channe1) firstData
    = do
        setME s
        case wME of
        ItsaNoteOff -> do v <- get1
        return (NoteOff channel firstData v)
        ItsaNoteOn -> do v <- get1
        case v of
                            0 -> return (NoteOff channe1 firstData 0)
                            _ -> return (NoteOn channel firstData v)
        ItsaPolyAfter -> do v <- get1
        return (PolyAfter channe1 firstData v)
        ItsaContro1 -> do v <- get1
        return (Control channel firstData v)
        ItsaProgChange -> return (ProgChange channel firstData)
        ItsaMonoAfter -> return (MonoAfter channel firstData)
        ItsaPitchBend -> do msb <- get1
        return (PitchBend channe1 (firstData+256*msb))
        AtBeginning -> error "AtBeginning"
```

Parse a MetaEvent.
> metaEvent :: Int -> Int -> MidiReader MetaEvent

```
> metaEvent 0 _ = do x <- get2; return (SequenceNum x)
> metaEvent 1 size = do s <- getN size; return (TextEvent s)
> metaEvent 2 size = do s <- getN size; return (Copyright s)
> metaEvent 3 size = do s <- getN size; return (TrackName s)
> metaEvent 4 size = do s <- getN size; return (InstrName s)
> metaEvent 5 size = do s <- getN size; return (Lyric s)
> metaEvent 6 size = do s <- getN size; return (Marker s)
> metaEvent 7 size = do s <- getN size; return (CuePoint s)
>
> metaEvent 32 _ = do c <- get1; return (MIDIPrefix c)
> metaEvent 47 _ = return (EndOfTrack )
> metaEvent 81 _ = do t <- get3; return (SetTempo t)
> metaEvent 84 _ = do {hrs <- get1 ; mins <- get1 ; secs <- get1;
> frames <- get1 ; bits <- get1 ;
> return (SMPTEOffset hrs mins secs frames bits)}
>
metaEvent 88 _ = do
> n <- get1
> d <- get1
> c <- get1
> b <- get1
> return (TimeSig n d c b)
> metaEvent 89 _ = do
> sf <- get1
    mi <- get1
    return (KeySig (toKeyName sf) (toMode mi))
    metaEvent 127 size = do
    contents <- getN size
    return (SequencerSpecific (map fromEnum contents))
metaEvent _ size = do {s <- getN size; return (Unknown s)}
>
> toKeyName sf = [ KeyCf .. ] !! ((sf+7) `mod` 15)
```

Convert a track using delta time to one using absolute time.

```
> undelta :: Track -> Track
> undelta trk = undelta' trk 0
> where
> undelta' :: Track -> Int -> Track
```

```
> unde7ta' [] _ = []
> undelta' ((MidiEvent dt e):es) t = (MidiEvent (dt+t) e):undelta' es (dt+t)
> undelta' ((MetaEvent dt e):es) t = (MetaEvent (dt+t) e):undelta' es (dt+t)
> undelta' (_:es) t = undelta' es t
>
toMode :: Int -> Mode
toMode 0 = Major
toMode 1 = Minor
```

getCh gets a single character (a byte) from the input.
getCh :: MidiReader Char
getCh = do \{sub1Size; tokenP myHead\}
where myHead ([] ,_ ,_ ) = Nothing
myHead ((c:cs),st,sz) = Just (c,cs,st,sz)
getN n returns n characters (bytes) from the input.

```
> getN :: Int -> MidiReader String
> getN 0 = return []
> getN n = do
> a <- getCh
> b <- (getN (n-1))
> return (a:b)
```

get1, get2, get3, and get4 take 1-, 2-, 3-, or 4-byte numbers from the input (respectively), convert the base-256 data into a single number, and return.

```
> get1 :: MidiReader Int
get1 = do
> c <- getCh
> return (fromEnum c)
>
> get2 :: MidiReader Int
> get2 = do
> x1 <- get1
> x2 <- get1
> return (fromBytes [x1,x2])
>
```

```
get3 :: MidiReader Int
get3 = do
x1 <- get1
    x2 <- get1
    x3 <- get1
    return (fromBytes [x1,x2,x3])
get4 :: MidiReader Int
get4 = do
    x1 <- get1
    x2 <- get1
    x3 <- get1
    x4 <- get1
    return (fromBytes [x1,x2,x3,x4])
```

Variable-length quantities are used often in MIDI notation. They are represented in the following way. Each byte (containing 8 bits) uses the 7 least significant bits to store information. The most significant bit is used to signal whether or not more information is coming. If it's 1 , another byte is coming. If it's 0 , that byte is the last one. getVar gets a variable-length quantity from the input.

```
> getVar :: MidiReader Int
> getVar = getVarAux 0
> where getVarAux n = do
> digit <- get1
> if digit < 128 -- if it's the last byte
> then return ((bshift1 7 n) + digit)
> else getVarAux ((bshift1 7 n) + (bTrunc 7 digit))
```

Functions to show the decoded contents of a Midi file in an easy-to-read format.

```
> showMidiFile :: String -> IO ()
> showMidiFile file = do
> s <- readBinaryFile file
> putStr (showChunks s)
>
> showChunks :: String -> String
> showChunks mf = showMR chunks (unlinesS . map pp) (mf, (AtBeginning,0),-1) ""
> where
> pp :: (String, String, Status, Int) -> ShowS
> pp ("MThd",contents,st,sz) =
```

```
> showString "Header: " .
> showMR header shows (contents,st,sz)
> pp ("MTrk",contents,st,sz) =
    showString "Track:\n" .
    showMR plainTrack (un7inesS . map showTrackEvent) (contents,st,sz)
    pp (ty,contents,_,_) =
        showString "Chunk: " .
    showString ty .
    showString " " .
    shows (map fromEnum contents) .
    showString "\n"
showTrackEvent :: (ElapsedTime, MidiFileEvent) -> ShowS
showTrackEvent (t,e) =
    rightS 10 (shows t) . showString " : " . showEvent e
showEvent :: MidiFileEvent -> ShowS
showEvent (Midi e) =
    showString "MidiEvent " .
    shows e
showEvent (Meta e) =
    showString "MetaEvent " .
    shows e
showEvent (SysexStart s) =
    showString "SysexStart " . concatS (map (shows.fromEnum) s)
showEvent (SysexCont s) =
    showString "SysexCont " . concatS (map (shows.fromEnum) s)
showMR :: MidiReader a -> (a->ShowS) -> (String, Status, Int) -> ShowS
showMR m pp (s,st,sz) =
    case runP m (s,st,sz) of
> Nothing -> showString "Parse failed: " . shows (map fromEnum s)
> Just (a,[] ,_,_) -> pp a
> Just (a,junk,_,_) -> pp a . showString "Junk: " . shows (map fromEnum junk)
```

These two functions, the plainChunk and chunks parsers, do not combine directly into a single master parser. Rather, they should be used to chop parts of a midi file up into chunks of bytes which can be outputted separately.

Chop a Midi file into chunks returning:

- list of "chunk-type"-contents-running status triples; and
- leftover slop (should be empty in correctly formatted file)

```
chunks :: MidiReader [(String, String, Status, Int)]
chunks = zeroOrMore plainChunk
plainChunk :: MidiReader (String, String, Status, Int)
plainChunk = do
    ty <- getN 4 -- chunk type: header or track
    size <- get4 -- size of what's next
    contents <- getN size -- what's next
    status <- readME -- running status
    return (ty, contents, status, -1) -- Don't worry about size
```

The following parser monad parses a Midi File. As it parses, it keeps track of these things:

- (w, c) a.k.a. st Running status. In MIDI, a shortcut is used for long strings of similar MIDI events: if a stream of consecutive events all have the same type and channel, the type and channel can be ommitted for all but the first event. To implement this "feature," the parser must keep track of the type and channel of the most recent Midi Event.
- sz The size, in bytes, of what's left to parse, so that it knows when it’s done.

```
type MidiReader a = Parser String WhichMidiEvent Int Int a
data Parser s w c sz a = P ((s,(w,c),sz) -> Maybe (a,s,(w,c),sz))
unP :: Parser s w c sz a -> ((s,(w,c),sz) -> Maybe (a,s,(w,c),sz))
unP (P a) = a
-- Access to state
tokenP :: ((s,(w,c),sz) -> Maybe (a,s,(w,c),sz)) -> Parser s w c sz a
runP :: Parser s w c sz a -> (s,(w,c),sz) -> Maybe (a,s, (w,c),sz)
tokenP get = P $ get
runP m (s,st,sz) = (unP m) (s,st,sz)
instance Monad (Parser s w c sz) where
    m >= k = P $ \ (s,st,sz) -> do
                                    (a,s',st',sz') <- unP m (s,st,sz)
                                    unP (k a) (s',st',sz')
    m > k = P $ \ (s,st,sz) -> do
                                    (_,s',st',sz') <- unP m (s,st,sz)
                                    unP k (s',st',sz')
return a = P $ \ (s,st,sz) -> return (a,s,st,sz)
```

```
>
setME :: Status -> MidiReader ()
setME st' = P $ \ (s,st,sz) -> return ((),s,st',sz)
readME :: MidiReader Status
readME = P $ \ (s,st,sz) -> return (st,s,st,sz)
>
setSize :: Int -> MidiReader ()
setSize sz' = P $ \ (s,st,sz) -> return ((),s,st,sz')
sub1Size :: MidiReader ()
sub1Size = P $ \ (s,st,sz) -> return ((),s,st,(sz-1))
readSize :: MidiReader Int
readSize = P $ \ (s,st,sz) -> return (sz,s,st,sz)
-- instance MonadZero (Parser s w c sz) where
instance MonadPlus (Parser s w c sz) where
    mzero = P $ \ (s,st,sz) -> mzero
    p `mplus` q = P $ \ (s,st,sz) -> unP p (s,st,sz) `mplus` unP q (s,st,sz)
-- Wadler's force function
force :: Parser s w c sz a -> Parser s w c sz a
force (P p) = P $ \ (s,st,sz) -> let x = p (s,st,sz)
                                    in Just (fromJust x)
zeroOrMore :: Parser s w c sz a -> Parser s w c sz [a]
zeroOrMore p = force (oneOrMore p `mplus` return [])
oneOrMore :: Parser s w c sz a -> Parser s w c sz [a]
oneOrMore p = do {x <- p; xs <- zeroOrMore p; return (x:xs)}
```


## 9 Reading Midi files

Now that we have translated a raw Midi file into a MidiFile data type, we can translate that MidiFile into a Music object.

```
> module ReadMidi (readMidi, module Basics, module Performance,
> module MidiFile)
> where
```

```
>
> import Ratio
> import Basics
> import Performance
> import MidiFile
> import HaskToMidi
> import GeneralMidi
```

The main function. Note that we output a Context and a UserPatchMap as well as a Music object.

```
> readMidi :: MidiFile -> (Music, Context, UserPatchMap)
readMidi mf@(MidiFile _ d trks) =
> let trk1 = removeEOTs $ format mf
> upm = makeUPM trks
> (trk2,dur) = getFirst trk1
> trksrs = getRest dur (newTracks dur (splitBy (isTempoChg) trk2))
> read (t,r) = Tempo r (readTrack (tDiv d) upm t)
> m = optimize $ line $ map read trksrs
> in (m,
> Context {cTime = 0,
> CPlayer = fancyPlayer,
> cInst = getInst upm,
> cDur = float dur / 250000,
> cKey = 0,
> cVo1 = 100},
> upm)
```

Remove the EndOfTrack's that end up in the middle of merged or sequenced tracks.

```
> removeEOTs :: Track -> Track
> removeEOTs ((MetaEvent _ EndOfTrack):es) = removeEOTs es
> removeEOTs (e:es) = e:(removeEOTs es)
> removeEOTs [] = []
```

Make one big track out of the individual tracks of a MidiFile, using different methods depending on the format of the MidiFile.
> format (MidiFile 0 _ [trk]) $=$ trk

```
> format (MidiFile 1 _ trks) = mergeTracks trks
> format (MidiFile 2 _ trks) = seqTracks trks
> format _ = error ("readMidi: unfamiliar format or file corrupt")
```

Used for a list of simultaneous tracks (format 1). It merges together the events of all the tracks, putting the events in order by their elapsed time. A generalization of this algorithm could be useful as a merge of an arbitrary number of lists.

```
> mergeTracks :: [Track] -> Track
> mergeTracks [[]] = []
> mergeTracks ([]:trks) = mergeTracks trks
> mergeTracks trks = 1et minTs = fold11 minT trks
>> >
> minT (trk1@(e1:e1s)) (trk2@(e2:e2s)) =
                                if getTime e1 <= getTime e2 then trk1 e1se trk2
    newList (trk@(e:es):trks) trk'
        | e == (head trk') = es:trks
            | otherwise = trk:(newList trks trk')
    newList ([]:trks) trk' = []:newList trks trk'
    newList [[]] _ = [[]]
        in
        (head (minTs)) : mergeTracks (newList trks (minTs))
```

Used for a list of sequential tracks (format 2). It puts the events of the tracks in order one after the other. About its auxiliary functions: seqTracks' is the same as seqTracks, only it keeps a running total of the time increment; incTrack increments the elapsed-time of each event in the given track; incEvent increments the elapsed-time of a single event.

```
> seqTracks :: [Track] -> Track
> seqTracks trks = seqTracks' 0 trks
> where
> seqTracks' _ [] = []
> seqTracks' _ [[]] = []
> seqTracks' n ([]:trks) = seqTracks' n trks
> seqTracks' n (trk:trks) = a ++ seqTracks' (b+n) trks
> where (a,b) = incTrack n trk
> incTrack n [] = error ("incTrack: Empty list")
> incTrack n [e] = ([incEvent n e],getTime e)
> incTrack n (e:es) = ((incEvent n e):a, b)
> where (a,b) = incTrack n es
```

```
> incEvent n (MidiEvent t x) = MidiEvent (t+n) x
> incEvent n (MetaEvent t x) = MetaEvent ( }\textrm{t}+\textrm{n}\mathrm{ ) x
```

Get the elapsed time of an event.

```
> getTime :: MEvent -> ElapsedTime
getTime (MidiEvent t _) = t
getTime (MetaEvent t _) = t
```

Look through the given tracks, using Program Changes to make a UserPatchMap.

```
makeUPM :: [Track] -> UserPatchMap
makeUPM trks = removeDups $ map makeTriple $ concatMap searchPC trks
    where makeTriple (ch,num) = let gmName = lookupInst genMidiMap num
                                    iName = case ch of
                                    9 -> "drums"
                                    _ -> gmName
                            in (iName, gmName, ch)
        searchPC ((MidiEvent _ (ProgChange ch num)):es) = (ch, num) : searchPC es
        searchPC ((MidiEvent t (NoteOn _ _ _)):es) | t > 0 = []
        searchPC (e:es) = searchPC es
        searchPC [] = []
        searchInst ((MetaEvent _ (InstrName iName)):es) = Just iName
        searchInst ((MidiEvent t (NoteOn _ _ _)):es) | t > 0 = Nothing
        searchInst (e:es) = searchInst es
        searchInst [] = Nothing
```

Remove consecutive duplicates from a list, used in this case to take out redundant elements in the UserPatchMap.

```
> removeDups :: Eq a => [a] -> [a]
> removeDups [] = []
> removeDups [x] = [x]
> removeDups (x:y:xys) | x == y = removeDups (y:xys)
> | otherwise = x : removeDups (y:xys)
```

Translate Divisions into the number of ticks per quarter note.

```
> tDiv :: Division -> Int
> tDiv (Ticks x) = x
> tDiv (SMPTE _ _) = error "Sorry, SMPTE not yet implemented."
```

getFirst gets the information that occurs at the beginning of the piece: the default tempo and the default key signature. A SetTempo in the middle of the piece should translate to a tempo change (Tempo $\times$ y m), but a SetTempo at time 0 should set the default tempo for the entire piece, by translating to Context tempo. getFirst takes care of all events that occur at time 0 so that if any SetTempo appears at time 0 , it can translate to Context even if it is not the very first event.

```
> getFirst :: Track -> (Track, Int)
> getFirst trk1 = getFirst' trk1 []
> where
> getFirst' :: Track -> Track -> (Track, Int)
> getFirst' ((MetaEvent 0 (SetTempo tempo)):es) tAcc = (((reverse tAcc)++es), tempo)
> getFirst' (e@(MetaEvent 0 _):es) tAcc = getFirst' es (e:tAcc)
> getFirst' (e@(MidiEvent 0 _):es) tAcc = getFirst' es (e:tAcc)
> getFirst' es tAcc = (((reverse tAcc)++es), defST)
> getFirst' [] _ = ([] , defST)
```

Manages the tempo changes in the piece; it translates each MidiFile SetTempo into a ratio between the new tempo and the tempo at the beginning.

```
> getRest :: Int -> [Track] -> [(Track,Ratio Int)]
> getRest d (((MetaEvent _ (SetTempo tempo)):es):trks) = (es,r) : getRest d trks
> where r = d % tempo
> getRest d (trk:trks) = (trk,1) : getRest d trks
> getRest _ [] = []
```

Get the first instrument from the UserPatchMap, to use as the default in the Context.

```
> getInst :: UserPatchMap -> String
> getInst ((iName, gmName, channe1):xs) = iName
> getInst [] = "piano"
```

splitBy takes a boolean test and a list; it divides up the list and turns it into a list of sublists; each sub-list consists of (1) one element for which the test is true (or the first element in the list), and (2) all elements after that element for which the test is false. Used to split a track into sub-tracks by tempo. For example, splitBy (>10) [27, 0, 2, 1, 15, 3, 42, 4] yields [ [27,0,2,1], [15,3], [42,4] ].

```
> splitBy :: (a -> Bool) -> [a] -> [[a]]
> splitBy test xs = splitBy' test xs []
> where splitBy' :: (a -> Bool) -> [a] -> [a] -> [[a]]
> splitBy' test (x1:x2:xs) acc =
> if (test x2)
> then (reverse (x1:acc)) : (sp1itBy' test (x2:xs) [])
> else splitBy' test (x2:xs) (x1:acc)
> splitBy' test [x] acc = [(reverse (x:acc))]
> splitBy' _ [] _ = [[]]
> isTempoChg :: MEvent -> Bool
> isTempoChg (MetaEvent _ (SetTempo _)) = True
> isTempoChg _ = False
```

readTrack is the heart of the readMidi operation. It reads a track that has been "processed" by newTracks, and returns the track as Music.

```
> readTrack :: Int -> UserPatchMap -> Track -> Music
> readTrack _ _ []
> = Rest 0
> readTrack ticks upm (nOn @(MidiEvent t1 (NoteOn ch p v))
    : nOff@(MidiEvent t2 (NoteOff _ _ _))
    : nOn2@(MidiEvent t3 (NoteOn _ _ _))
    : es)
    | (t2 < t1) || (t3 < t1) = error "readTrack: elapsed time values out of order"
    | t2 == t3 = n :+: readTrack ticks upm (nOn2:es)
    | t1 == t3 = n :=: readTrack ticks upm (nOn2:es)
    | t3 < t2 = n :=: (Rest (diff ticks t1 t3) :+: readTrack ticks upm (nOn2:es
    | t3 > t2 = n :+: Rest (diff ticks t2 t3) :+: readTrack ticks upm (nOn2:es
        where plainNote = Note (pitch p) (diff ticks t1 t2) (makeVol v)
            n = if ch == 9 then Instr "drums" plainNote
                else Instr (lookupUPM ch upm) plainNote
readTrack ticks upm (nOn@ (MidiEvent _ (NoteOn _ _ _))
    : nOff@(MidiEvent _ (NoteOff _ _ _))
    : x
    : es)
```

```
> = readTrack ticks upm (x : nOn : nOff : es)
> readTrack ticks upm (MidiEvent t1 (NoteOn ch p v)
    : MidiEvent t2 (NoteOff _ _ _)
    : es)
    | t2 < t1 = error "readTrack: elapsed time values of last note out of order"
    | otherwise = n :+: readTrack ticks upm es
        where plainNote = Note (pitch p) (diff ticks t1 t2) (makeVol v)
            n = if ch == 9 then Instr "drums" plainNote
                                else Instr (lookupUPM ch upm) plainNote
readTrack ticks upm ((MidiEvent t (ProgChange ch num)):es)
    = readTrack ticks (progChange ch num upm) es
    readTrack ticks upm (e:es)
> = readTrack ticks upm es
```

Take the division in ticks and two time values and calculates the note duration (quarter note, eighth note, etc.) that expresses their difference.

```
> diff :: Int -> Int -> Int -> Dur
```

$>$ diff ticks t1 t2 = ((t2 - t1) \% ticks) / 8

Look up an instrument name from a UserPatchMap given its channel number.

```
> lookupUPM :: Int -> UserPatchMap -> String
> lookupUPM ch ((name,_,midiChan):xs)
> | ch == midiChan = name
> | otherwise = lookupUPM ch xs
> lookupUPM ch [] = error ("Invalid channe1 in user patch map")
```

Implement a Program Change: a change in the UserPatchMap in which a channel changes from one instrument to another.

```
> progChange :: Int -> Int -> UserPatchMap -> UserPatchMap
> progChange ch num ((_, _, midiChan):xs) | ch == midiChan =
> let n = lookupInst genMidiMap num
> in ((n,n,ch):xs)
> progChange ch num (x:xs) = x:(progChange ch num xs)
> progChange _ _ [] = []
> progChange _ _ _ = error "progChange: Uh oh."
```

Look up an instrument in the General Midi table given its program number.

```
> lookupInst :: GenMidiTable -> Int -> IName
> lookupInst ((inst,num):xs) n | (num == n) = inst
> | otherwise = lookupInst xs n
```

Interpret a MidiFile velocity, creating a Music NoteAttribute. The MIDI specification calls for some sort of exponential scale, but for now it's just linear.

```
> makeVol :: Int -> [NoteAttribute]
> makeVol x = [Volume (float x)]
```

The newTracks function changes the order of the events in a list of sub-tracks so that they can be handled by readTrack: each NoteOff is put directly after its corresponding NoteOn. Its first and second arguments are the elapsed time and value (in microseconds per quarter note) of the SetTempo currently in effect.

```
> newTracks :: Int -> [Track] -> [Track]
> newTracks dur = newTracks' 0 dur
    where
        newTracks' :: Int -> Int -> [Track] -> [Track]
        newTracks' - _ [] = []
        newTracks' _ _ [[]] = [[]]
        newTracks' stt stv ([]:trks) = [] : newTracks' stt stv trks
        newTracks' _ _ ((e@(MetaEvent newStt (SetTempo newStv)):es):trks) =
            let (trk':trks') = newTracks' newStt newStv (es:trks)
            in ((e:trk'):trks')
        newTracks' stt stv ((e1@(MidiEvent tno (NoteOn c p _)):es):trks) =
            let (e2, leftover) = search (stt-tno) stv stv tno stt c p (es:trks) [[]]
                (trk':trks') = newTracks' stt stv leftover
            in ((e1:e2:trk'):trks')
        newTracks' stt stv ((e:es):trks) = let (trk':trks') = newTracks' stt stv (es:trks
                                    in ((e:trk'):trks')
        newTracks' stt stv (trk:trks) = trk:(newTracks' stt stv trks)
```

search takes a track that has been divided by tempo into sub-lists; it looks through the list of sub-tracks to find the NoteOff corresponding to the given NoteOn. A NoteOff corresponds to an earlier NoteOn if it is the first in the track to have the same channel and pitch. If it's in a different sub-track than its NoteOn, it puts it in the NoteOn's subtrack and calculates
what the new elapsed-time should be after one or more tempo changes. This function takes a ridiculous number of arguments, I know, but I don't think it can do without any of the information. Maybe there is a simpler way.

```
> search :: Int -> -- time interval between NoteOn and most recent SetTempo,
> -- in terms of the tempo at the NoteOn
> Int -> Int -> -- SetTempo values: the one at the NoteOn and the most
> -- recent one
> Int -> Int -> -- elapsed times of the NoteOn and the most recent SetTemf
> Int -> Int -> -- channel and pitch of NoteOn (NoteOff must match)
> [Track] -> -- the tracks left to be searched
> [Track] -> -- accumulator: what's left after NoteOff is found
> (MEvent,[Track]) -- the needed event and the remainder of the tracks
> search int ost nst tno stt c1 p1 ((e@(MidiEvent t (NoteOff c2 p2 v)):es):trks)
> | c1 == c2 && p1 == p2
> = ((MidiEvent eTime (NoteOff c2 p2 v)),
> (reverse aTrks)++(((reverse aTrk)++es):trks))
> where eTime = tno + int + round ((float (t-stt))*(float ost)/(float nst))
> search int ost nst tno stt c p ((e@(MetaEvent t (SetTempo st)):es):trks)
```

```
> search newInt ost st tno t c p (es:trks) ((e:aTrk):aTrks)
> where newInt = int + round ((float (t-stt))*(float ost)/(float nst))
> search _ _ _ tno _ c p [[]] (aTrk:aTrks) =
> ((MidiEvent tno (NoteOff c p 0)), aTrks++[reverse aTrk])
search _ _ _ tno _ c p [] (aTrk:aTrks) =
> ((MidiEvent tno (NoteOff c p 0)), aTrks++[reverse aTrk])
search int ost nst tno stt c p ((e:es):trks) (aTrk:aTrks) =
> search int ost nst tno stt c p (es:trks) ((e:aTrk):aTrks)
search int ost nst tno stt c p ([] :trks) (aTrk:aTrks) =
> search int ost nst tno stt c p trks ([]:(reverse aTrk):aTrks)
> search _ _ _ _ _ _ _ [] =
> error "newTracks: search fe11 thru"
```

Music objects that come out of readMidi almost always contain redundancies, like rests of zero duration and redundant instrument specifications. optimize reduces the redundancy to make a Music file less cluttered and more efficient to use.

```
> optimize, optRests, optTempo, optInstr, optVol :: Music -> Music
optimize m = optTempo $ optVol $ optInstr $ optRests m
```

Remove rests of zero duration.

```
> optRests (m :+: Rest 0) = optRests m
> optRests (m :=: Rest 0) = optRests m
> optRests (Rest 0 :+: m) = optRests m
> optRests (Rest 0 :=: m) = optRests m
> optRests (m1 :+: m2) = optRests m1 :+: optRests m2
> optRests (m1 :=: m2) = optRests m1 :=: optRests m2
> optRests (Tempo a m) = Tempo a $ optRests m
> optRests (Trans a m) = Trans a $ optRests m
> optRests (Instr a m) = Instr a $ optRests m
> optRests (Player a m) = Player a $ optRests m
> optRests (Phrase a m) = Phrase a $ optRests m
> optRests x = x
```

Remove redundant Tempo’s.

```
> optTempo (Tempo 1 m) = optTempo m
> optTempo (Tempo a (Tempo b m)) = Tempo (a*b) $ optTempo m
> optTempo (m1 :+: m2) = optTempo m1 :+: optTempo m2
> optTempo (m1 :=: m2) = optTempo m1 :=: optTempo m2
> optTempo (Tempo a m) = Tempo a $ optTempo m
> optTempo (Trans a m) = Trans a $ optTempo m
> optTempo (Instr a m) = Instr a $ optTempo m
> optTempo (Player a m) = Player a $ optTempo m
> optTempo (Phrase a m) = Phrase a $ optTempo m
> optTempo x = x
```

Remove unnecessary Instr's.

```
> optInstr m1 = let (m2,changed) = optInstr' m1
> in if changed then optInstr m2
                        e1se m2
    where
        optInstr' :: Music -> (Music, Bool)
        optInstr' ((m1 :+: m2) :+: m3) = (m1 :+: m2 :+: m3, True)
    optInstr' ((m1 :=: m2) :=: m3) = (m1 :=: m2 :=: m3, True)
    optInstr' (Instr a (Instr b m)) = (Instr a m, True)
    optInstr' (Instr a m1 :+: Instr b m2) | a == b = (Instr a (m1 :+: m2), True)
    optInstr' (Instr a m1 :=: Instr b m2) | a == b = (Instr a (m1 :=: m2), True)
    optInstr' (Instr a m1 :+: Instr b m2 :+: x)
        | a == b = (Instr a (m1 :+: m2) :+: x, True)
```

```
> optInstr' (Instr a m1 :=: Instr b m2 :=: x)
    | a == b = (Instr a (m1 :=: m2) :=: x, True)
    optInstr' (Instr a m :+: Rest r) = (Instr a (m :+: Rest r), True)
    optInstr' (Instr a m :=: Rest r) = (Instr a (m :=: Rest r), True)
    optInstr' (Rest r :+: Instr a m) = (Instr a (Rest r :+: m), True)
    optInstr' (Rest r :=: Instr a m) = (Instr a (Rest r :=: m), True)
    optInstr' (m1 :+: m2) = let (m3,c3) = optInstr' m1
    (m4,c4) = optInstr' m2
    in (m3 :+: m4, c3 || c4)
optInstr' (m1 :=: m2) = 1et (m3,c3) = optInstr' m1
    (m4,c4) = optInstr' m2
    in (m3 :=: m4, c3 || c4)
    optInstr' (Tempo a m) = (Tempo a $ optInstr m, False)
    optInstr' (Trans a m) = (Trans a $ optInstr m, False)
    optInstr' (Instr a m) = (Instr a $ optInstr m, False)
    optInstr' (Player a m) = (Player a $ optInstr m, False)
    optInstr' (Phrase a m) = (Phrase a $ optInstr m, False)
    optInstr' x = (x, False)
```

Change repeated Volume Note Attributes to Phrase Attributes.

```
optVo1 m1 = let (m2,changed) = optVol' m1
        in if changed then optVol m2
            else m2
optVol' ((m1 :+: m2) :+: m3) = (m1 :+: m2 :+: m3, True)
optVol' ((m1 :=: m2) :=: m3) = (m1 :=: m2 :=: m3, True)
optVol' (Phrase p1 (Phrase p2 m)) | p1 == p2 = (Phrase p1 m, True)
                            otherwise = (Phrase (p1 ++ p2) m, True)
optVol' (Note p1 d1 [Volume v1] :+: Note p2 d2 [Volume v2]) | v1 == v2 =
    (Phrase [Dyn (Loudness v1)] (Note p1 d1 [] :+: Note p2 d2 []), True)
optVol' (Note p1 d1 [Volume v1] :=: Note p2 d2 [Volume v2]) | v1 == v2 =
    (Phrase [Dyn (Loudness v1)] (Note p1 d1 [] :=: Note p2 d2 []), True)
optVol' (Note p1 d1 [Volume v1] :+: Note p2 d2 [Volume v2] :+: x) | v1 == v2 =
    ((Phrase [Dyn (Loudness v1)] (Note p1 d1 [] :+: Note p2 d2 [])) :+: x, True)
optVol' (Note p1 d1 [Volume v1] :=: Note p2 d2 [Volume v2] :=: x) | v1 == v2 =
    ((Phrase [Dyn (Loudness v1)] (Note p1 d1 [] :=: Note p2 d2 [])) :=: x, True)
optVol' (Phrase p1 m1 :+: Phrase p2 m2) | p1 == p2 = (Phrase p1 (m1 :+: m2), True)
optVol' (Phrase p1 m1 :=: Phrase p2 m2) | p1 == p2 = (Phrase p1 (m1 :=: m2), True)
optVol' (Phrase p1 m1 :+: Phrase p2 m2 :+: x) | p1 == p2 =
    (Phrase p1 (m1 :+: m2 :+: x),True)
optVol' (Phrase p1 m1 :=: Phrase p2 m2 :=: x) | p1 == p2 =
    (Phrase p1 (m1 :=: m2 :=: x),True)
optVol' (Phrase phr@[Dyn (Loudness 1)] m :+: Note p d [Volume v]) | 1 == v =
```

```
> (Phrase phr (m :+: Note p d []), True)
optVol' (Phrase phr@[Dyn (Loudness 1)] m :=: Note p d [Volume v]) | 1 == v =
(Phrase phr (m :=: Note p d []), True)
optVol' (Phrase phr@[Dyn (Loudness 1)] m :+: Note p d [Volume v] :+: x) | 1 == v =
(Phrase phr (m :+: Note p d []) :+: x,True)
optVol' (Phrase phr@[Dyn (Loudness 1)] m :=: Note p d [Volume v] :=: x) | 1 == v =
(Phrase phr (m :=: Note p d []) :=: x,True)
optVol' (Note p d [Volume v] :+: Phrase phr@[Dyn (Loudness 1)] m) | 1 == v =
(Phrase phr (Note p d [] :+: m),True)
optVol' (Note p d [Volume v] :=: Phrase phr@[Dyn (Loudness 1)] m) | 1 == v =
    (Phrase phr (Note p d [] :=: m),True)
optVol' (Note p d [Volume v] :+: Phrase phr@[Dyn (Loudness 1)] m :+: x) | 1 == v =
    ((Phrase phr (Note p d [] :+: m)) :+: x, True)
optVol' (Note p d [Volume v] :=: Phrase phr@[Dyn (Loudness 1)] m :=: x) | 1 == v =
    ((Phrase phr (Note p d [] :=: m)) :=: x, True)
optVol' (Rest r :+: Note p d [Volume v]) =
    (Phrase [Dyn (Loudness v)] (Rest r :+: Note p d []),True)
optVol' (Rest r :=: Note p d [Volume v]) =
    (Phrase [Dyn (Loudness v)] (Rest r :=: Note p d []),True)
optVol' (Note p d [Volume v] :+: Rest r) =
    (Phrase [Dyn (Loudness v)] (Note p d [] :+: Rest r),True)
optVol' (Note p d [Volume v] :=: Rest r) =
    (Phrase [Dyn (Loudness v)] (Note p d [] :=: Rest r),True)
    optVol' (Rest r :+: Phrase p m) = (Phrase p (Rest r :+: m), True)
optVol' (Rest r :=: Phrase p m) = (Phrase p (Rest r :=: m), True)
optVol' (Phrase p m :+: Rest r) = (Phrase p (m :+: Rest r), True)
> optVol' (Phrase p m :=: Rest r) = (Phrase p (m :=: Rest r), True)
> optVol' (m1 :+: m2) = let (m3,c3) = optVol' m1
                                    (m4,c4) = optVol' m2
    in (m3 :+: m4, c3 || c4)
> optVol' (m1 :=: m2) = let (m3,c3) = optVol' m1
    (m4,c4) = optVol' m2
    in (m3 :=: m4, c3 || c4)
> optVol' (Tempo a m) = (Tempo a $ optVol m, False)
> optVol' (Trans a m) = (Trans a $ optVol m, False)
> optVol' (Instr a m) = (Instr a $ optVol m, False)
> optVol' (Player a m) = (Player a $ optVol m, False)
> optVol' (Phrase a m) = (Phrase a $ optVol m, False)
> optVol'x = (x, False)
```


## 10 Representing Chords

Earlier I described how to represent chords as values of type Music. However, sometimes it is convenient to treat chords more abstractly. Rather than think of a chord in terms of its actual notes, it is useful to think of it in terms of its chord "quality," coupled with the key it is played in and the particular voicing used. For example, we can describe a chord as being a "major triad in root position, with root middle C." Several approaches have been put forth for representing this information, and we cannot cover all of them here. Rather, I will describe two basic representations, leaving other alternatives to the skill and imagination of the reader. ${ }^{3}$

First, one could use a pitch representation, where each note is represented as its distance from some fixed pitch. 0 is the obvious fixed pitch to use, and thus, for example, $[0,4,7]$ represents a major triad in root position. The first zero is in some sense redundant, of course, but it serves to remind us that the chord is in "normal form." For example, when forming and transforming chords, we may end up with a representation such as $[2,6,9]$, which is not normalized; its normal form is in fact $[0,4,7]$. Thus we define:

A chord is in pitch normal form if the first pitch is zero, and the subsequent pitches are monotonically increasing.

One could also represent a chord intervalically; i.e. as a sequence of intervals. A major triad in root position, for example, would be represented as $[4,3,-7]$, where the last interval "returns" us to the "origin." Like the 0 in the pitch representation, the last interval is redundant, but allows us to define another sense of normal form:

A chord is in interval normal form if the intervals are all greater than zero, except for the last which must be equal to the negation of the sum of the others.

In either case, we can define a chord type as:

```
> type Chord = [AbsPitch]
```

We might ask whether there is some advantage, computationally, of using one of these representations over the other. However, there is an invertible linear transformation between them, as defined by the following functions, and thus there is in fact little advantage of one over the other:

```
> pitToInt :: Chord -> Chord
> pitToInt ch = aux ch
> where aux (n1:n2:ns) = (n2-n1) : aux (n2:ns)
```

[^2]```
> aux [n] = [head ch - n]
>
> intToPit :: Chord -> Chord
> intToPit ch = 0 : aux 0 ch
> where aux p [n] = []
> aux p (n:ns) = n' : aux n' ns where n' = p+n
```

Exercise 9 Show that pitToInt and intToPit are inverses in the following sense: for any chord ch1 in pitch normal form, and ch2 in interval normal form, each of length at least two:

```
intToPit (pitToInt ch1) = ch1
pitToInt (intToPit ch2) = ch2
```

Another operation we may wish to perform is a test for equality on chords, which can be done at many levels: based only on chord quality, taking inversion into account, absolute equality, etc. Since the above normal forms guarantee a unique representation, equality of chords with respect to chord quality and inversion is simple: it is just the standard (overloaded) equality operator on lists. On the other hand, to measure equality based on chord quality alone, we need to account for the notion of an inversion.

Using the pitch representation, the inversion of a chord can be defined as follows:

```
> pitInvert (p1:p2:ps) = 0 : map (subtract p2) ps ++ [12-p2]
```

Although we could also directly define a function to invert a chord given in interval representation, we will simply define it in terms of functions already defined:

```
> intInvert = pitToInt . pitInvert . intToPit
```

We can now determine whether a chord in normal form has the same quality (but possibly different inversion) as another chord in normal form, as follows: simply test whether one chord is equal either to the other chord or to one of its inversions. Since there is only a finite number of inversions, this is well defined. In Haskell:

```
> samePitChord ch1 ch2 =
> let invs = take (length ch1) (iterate pitInvert ch1)
> in ch2 'elem` invs
>
> sameIntChord ch1 ch2 =
> let invs = take (length ch1) (iterate intInvert ch1)
> in ch2 'elem` invs
```

For example, samePitChord [0,4, 7 ] [0,5,9] returns True (since $[0,5,9]$ is the pitch normal form for the second inversion of $[0,4,7]$ ).

## 11 Equivalence of Literal Performances

A literal performance is one in which no aesthetic interpretation is given to a musical object. The function perform in fact yields a literal performance; aesthetic nuances must be expressed explicitly using note and phrase attributes.

There are many musical objects whose literal performances we expect to be equivalent. For example, the following two musical objects are certainly not equal as data structures, but we would expect their literal performances to be identical:

```
(m1 :+: m2) :+: (m3 :+: m4)
    m1 :+: m2 :+: m3 :+: m4
```

Thus we define a notion of equivalence:

Definition: Two musical objects $m 1$ and $m 2$ are equivalent, written $m 1 \equiv m 2$, if and only if:

```
(\forallimap,c) perform imap c m1 = perform imap c m2
```

where "=" is equality on values (which in Haskell is defined by the underlying equational logic).
One of the most useful things we can do with this notion of equivalence is establish the validity of certain transformations on musical objects. A transformation is valid if the result of the transformation is equivalent (in the sense defined above) to the original musical object; i.e. it is "meaning preserving."

The most basic of these transformation we treat as axioms in an algebra of music. For example:

Axiom 1 For any r1, r2, r3, r4, and m:

$$
\text { Tempo r1 r2 (Tempo r3 r4 m) } \equiv \text { Tempo (r1*r3) (r2*r4) m }
$$

To prove this axiom, we use conventional equational reasoning (for clarity we omit imap and simplify the context to just dt ):

## Proof:

```
perform dt (Tempo r1 r2 (Tempo r3 r4 m))
= perform (r2*dt/r1) (Tempo r3 r4 m) -- unfolding perform
= perform (r4*(r2*dt/r1)/r3) m -- unfolding perform
= perform ((r2*r4)*dt/(r1*r3)) m -- simple arithmetic
= perform dt (Tempo (r1*r3) (r2*r4) m) -- folding perform
```

Here is another useful transformation and its validity proof (for clarity in the proof we omit imap and simplify the context to just ( $\mathrm{t}, \mathrm{dt}$ ) ):

Axiom 2 For any $\mathrm{r} 1, \mathrm{r} 2, \mathrm{~m} 1$, and m 2 :

```
Tempo r1 r2 (m1 :+: m2) \equiv Tempo r1 r2 m1 :+: Tempo r1 r2 m2
```

In other words, tempo scaling distributes over sequential composition.

## Proof:

```
perform (t,dt) (Tempo r1 r2 (m1 :+: m2))
= perform (t,r2*dt/r1) (m1 :+: m2) -- unfolding perform
= perform (t,r2*dt/r1) m1 ++ perform (t',r2*dt/r1) m2 -- unfolding perform
= perform (t,dt) (Tempo r1 r2 m1) ++
    perform (t',dt) (Tempo r1 r2 m2) -- folding perform
= perform (t,dt) (Tempo r1 r2 m1) ++
    perform (t'',dt) (Tempo r1 r2 m2) -- folding dur
= perform (t,dt) (Tempo r1 r2 m1 :+: Tempo r1 r2 m2) -- folding perform
where t' = t + (dur m1)*r2*dt/r1
    t'' = t + (dur (Tempo r1 r2 m1))*dt
```

An even simpler axiom is given by:

Axiom 3 For any r and m :

$$
\text { Tempo r r m } \equiv \mathrm{m}
$$

In other words, unit tempo scaling is the identity.


Figure 9: Equivalent Phrases

## Proof:

```
perform (t,dt) (Tempo r r m)
= perform (t,r*dt/r) m -- unfolding perform
= perform (t,dt) m -- simple arithmetic
```

Note that the above proofs, being used to establish axioms, all involve the definition of perform. In contrast, we can also establish theorems whose proofs involve only the axioms. For example, Axioms 1, 2, and 3 are all needed to prove the following:

Theorem 1 For any r1, r2, m1, and m2:

```
Tempo r1 r2 m1 :+: m2 \equiv Tempo r1 r2 (m1 :+: Tempo r2 r1 m2)
```


## Proof:

```
Tempo r1 r2 (m1 :+: Tempo r2 r1 m2)
= Tempo r1 r2 m1 :+: Tempo r1 r2 (Tempo r2 r1 m2) -- by Axiom 1
= Tempo r1 r2 m1 :+: Tempo (r1*r2) (r2*r1) m2 -- by Axiom 2
= Tempo r1 r2 m1 :+: Tempo (r1*r2) (r1*r2) m2 -- simp1e arithmetic
= Tempo r1 r2 m1 :+: m2 -- by Axiom 3
```

For example, this fact justifies the equivalence of the two phrases shown in Figure 9.
Many other interesting transformations of Haskore musical objects can be stated and proved correct using equational reasoning. We leave as an exercise for the reader the proof of the following axioms (which include the above axioms as special cases).

Axiom 4 Tempo is multiplicative and Transpose is additive. That is, for any r1, r2, r3, r4, p, and m :

$$
\begin{gathered}
\text { Tempo r1 r2 (Tempo r3 r4m) } \equiv \text { Tempo (r1*r3) (r2*r4) m } \\
\text { Trans p1 (Trans p2 m) } \equiv \text { Trans (p1+p2) m }
\end{gathered}
$$

Axiom 5 Function composition is commutative with respect to both tempo scaling and transposition. That is, for any r1, r2, r3, r4, p1 and p2:

```
Tempo r1 r2 . Tempo r3 r4 \equiv Tempo r3 r4 . Tempo r1 r2
            Trans p1. Trans p2 \equiv Trans p2 . Trans p1
    Tempo r1 r2 . Trans p1 \equiv Trans p1 . Tempo r1 r2
```

Axiom 6 Tempo scaling and transposition are distributive over both sequential and parallel composition. That is, for any r1, r2, p, m1, and m2:

```
Tempo r1 r2 (m1 :+: m2) \equiv Tempo r1 r2 m1 :+: Tempo r1 r2 m2
Tempo r1 r2 (m1 :=: m2) \equiv Tempo r1 r2 m1 :=: Tempo r1 r2 m2
    Trans p (m1 :+: m2) \equiv Trans p m1 :+: Trans p m2
    Trans p (m1 :=: m2) \equiv Trans p m1 :=: Trans p m2
```

Axiom 7 Sequential and parallel composition are associative. That is, for any m0, m1, and m 2 :

$$
\begin{aligned}
& m 0:+:(m 1:+: m 2) \equiv(m 0:+: m 1):+: m 2 \\
& m 0:=:(m 1:=: m 2) \equiv(m 0:=: m 1):=: m 2
\end{aligned}
$$

Axiom 8 Parallel composition is commutative. That is, for any m0 and m1:
m0 :=: m1 三 m1 :=: m0

Axiom 9 Rest 0 is a unit for Tempo and Trans, and a zero for sequential and parallel composition. That is, for any $\mathrm{r} 1, \mathrm{r} 2, \mathrm{p}$, and m :

$$
\begin{aligned}
& \text { Tempo r1 r2 (Rest } 0) \equiv \text { Rest } 0 \\
& \text { Trans p (Rest } 0) \equiv \text { Rest } 0 \\
& m:+: R e s t 0 \equiv m \equiv \operatorname{Rest} 0:+: m \\
& m:=: \text { Rest } 0 \equiv m \equiv \text { Rest } 0:=: m
\end{aligned}
$$

Exercise 10 Establish the validity of each of the above axioms.

## 12 Haskore Support for CSound

```
> module CSound where
> import Performance
> import IO
> import List (find, nub)
```

[Note: if this module is loaded into Hugs98, the following error message results:

```
Reading file "CSound.1hs":
ERROR "CSound.lhs" (line 707):
*** Cannot derive Eq OrcExp after 40 iterations.
*** This may indicate that the problem is undecidable. However,
*** you may still try to increase the cutoff limit using the -c
*** option and then try again. (The current setting is -c40)
```

This is apparently due to the size of the OrcExp data type. For correct operation, start Hugs with a larger cutoff limit, such as -c1000.]

CSound is a software synthesizer that allows its user to create a virtually unlimited number of sounds and instruments. It is extremely portable because it is written entirely in C. Its strength lies mainly in the fact that all computations are performed in software, so it is not reliant on sophisticated musical hardware. The output of a CSound computation is a file representing the signal which can be played by an independent application, so there is no hard upper limit on computation time. This is important because many sophisticated signals take much longer to compute than to play. The purpose of this module is to create an interface between Haskore and CSound in order to give the Haskore user access to all the powerful features of a software sound synthesizer.

CSound takes as input two plain text files: a score (.sco) file and an orchestra (.orc) file. The score file is similar to a Midi file, and the orchestra file defines one or more instruments that are referenced from the score file (the orchestra file can thus be thought of as the software equivalent of Midi hardware). The CSound program takes these two files as input, and produces a sound file as output, usually in .wav format. Sound files are generally much larger than Midi files, since they describe the actual sound to be generated, represented as a sequence of values (typically 44,100 of them for each second of music), which are converted directly into voltages that drive the audio speakers. Sound files can be played by any standard media player found on conventional PC's.

Each of these files is described in detail in the following sections.

### 12.1 The Score File

We will represent a score file as a sequence of score statements:

```
> type Score = [ScoreStmt]
```

The ScoreStmt data type is designed to simulate CSound's three kinds of score statements:

1. A tempo statement, which sets the tempo. In the absence of a tempo statement, the tempo defaults to 60 beats per minute.
2. A note event, which defines the start time, pitch, duration (in beats), volume (in decibels), and instrument to play a note (and is thus more like a Haskore Event than a Midi event, thus making the conversion to CSound easier than to Midi, as we shall see later). Each note event also contains a number of optional arguments called $p$-fields, which determine other properties of the note, and whose interpretation depends on the instrument that plays the note. This will be discussed further in a later section.
3. Function table definitions. A function table is used by instruments to produce audio signals. For example, sequencing through a table containing a perfect sine wave will produce a very pure tone, while a table containing an elaborate polynomial will produce a complex sound with many overtones. The tables can also be used to produce control signals that modify other signals. Perhaps the simplest example of this is a tremolo or vibrato effect, but more complex sound effects, and FM (frequency modulation) synthesis in general, is possible.
```
> data ScoreStmt = CSTempo Bpm
> | CSNote Inst StartTime Duration Pch V1m [Pfield]
> | CSTable Tbl CreatTime TblSize Normalize GenRoutine
> deriving Show
>
> type Bpm = Int
> type Inst = Int
> type StartTime = Float
> type Duration = Float
> type Pch = Float
> type V1m = Float
> type Pfield = Float
> type Tb1 = Int
> type CreatTime = Float
> type TblSize = Int
> type Normalize = Bool
```

This is all rather straightforward, except for function table generation, which requires further explanation.

### 12.1.1 Function Tables

Each function table must have a unique integer ID (Tb1), creation time (usually 0), size (which must be a power of 2), and a Normalize flag. Most tables in CSound are normalized, i.e. rescaled to a maximum absolute value of 1 . The normalization process can be skipped by setting the Normalize flag to False. Such a table may be desirable to generate a control or modifying signal, but is not very useful for audio signal generation.

Tables are simply arrays of floating point values. The values stored in the table are calculated by one of CSound's predefined generating routines, represented by the type GenRoutine:

```
> data GenRoutine = GenRoutine GenNum [GenArg]
> | SoundFile SFName SkipTime ChanNum
> deriving Show
> type SFName = String
> type SkipTime = Float
> type ChanNum = Float
> type GenNum = Int
> type GenArg = Float
```

GenRoutine n args refers to CSound's generating routine $n$ (an integer), called with floating point arguments args. There is only one generating routine (called GEN01) in CSound that takes an argument type other than floating point, and thus we represent this using the special constructor SoundFile, whose functionality will be described shortly.

Knowing which of CSound's generating routines to use and with what arguments can be a daunting task. The newest version of CSound (version 4.01) provides 23 different generating routines, and each one of them assigns special meanings to its arguments. To avoid having to reference routines using integer ids, the following functions are defined for the most often-used generating routines. A brief discussion of each routine is also included. For a full description of these and other routines, refer to the CSound manual or consult the following webpage: http://www.1eeds.ac.uk/music/Man/Csound/Function/GENS.htm1. The user familiar with CSound is free to write helper functions like the ones below to capture other generating routines.

GEN01. Transfers data from a soundfile into a function table. Recall that the size of the function table in CSound must be a power of two. If the soundfile is larger than the table size, reading stops when the table is full; if it is smaller, then the table is padded with zeros. One exception is allowed: if the file is of type AIFF and the table size is set to zero, the size of the function table is allocated dynamically as the number of points in the soundfile. The table is then unusable by normal oscillators, but can be used by a special Samp0sc constructor (discussed in Section 12.2). The first argument passed to the GEN01 subroutine is a string
containing the name of the source file. The second argument is skip time, which is the number of seconds into the file that the reading begins. Finally there is an argument for the channel number, with 0 meaning read all channels. GEN01 is represented in Haskore as SoundFile SFName SkipTime ChanNum, as discussed earlier. To make the use of SoundFile consistent with the use of other functions to be described shortly, we define a simple equivalent:

```
> soundFile :: SFName -> SkipTime -> ChanNum -> GenRoutine
> soundFile = SoundFile
```

GEN02. Transfers data from its argument fields directly into the function table. We represent its functionality as follows:

```
> tableValues :: [GenArg] -> GenRoutine
> tableValues gas = GenRoutine 2 gas
```

GEN03. Fills the table by evaluating a polynomial over a specified interval and with given coefficients. For example, calling GEN03 with an interval of $(-1,1)$ and coefficients $5,4,3$, $2,0,1$ will generate values of the function $5+4 x+3 x^{2}+2 x^{3}+x^{5}$ over the interval -1 to 1. The number of values generated is equal to the size of the table. Let's express this by the following function:

```
> polynomia1 :: Interva1 -> Coefficients -> GenRoutine
> polynomial (x1,x2) cfs = GenRoutine 3 (x1:x2:cfs)
>
> type Interval = (Float, Float)
> type Coefficients = [Float]
```

GEN05. Constructs a table from segments of exponential curves. The first argument is the starting point. The meaning of the subsequent arguments alternates between the length of a segment in samples, and the endpoint of the segment. The endpoint of one segment is the starting point of the next. The sum of all the segment lengths normally equals the size of the table: if it is less the table is padded with zeros, if it is more, only the first Tb1Size locations will be stored in the table.

```
> exponentia11 :: StartPt -> [(SegLength, EndPt)] -> GenRoutine
```

```
> exponential1 sp xs = GenRoutine 5 (sp : flattenTuples2 xs)
>
> type StartPt = Float
> type SegLength = Float
> type EndPt = Float
```

f1attenTup1es2 flattens a list of pairs into a list. Similarly, f1attenTup1es3 flattens a list of 3 -tuples into a list, and so on.

```
> flattenTuples2 :: [(a,a)] -> [a]
> flattenTuples3 :: [(a,a,a)] -> [a]
> flattenTup1es4 :: [(a,a,a,a)] -> [a]
>
> f1attenTup1es2 [] = []
> flattenTuples2 ((x,y) : xs) = x : y : flattenTuples2 xs
>
> flattenTuples3 [] = []
> flattenTuples3 ((x,y,z) : xs) = x : y : z : flattenTuples3 xs
>
> flattenTuples4 [] = []
> flattenTuples4 ((x, y, z, w) : xs) = x : y : z : w : flattenTuples4 xs
```

GEN25. Similar to GEN05 in that it produces segments of exponential curves, but instead of representing the lengths of segments and their endpoints, its arguments represent ( $x, y$ ) coordinates in the table, and the subroutine produces curves between successive locations. The $x$-coordinates must be in increasing order.

```
> exponentia12 :: [Point] -> GenRoutine
> exponential2 pts = GenRoutine 25 (flattenTuples2 pts)
>
> type Point = (Float,Float)
```

GEN06. Generates a table from segments of cubic polynomial functions, spanning three points at a time. We define a function cubic with two arguments: a starting position and a list of segment length (in number of samples) and segment endpoint pairs. The endpoint of one segment is the starting point of the next. The meaning of the segment endpoint alternates between a local minimum/maximum and point of inflexion. Whether a point is a maximum or a minimum is determined by its relation to the next point of inflexion. Also note that for
two successive minima or maxima, the inflexion points will be jagged, whereas for alternating maxima and minima, they will be smooth. The slope of the two segments is independent at the point of inflection and will likely vary. The starting point is a local minimum or maximum (if the following point is greater than the starting point, then the starting point is a minimum, otherwise it is a maximum). The first pair of numbers will in essence indicate the position of the first inflexion point in ( $x, y$ ) coordinates. The folowing pair will determine the next local minimum/maximum, followed by the second point of inflexion, etc.

```
> cubic :: StartPt -> [(SegLength, EndPt)] -> GenRoutine
> cubic sp pts = GenRoutine 6 (sp : flattenTuples2 pts)
```

GEN07. Similar to GEN05, except that it generates straight lines instead of exponential curve segments. All other issues discussed about GEN05 also apply to GEN07. We represent it as:

```
> lineSeg1 :: StartPt -> [(SegLength, EndPt)] -> GenRoutine
> lineSeg1 sp pts = GenRoutine 7 (sp : flattenTuples2 pts)
```

GEN27. As with GEN05 and GEN25, produces straight line segments between points whose locations are given as $(x, y)$ coordinates, rather than a list of segment length, endpoint pairs.

```
> lineSeg2 :: [Point] -> GenRoutine
> 1ineSeg2 pts = GenRoutine 27 (flattenTup1es2 pts)
```

GEN08. Produces a smooth piecewise cubic spline curve through the specified points. Neighboring segments have the same slope at the common points, and it is that of a parabola through that point and its two neighbors. The slope is zero at the ends.

```
> cubicSpline :: StartPt -> [(SegLength, EndPt)] -> GenRoutine
> cubicSpline sp pts = GenRoutine 8 (sp : flattenTuples2 pts)
```

GEN10. Produces a composite sinusoid. It takes a list of relative strengths of harmonic partials $1,2,3$, etc. Partials not required should be given strength of zero.

```
> compSine1 :: [PStrength] -> GenRoutine
> compSine1 pss = GenRoutine 10 pss
>
> type PStrength = Float
```

GEN09. Also produces a composite sinusoid, but requires three arguments to specify each contributing partial. The arguments specify the partial number, which doesn't have to be an integer (i.e. inharmonic partials are allowed), the relative partial strength, and the initial phase offset of each partial, expressed in degrees.

```
> compSine2 :: [(PNum, PStrength, PhaseOffset)] -> GenRoutine
> compSine2 args = GenRoutine 9 (flattenTup1es3 args)
>
> type PNum = Float
> type PhaseOffset = Float
```

GEN19. Provides all of the functionality of GEN09, but in addition a DC offset must be specified for each partial. The DC offset is a vertical displacement, so that a value of 2 will lift a 2 -strength partial from range $[-2,2]$ to range $[0,4]$ before further scaling.

```
> compSine3 :: [(PNum, PStrength, PhaseOffset, DCOffset)] -> GenRoutine
> compSine3 args = GenRoutine 19 (flattenTuples4 args)
>
> type DCOffset = Float
```

GEN11. Produces an additive set of harmonic cosine partials, similar to GEN10. We will represent it by a function that takes three arguments: the number of harmonics present, the lowest harmonic present, and a multiplier in an exponential series of harmonics amplitudes (if the $x$ 'th harmonic has strength coefficient of $A$, then the $(x+n)$ 'th harmonic will have a strength of $A *\left(r^{n}\right)$, where $r$ is the multiplier).

```
> cosineHarms :: NHarms -> LowestHarm -> Mult -> GenRoutine
> cosineHarms n 1 m = GenRoutine 11 [float n, float 1, m]
>
> type NHarms = Int
> type LowestHarm = Int
> type Mult = Float
```

GEN21. Produces tables having selected random distributions.

```
> randomTable :: RandDist -> GenRoutine
> randomTable rd = GenRoutine 21 [float rd]
>
> type RandDist = Int
>
> uniform, linear, triangular, expon,
> biexpon, gaussian, cauchy, posCauchy :: Int
> uniform = 1
> linear = 2
> triangular = 3
> expon = 4
> biexpon = 5
> gaussian = 6
> cauchy = 7
> posCauchy = 8
```

Common Tables For convenience, here are some common function tables, which take as argument the identifier integer:

```
> simpleSine, square, sawtooth, triangle, whiteNoise :: Tbl -> ScoreStmt
>
> simpleSine n = CSTable n 0 8192 True
> (compSine1 [1])
> square n = CSTable n 0 1024 True
> (lineSeg1 1 [(128, 1), (0, -1), (128, -1)])
> sawtooth n = CSTable n 0 1024 True
> (lineSeg1 0 [(128, 1), (0, -1), (128, 0)])
> triangle n = CSTable n 0 1024 True
> (lineSeg1 0 [(64, 1), (128, -1), (64, 0)])
> whiteNoise n = CSTable n O 1024 True
> (randomTable uniform)
```

The following function for a composite sine has an extra argument, a list of harmonic partial strengths:

```
> compSine :: Tb1 -> [PStrength] -> ScoreStmt
> compSine n s = CSTable 6 0 8192 True (compSine1 s)
```


### 12.1.2 Naming Instruments and Tables

In CSound, each table and instrument has a unique identifying integer associated with it. Haskore, on the other hand, uses strings to name instruments. What we need is a way to convert Haskore instrument names to identifier integers that CSound can use. Similar to Haskore's player maps, we define a notion of a CSound name map for this purpose.

```
> type NameMap = [(Name, Int)]
> type Name = String
```

A name map can be provided directly in the form [("name1", int1), ("name2", int2), ...], or the programmer can define auxiliary functions to make map construction easier. For example:

```
> makeNameMap :: [Name] -> NameMap
> makeNameMap nms = zip nms [1..]
```

The following function will add a name to an existing name map. If the name is already in the map, an error results.

```
> addToMap :: NameMap -> Name -> Int -> NameMap
> addToMap nmap nm i =
> case (getId nmap nm) of
> Nothing -> (nm,i) : nmap
> Just _ -> error (" addToMap: the name " ++ nm ++
> " is already in the name map")
>
> getId :: NameMap -> Name -> Maybe Int
> getId nmap nm =
> case (find (\(n,_) -> n==nm) nmap) of
> Nothing -> Nothing
> Just (n,i) -> Just i
```

Note the use of the function find imported from the List library.

### 12.1.3 Converting Haskore Music to a CSound Score File

To convert a Music value into a CSound score file, we need to:

## 1. Convert the Music value to a Performance.

2. Convert the Performance value to a Score.
3. Write the Score value to a CSound score file.

We already know how to do the first step. Steps two and three will be achieved by the following two functions:

```
perfToScore :: NameMap -> Performance -> Score
writeScore :: Score -> IO ()
```

The three steps can be put together in whatever way the user wishes, but the most general way would be this:

```
> musicToCSound :: NameMap -> PMap -> Tables -> Context -> Music -> IO ()
> musicToCSound nmap pmap tables cont m =
> writeScore (tables ++ perfToScore nmap (perform pmap cont m))
>
> type Tables = Score
```

The Tables argument is a user-defined set of function tables, represented as a sequence of ScoreStmts (specifically, CSTable constructors). (See Section ??.)

From Performance to Score The translation between performance Events and score CSNotes is straightforward, the only tricky parts being:

- The unit of time in a Performance is the second, whereas in a Score it is the beat. However, the default CSound tempo is 60 beats per minute, or one beat per second, as was already mentioned, and we use this default for our Score files. Thus the two are equivalent, and no translation is necessary.
- In a Performance, pitch is represented as an AbsPitch value, an integer denoting the absolute position of the pitch in semitones (MIDI uses the same representation). CSound, however, uses different pitch representations, one of them being octave point pitch class, or "pch." Using pch, a pitch is represented by a decimal number whose integer part represents the octave, and decimal part represents the semitone within the octave (thus 0.00 is the lowest C, 0.11 the lowest B, 8.00 is middle C, etc.). The function absToPch defined below performs the required conversion for us.

```
perfToScore :: NameMap -> Performance -> Score
perfToScore _ [] = []
perfToScore nmap (Event t i p d v pfs : evs) =
    case (getId nmap i) of
        Nothing -> error ("perfToScore: instrument " ++ i ++ " is unknown")
        Just num -> CSNote num t d (absToPch p) v pfs : perfToScore nmap evs
>
> absToPch :: AbsPitch -> Pch
> absToPch ap = float (ap `quot` 12) + (f1oat (ap `mod` 12) / 100.0)
```

From Score to Score File Now that we have a value of type Score, we must write it into a plain text ASCII file with an extension . sco in a way that CSound will recognize. This is done by the following function:

```
> writeScore :: Score -> IO ()
> writeScore s = do putStr "\nName your score file "
> putStr "(.sco extension will be added): "
> name <- getLine
> h <- openFile (name ++ ".sco") WriteMode
> printScore h s
> hClose h
```

This function asks the user for the name of the score file, opens that file for writing, writes the score into the file using the function printScore, and then closes the file.

The score file is a plain text file containing one statement per line. Each statement consists of an opcode, which is a single letter that determines the action to be taken, and a number of arguments. The opcodes we will use are "e" for end of score, " t " to set tempo, " f " to create a function table, and "i" for note events.

```
> printScore :: Handle -> Score -> IO ()
> printScore h [] = hPutStr h "e\n" -- end of score
> printScore h (s : ss) = do printStatement h s
> printScore h ss
```

In the following we will come across several instances where we will need to write a list of floating point numbers into the file, one number at a time, separated by spaces. To do this, we will need to convert the list to a string. This is done by the following function:

```
> listToString :: [Float] -> String
> listToString [] = ""
> 1istToString (n : ns) = " " ++ show n ++ listToString ns
```

Finally, the printStatement function:

```
> printStatement :: Handle -> ScoreStmt -> IO ()
> printStatement h (CSTempo t) =
> hPutStr h ("t 0 " ++ show t ++ "\n")
> printStatement h (CSNote i st d p v pfs) =
> hPutStr h
> ("i " ++ show i ++ " " ++ show st ++ " " ++ show d ++ " " ++
> show p ++ " " ++ show v ++ listToString pfs ++ "\n")
printStatement h (CSTable t ct s n gr) =
    hPutStr h
        ("f " ++ show t ++ " " ++ show ct ++ " " ++ show s ++
            (if n then " " else " -") ++ showGenRoutine gr ++ "\n")
    where showGenRoutine (SoundFile nm st cn) =
                                    "1 " ++ nm ++ " " ++ show st ++ " 0 " ++ show cn
                showGenRoutine (GenRoutine gn gas) =
                    show gn ++ listToString gas
```


### 12.2 The Orchestra File

The orchestra file consists of two parts: a header, and one or more instrument blocks. The header sets global parameters controlling sampling rate, control rate, and number of output channels. The instrument blocks define instruments, each identified by a unique integer ID, and containing statements modifying or generating various audio signals. Each note statement in a score file passes all its arguments-including the p-fields-to its corresponding instrument in the orchestra file. While some properties vary from note to note, and should therefore be designed as p-fields, many can be defined within the instrument; the choice is up to the user.

The orchestra file is represented as:

```
> type Orchestra = (Header, [InstBlock])
```

The orchestra header sets the audio rate, control rate, and number of output channels:

```
> type Header = (AudRate, Ctr1Rate, Chn1s)
>
> type AudRate = Int -- samples per second
> type CtrlRate = Int -- samples per second
> type Chn1s = Int -- mono=1, stereo=2, surround=4
```

Digital computers represent continuous analog audio waveforms as a sequence of discrete samples. The audio rate (AudRate) is the number of these samples calculated each second. Theoretically, the maximum frequency that can be represented is equal to one-half the audio rate. Audio CDs contain 44,100 samples per second of music, giving them a maximum sound frequency of $22,050 \mathrm{~Hz}$, which is as high as most human ears are able to hear.

Computing 44,100 values each second can be a demanding task for a CPU, even by today's standards. However, some signals used as inputs to other signal generating routines don't require such a high resolution, and can thus be generated at a lower rate. A good example of this is an amplitude envelope, which changes relatively slowly, and thus can be generated at a rate much lower than the audio rate. This rate is called the control rate (Ctr1Rate), and is set in the orchestra file header. The audio rate is usually a multiple of the control rate, but this is not a requirement.

Each instrument block contains a unique identifying integer, as well as an orchestra expression that defines the audio signal characterizing that instrument:

```
> type InstBlock = (Inst, OrcExp)
```

Recall that Inst is a type synonym for an Int. This value may be obtained from a string name and a name map using the function getId :: NameMap -> Name -> Maybe Int discussed earlier.

### 12.2.1 Orchestra Expressions

The data type OrcExp is the largest deviation that we will make from the actual CSound design. In CSound, instruments are defined using a sequence of statements that, in a piecemeal manner, define the various oscillators, summers, constants, etc. that make up an instrument. These pieces can be given names, and these names can be referenced from other statements. But despite this rather imperative, statement-oriented approach, it is acually completely functional. In other words, every CSound instrument can be rewritten as a single expression. It is this "expression language" that we capture in OrcExp. A pleasant attribute of the result is that CSound's ad hoc naming mechanism is replaced with Haskell's conventional way of naming things.

The entire OrcExp data type declaration is shown in Figure 10. In what follows, we describe each of the various constructors in turn.

```
data OrcExp = Const Float
            | Pfield Int
    | Plus OrcExp OrcExp
    | Minus OrcExp OrcExp
    | Times OrcExp OrcExp
    | Divide OrcExp OrcExp
    | Power OrcExp OrcExp
    | Modulo OrcExp OrcExp
    | Int OrcExp
    | Frac OrcExp
    | Neg OrcExp
    | Abs OrcExp
    | Sqrt OrcExp
    | Sin OrcExp
    | Cos OrcExp
    | Exp OrcExp
    Log OrcExp
    | AmpToDb OrcExp
    | DbToAmp OrcExp
    | PchToHz OrcExp
    | HzToPch OrcExp
    | GreaterThan OrcExp OrcExp OrcExp OrcExp
    | LessThan OrcExp OrcExp OrcExp OrcExp
    | GreaterOrEqTo OrcExp OrcExp OrcExp OrcExp
    | LessOrEqTo OrcExp OrcExp OrcExp OrcExp
    | Equals OrcExp OrcExp OrcExp OrcExp
    | NotEquals OrcExp OrcExp OrcExp OrcExp
    | MonoOut OrcExp
    | LeftOut OrcExp
    RightOut OrcExp
    | StereoOut OrcExp OrcExp
    | FrontLeftOut OrcExp
    | FrontRightOut OrcExp
    | RearRightOut OrcExp
    | RearLeftOut OrcExp
    | QuadOut OrcExp OrcExp OrcExp OrcExp
    | Line EvalRate Start Durn Finish
    | Expon EvalRate Start Durn Finish
    | LineSeg EvalRate Start Durn Finish [(Durn, Finish)]
    | ExponSeg EvalRate Start Durn Finish [(Durn, Finish)]
    | Env EvalRate Sig RTime Durn DTime RShape SAttn DAttn Steep
    | Phasor EvalRate Freq InitPhase
    | Tb7Lookup EvalRate Index Table IndexMode
    | TblLookupI EvalRate Index Table IndexMode
    | Osc EvalRate Amp Freq Table
    | OscI EvalRate Amp Freq Table
    | FMOsc Amp Freq CarFreq ModFreq ModIndex Table
    FMOscI Amp Freq CarFreq ModFreq ModIndex Table
    | SampOsc Amp Freq Table
    | Random EvalRate Amp
    | RandomHold EvalRate Amp HoldHz
    | RandomI EvalRate Amp HoldHz
    | GenBuzz Amp Freq NumHarms LoHarm Multiplier Table
    | Buzz Amp Freq NumHarms Table
    | Pluck Amp Freq Table DecayMethod DecArg1 DecArg2
    | Delay MaxDe1 AudioSig
    | DelTap TapTime DelLine
    | DelTapI TapTime DelLine
    | DelayW AudioSig
    | Comb AudioSig RevTime LoopTime
    | AlPass AudioSig RevTime LoopTime
    | Reverb AudioSig RevTime LoopTime
deriving (Show, Eq)
```


## Constants Const x represents the floating-point constant x .

P-field Arguments Pfie1d n refers to the $n$th p-field argument. Recall that all note characteristics, including pitch, volume, and duration, are passed into the orchestra file as p-fields. For example, to access the pitch, one would write Pfie1d 4. To make the access of these most common p-fields easier, we define the following constants:

```
> noteDur, notePit, noteVol :: OrcExp
> noteDur = Pfield 3
> notePit = Pfield 4
> noteVol = Pfield 5
```

It is also useful to define the following standard names, which are identical to those used in CSound:

```
> p1,p2,p3,p4,p5,p6,p7,p8,p9 :: OrcExp
> p1 = Pfield 1
> p2 = Pfield 2
> p3 = Pfield 3
> p4 = Pfield 4
> p5 = Pfield 5
> p6 = Pfield 6
> p7 = Pfield 7
> p8 = Pfield 8
> p9 = Pfield 9
```

Arithmetic and Transcendental Functions Arithmetic expressions are represented by the constructors Plus, Minus, Times, Divide, Power, and Modulo, each taking two OrcExps as arguments. In addition, there are a number of unary arithmatic functions: Int $x$ and Frac $x$ represent the integer and fractional parts, respectively, of $x$. Abs $x$, Neg $x$, Sqrt $x$, Sin $x$, and $\operatorname{Cos} x$ represent the absolute value, negation, square root, sine and cosine (in radians) of x , respectively. Exp x represents $e$ raised to the power x , and $\log \mathrm{x}$ is the natural logarithm of $x$.

To facilitate the use of these arithmetic functions, we can make OrcExp an instance of certain numeric type classes, thus providing more conventional names for the various operations.

```
> instance Num OrcExp where
```

```
> (+) = Plus
> (-) = Minus
> (*) = Times
> negate = Neg
> abs = Abs
> fromInteger i = Const (fromInteger i)
> -- omitted: signum
> instance Fractional OrcExp where
> (/) = Divide
> fromRational x = Const (fromRational x)
> instance Floating OrcExp where
> exp = Exp
> log = Log
> sqrt = Sqrt
> (**) = Power
> sin = Sin
> cos = Cos
> pi = Const pi
> tan x = Sin x / Cos x
> -- omitted: asin, acos, atan :: a -> a
> -- sinh, cosh, tanh :: a -> a
> -- asinh, acosh, atanh :: a -> a
```

For example, Plus (Pfield 3) (Power (Sin (Pfield 6)) (Const 2)) can nowbewritten simply as noteDur $+\sin p 6$ ** 2.

Pitch and Volume Coercions The next set of constructors represent functions that convert between different CSound pitch and volume representations. Recall that the CSNote constructor uses decibels for volume and "pch" notation for pitch. If these values are to be used as inputs into a signal generating or modifying routine, they must first be converted into units of raw amplitude and hertz, respectively. This is accomplished by the functions represented by DbToAmp and PchToHz, and their inverses are AmpToDb and HzToPch. Thus note that PchToHz (notePit +0.01 ) raises the pitch by one semitone, whereas PchToHz notePit + 0.01 raises the pitch by 0.01 Hz .

Comparison Operators OrcExp also includes comparison constructors GreaterThan, LessThan, GreaterOrEqTo, LessOrEqTo, Equa1s, and NotEqua1s. Each takes four OrcExp arguments: the values of the first two are compared, and if the result is true, the expression evaluates to the third argument; otherwise, it takes on the value of the fourth. ${ }^{4}$

[^3]Output Operators The next group of constructors represent CSound's output statements. The constructors are MonoOut, LeftOut, RightOut, StereoOut, FrontLeftOut, FrontRightOut, RearRightOut, RearLeft0ut, and QuadOut. StereoOut takes two OrcExp arguments, QuadOut takes four, and the rest take one.

The top-level of an instrument's OrcExp (i.e., the one in the InstB1ock value) will normally be an application of one of these. Furthermore, the constructor used must be in agreement with the number of output channels specified in the orchestra header-for example, using LeftOut when the header declares the resulting sound file to be mono will result in an error.

Signal Generation and Modification The most sophisticated OrcExp constructors are those that emulate CSound's signal generation and modification functions. There are quite a few of them, and they are all described here, although the reader is encouraged to read the CSound manual for further details.

Before defining each constructor, however, there are two general issues to discuss:
First, signals in CSound can be generated at three rates: the note rate (i.e., with every note event), the control rate, and the audio rate (we discussed the latter two earlier). Many of the signal generating routines can produce signals at more than one rate, so the rate must be specified as an argument. The following simple data structure serves this purpose:

```
> data EvalRate = NR -- note rate
> | CR -- control rate
> | AR -- audio rate
> deriving (Show, Eq)
```

Second, note in Figure 10 that this collection of constructors uses quite a few other type names. In all cases, however, these are simply type synonyms for OrcExp, and are used only for clarity. These type synonyms are listed here in one fell swoop:

```
> type Start = OrcExp
> type Durn = OrcExp
> type Finish = OrcExp
> type Sig = OrcExp
> type RTime = OrcExp
> type DTime = OrcExp
> type RShape = OrcExp
> type SAttn = OrcExp
> type DAttn = OrcExp
> type Steep = OrcExp
```

then add Boolean operators such as "and", "or", etc. It seems possible to do this in Haskore, but its translation into CSound would be more difficult, and thus we take the more conservative approach for now.

```
> type Freq = OrcExp
> type InitPhase = OrcExp
> type Index = OrcExp
> type Table = OrcExp
> type IndexMode = OrcExp
> type Amp = OrcExp
> type CarFreq = OrcExp
> type ModFreq = OrcExp
> type ModIndex = OrcExp
> type HoldHz = OrcExp
> type NumHarms = OrcExp
> type LoHarm = OrcExp
> type Multiplier = OrcExp
> type DecayMethod = OrcExp
> type DecArg1 = OrcExp
> type DecArg2 = OrcExp
> type MaxDe1 = OrcExp
> type AudioSig = OrcExp
> type TapTime = OrcExp
> type DelLine = OrcExp
> type RevTime = OrcExp
> type LoopTime = OrcExp
```

We can now discuss each constructor in turn:

1. Line evalrate start durn finish produces values along a straight line from start to finish. The values can be generated either at control or audio rate, and the line covers a period of time equal to durn seconds.
2. Expon is similar to Line, but Expon evalrate start durn finish produces an exponential curve instead of a straight line.
3. If a more elaborate signal is required, one can use the constructors LineSeg or ExponSeg, which take arguments of type EvalRate, Start, Durn, and Finish, as above, and also [(Durn, Finish)]. The first four arguments work as before, but only for the first of a number of segments. The subsequent segment lengths and endpoints are given in the fifth argument. A signal containing both straight line and exponential segments can be obtained by adding a LineSeg signal and ExponSeg signal together in an appropriate way.
4. Env evalrate sig rtime durn dtime rshape sattn dattn steep modifies the signal sig by applying an envelope to it. ${ }^{5}$ rtime and dtime are the rise time and decay time, respectively (in seconds), and durn is the overall duration. rshape is the identifier integer

[^4]of a function table storing the rise shape. sattn is the pseudo-steady state attenuation factor. A value between 0 and 1 will cause the signal to exopnentially decay over the steady period, a value greater than 1 will cause the signal to exponentially rise, and a value of 1 is a true steady state maintained at the last rise value. steep, whose value is usually between -0.9 and +0.9 , influences the steepness of the exponential trajectory. dattn is the attenuation factor by which the closing steady state value is reduced exponentially over the decay period, with value usually around 0.01 .
5. Phasor evalrate freq initphase generates a signal moving from 0 to 1 at a given frequency and starting at the given initial phase offset. When used properly as the index to a table lookup unit, Phasor can simulate the behavior of an oscillator.
6. Table lookup constructors Tb1Lookup and Tb1LookupI both take Eva1Rate, Index, Table, and IndexMode arguments. The IndexMode is either 0 or 1, differentiating between raw index and normalized index (zero to one); for convenience we define:

```
> rawIndex, normalIndex :: OrcExp
 rawIndex = 0.0
 normalIndex = 1.0
```

Both Tb1Lookup and Tb1LookupI return values stored in the specified table at the given index. The difference is that TblLookupI uses the fractional part of the index to interpolate between adjacent table entries, which generates a smoother signal at a small cost in execution time.

As mentioned, the output of a Phasor can be used as input to a table lookup to simulate an oscillator whose frequencey is controlled by the note pitch. This can be accomplished easily by the following piece of Haskore code:

```
osc = let index = Phasor AR (PchToHz notePit) 0.0
    in TblLookupI AR index table normalIndex
```

where table is some given function table ID. If osc is given as argument to an output operator such as MonoOut, then this OrcExp coupled with an instrument ID number (say, 1) produces a complete instrument block:

```
i1 = (1, MonoOut osc)
```

Adding a suitable Header would then give us a complete, though somewhat sparse, Orchestra value.
7. Instead of the above design we could use one of the built-in CSound oscillators, Osc and OscI, which differ in the same way as Tb1Lookup and Tb1LookupI. Each oscillator constructor takes arguments of type Eva1Rate, Amp (in raw amplitude), Freq (in Hertz), and Table. The result is a signal that oscillates through the function table at the given frequency. Thus the following value is equivalent to osc above:

```
osc' = OscI AR 1 (PchToHz notePit) table
```

8. It is often desirable to use the output of one oscillator to modulate the frequency of another, a process known as frequency modulation. FMOsc amp freq carfreq modfreq modindex table is a signal whose effective modulating frequency is freq*modfreq, and whose carrier frequency is freq*carfreq. modindex is the index of modulation, usually a value between 0 and 4, which determines the timbre of the resulting signal. FMOscI behaves similarly. Note that there is no EvalRate argument, since these functions work at audio rate only. The given function table normally contains a sine wave. This oscillator setup is known as the chowning FM setup.
9. Samp0sc amp freq tab1e oscillates through a table containing an AIFF sampled sound segment. This is the only time a table can have a length that is not a power of two, as mentioned earlier. Like FMOsc, SampOsc can only generate values at the audio rate.
10. Random evalrate amp produces a random number series between -amp and +amp at either control or audio rate. RandomHold evalrate amp holdhz does the same but will hold each number for holdhz cycles before generating a new one. RandomI evalrate amp holdhz will in addition provide straight line interpolation between successive numbers.
All the remaining constructors only operate at audio rate, and thus do not have Eva1Rate arguments.
11. GenBuzz amp freq numharms loharm multiplier table generates a signal that is an additive set of harmonically related cosine partials. freq is the fundamental frequency, numharms is the number of harmonics, and loharm is the lowest harmonic present. The amplitude coefficients of the harmonics are given by the exponential series a, a * multiplier, a * multiplier**2,..., a * multiplier**(numharms-1). The value $a$ is chosen so that the sum of the amplitudes is amp. table is a function table containing a cosine wave.
12. Buzz is a special case of GenBuzz in which loharm $=1.0$ and Multiplier $=1.0$. table is a function table containing a sine wave.
Note that the above two constructors have an analog in the generating routine GEN11 and the related function cosineHarms (see Section 12.1.1). cosineHarms stores into a table the same waveform that would be generated by Buzz or GenBuzz. However, although cosineHarms is more efficient, it has fixed arguments and thus lacks the flexibility of Buzz and GenBuzz in being able to vary the argument values with time.
13. Pluck amp freq table decaymethod decarg1 decarg2 is an audio signal that simulates a plucked string or drum sound, constructed using the Karplus-Strong algorithm. The signal has amplitude amp and frequency freq. It is produced by iterating through an internal buffer that initially contains a copy of table and is smoothed on every pass to simulate the natural decay of a plucked string. If 0.0 is used for table, then the initial buffer is filled with a random sequence. There are six possible decay modes:
(a) simple smoothing, which ignores the two arguments;
(b) stretched smoothing, which stretches the smoothing time by a factor of decarg1, ignoring decarg2;
(c) simple drum, where decarg1 is a "roughness factor" ( 0 for pitch, 1 for white noise; a value of 0.5 gives an optimal snare drum sound);
(d) stretched drum, which contains both roughness (decarg1) and stretch (decarg2) factors;
(e) weighted smoothing, in which decarg1 gives the weight of the current sample and decarg2 the weight of the previous one (decarg1+decarg2 must be $\leq 1$ ); and
(f) recursive filter smoothing, which ignores both arguments.

Here again are some helpful constants:

```
> simpleSmooth, stretchSmooth, simpleDrum, stretchDrum,
    weightedSmooth, filterSmooth :: OrcExp
> simpleSmooth = 1.0
> stretchSmooth = 2.0
> simpleDrum = 3.0
> stretchDrum = 4.0
> weightedSmooth = 5.0
> filterSmooth = 6.0
```

14. Delay deltime audiosig establishes a digital delay line, where audiosig is the source, and deltime is the delay time in seconds.

The delay line can also be tappedby De1ayTap de1time de11ine and De1ayTapI de1time delline, where deltime is the tap delay, and de11ine must be a delay line created by the Delay constructor above. Again, DelayTapI uses interpolation, and may take up to twice as long as DelayTap to run, but produces higher precision results and thus a cleaner signal.
(Note: the constructor DelayW is used in the translation described later to mark the end of a sequence of delay taps, and is not intended for use by the user.)
15. Reverberation can be added to a signal using Comb audiosig revtime looptime, AlPass audiosig revtime looptime, and Reverb audiosig revtime. revtime is the time in seconds it takes a signal to decay to $1 / 1000$ th of its original amplitude, and looptime is the echo density. Comb produces a "colored" reverb, A1Pass a "flat" reverb, and Reverb a "natural room" reverb.

### 12.2.2 Converting Orchestra Values to Orchestra Files

We must now convert the OrcExp values into a form which can be written into a CSound . sco file. As mentioned earlier, each signal generation or modification statement in CSound assigns
its result a string name. This name is used whenever another statement takes the signal as an argument. Names of signals generated at note rate must begin with the letter "i", control rate with letter "k", and audio rate with letter "a". The output statements do not generate a signal so they do not have a result name.

The function mkList is shown in Figure 11, and creates an entry in the list for every signal generating, modifying, or outputting constructor. It uses the following auxiliary functions:

```
> mkListA11 :: [OrcExp] -> [(Eva1Rate, OrcExp)]
> mkListAll = foldr (++) [] . map mkList
>
> addNames :: [(Eva1Rate,OrcExp)] -> [(Name,OrcExp)]
> addNames 1s = zipWith counter 1s [1..]
> where counter (er,x) n =
> let var = case er of
> AR -> 'a' : show n
> CR -> 'k' : show n
> NR -> 'i' : show n
> in (var,x)
```

Putting all of the above together, here is a function that converts an OrcExp into a list of proper name / OrcExp pairs. Each one of these will eventually result in one statement in the CSound orchestra file. (The result of mkList is reversed to ensure that a definition exists before it is used; and this must be done before nub is applied (which removes duplicates), for the same reason.)

```
> processOrcExp :: OrcExp -> [(Name, OrcExp)]
> processOrcExp = addNames . nub . procDelay . reverse . mkList
```

The function procDelay is used to process delay lines, which require special treatment because a delay line followed by a number of taps must be ended in Csound with a De1ayW command.

```
> procDelay :: [(Eva1Rate, OrcExp)] -> [(Eva1Rate, OrcExp)]
> procDelay [] = []
> procDelay (x@(AR, d@(Delay _ _)) : xs) = [x] ++ procTaps d xs ++ procDelay xs
> procDelay (x : xs) = x : procDelay xs
>
> procTaps :: OrcExp -> [(Eva1Rate, OrcExp)] -> [(Eva1Rate, OrcExp)]
> procTaps d@(Delay t sig) [] = [(AR, DelayW sig)]
```



Figure 11: The $86{ }^{2} \mathrm{~L}$ List Function

```
> procTaps d (x@(AR,DelTap t d7) : xs) =
> if d == dl then (mkList t ++ [x] ++ procTaps d xs)
> else procTaps d xs
> procTaps d (x@(AR,De1TapI t d1): xs) =
> if d == dl then (mkList t ++ [x] ++ procTaps d xs)
> else procTaps d xs
> procTaps d (x : xs) = procTaps d xs
```

The functions that follow are used to write the orchestra file. writeOrchestra is similar to writeScore: it asks the user for a file name, opens the file, writes the given orchestra value to the file, and then closes the file.

```
> writeOrchestra :: Orchestra -> IO ()
> writeOrchestra orch = do putStr "\nName your orchestra file "
> putStr "(.orc extension will be added): "
> name <- getLine
> h <- openFile (name ++ ".orc") WriteMode
> writeOrc h orch
> hClose h
```

writeOrc splits the task of writing the orchestra into two parts: writing the header and writing the instrument blocks

```
> writeOrc :: Handle -> Orchestra -> IO ()
> writeOrc h (hdr,ibs) = do writeHeader h hdr
> writeIBlocks h ibs
```

Writing the header is relatively simple, and is accomplished by the following function:

```
> writeHeader :: Handle -> Header -> IO ()
> writeHeader h (a,k,nc) =
> hPutStrLn h ( "sr = " ++ show a ++
> "\nkr = " ++ show k ++
> "\nksmps = " ++ show (float a / float k) ++
> "\nnchn1s = " ++ show nc)
```

writeIBlocks writes each instrument block using the function writeIBlock:

```
> writeIBlocks :: Handle -> [InstBlock] -> IO ()
> writeIBlocks h = mapM_ (writeIBlock h)
```

writeIBlock writes a single instrument block.

```
> writeIBlock :: Handle -> InstBlock -> IO ()
> writeIBlock h (num,ox) =
> do hPutStrLn h ("\ninstr " ++ show num)
> writeOrcExps h (processOrcExp ox ++ [("",ox)])
> hPutStrLn h "endin"
```

Recall that after processing, the OrcExp becomes a list of (Name, OrcExp) pairs. The last few functions write each of these named OrcExps as a statement in the orchestra file. Whenever a signal generation/modification constructor is encountered in an argument list of another constructor, the argument's string name is used instead, as found in the list of (Name, OrcExp) pairs.

### 12.3 An Orchestra Example

Figure 14 shows a typical CSound orchestra file. Figure 15 shows how this same functionality would be achieved in Haskore using an Orchestra value. Finally, Figure 16 shows the result of applying writeOrchestra to orc1 shown in Figure 15. Figures 14 and 16 should be compared: you will note that except for name changes, they are the same, as they should be.

## 13 Related and Future Research

Many proposals have been put forth for programming languages targeted for computer music composition [Dan89, Sch83, Col84, AK92, DFV92, HS92, CR84, OFLB94], so many in fact that it would be difficult to describe them all here. None of them (perhaps surprisingly) are based on a pure functional language, with one exception: the recent work done by Orlarey et al. at GRAME [OFLB94], which uses a pure lambda calculus approach to music description, and bears some resemblance to our effort. There are some other related approaches based on variants of Lisp, most notably Dannenberg's Fugue language [DFV92], in which operators similar to ours can be found but where the emphasis is more on instrument synthesis rather than note-oriented composition. Fugue also highlights the utility of lazy evaluation in certain contexts, but extra effort is needed to make this work in Lisp, whereas in a non-strict language such as Haskell it essentially comes "for free." Other efforts based on Lisp utilize Lisp primarily as a convenient vehicle for "embedded language design," and the applicative nature of Lisp is not exploited

```
writeOrcExps :: Handle -> [(Name,OrcExp)] -> IO ()
writeOrcExps h noes = mapM_ writeOrcExp noes where
writeOrcExp :: (Name,OrcExp) -> IO ()
writeOrcExp (nm, MonoOut x) = hPutStr h "out " » writeArgs [x]
writeOrcExp (nm, LeftOut x) = hPutStr h "outs1 " » writeArgs [x]
writeOrcExp (nm, RightOut x)
writeOrcExp (nm, StereoOut x1 x2)
writeOrcExp (nm, FrontLeftOut x)
writeOrcExp (nm, FrontRightOut x)
writeOrcExp (nm, RearRightOut x)
writeOrcExp (nm, RearLft0ut x) = hPutStr h "outq4 " "writeArgs [x]
= hPutStr h "outq4 " » writeArgs [x]
writeOrcExp (nm, QuadOut x1 x2 x3 x4) = hPutStr h "outq " > writeArgs [x1,x2,x3,x4]
writeOrcExp (nm, Line er x1 x2 x3) =
    hPutStr h (nm ++ " line ") » writeArgs [x1,x2,x3]
writeOrcExp (nm, Expon er x1 x2 x3) =
        hPutStr h (nm ++ " expon ") » writeArgs [x1,x2,x3]
writeOrcExp (nm, LineSeg er x1 x2 x3 xlist) =
        hPutStr h (nm ++ " linseg ") » writeArgs [x1,x2,x3] »
            writeOrcExps h (flattenTup1es2 xlist)
writeOrcExp (nm, ExponSeg er x1 x2 x3 xlist) =
        hPutStr h (nm ++ " expseg ") » writeArgs [x1,x2,x3] »
            writeOrcExps h (flattenTuples2 xlist)
writeOrcExp (nm, Env er x1 x2 x3 x4 x5 x6 x7 x8) =
        hPutStr h (nm ++ " env7px ") > writeArgs [x1,x2,x3,x4,x5,x6,x7,x8]
writeOrcExp (nm, Phasor er x1 x2) =
        hPutStr h (nm ++ " phasor ") » writeArgs [x1,x2]
writeOrcExp (nm, TblLookup er x1 x2 x3) =
        hPutStr h (nm ++ " table ") > writeArgs [x1,x2,x3]
writeOrcExp (nm, TblLookupI er x1 x2 x3) =
        hPutStr h (nm ++ " tablei ") > writeArgs [x1,x2,x3]
writeOrcExp (nm, Osc er x1 x2 x3) =
        hPutStr h (nm ++ " oscil ") » writeArgs [x1,x2,x3]
writeOrcExp (nm, OscI er x1 x2 x3) =
        hPutStr h (nm ++ " oscili ") » writeArgs [x1,x2,x3]
writeOrcExp (nm, FMOsc x1 x2 x3 x4 x5 x6) =
        hPutStr h (nm ++ " fosci1 ") > writeArgs [x1,x2,x3,x4,x5,x6]
writeOrcExp (nm, FMOscI x1 x2 x3 x4 x5 x6) =
        hPutStr h (nm ++ " foscili ") > writeArgs [x1,x2,x3,x4,x5,x6]
writeOrcExp (nm, SampOsc x1 x2 x3) =
        hPutStr h (nm ++ " losci1 ") > writeArgs [x1,x2,x3]
writeOrcExp (nm, Random er x) =
        hPutStr h (nm ++ " rand ") > writeArgs [x]
writeOrcExp (nm, RandomHold er x1 x2) =
        hPutStr h (nm ++ " randh ") » writeArgs [x1,x2]
writeOrcExp (nm, RandomI er x1 x2) =
        hPutStr h (nm ++ " randi ") » writeArgs [x1,x2]
writeOrcExp (nm, GenBuzz x1 x2 x3 x4 x5 x6) =
        hPutStr h (nm ++ "gbuzz ") > writeArgs [x1,x2,x3,x4,x5,x6]
writeOrcExp (nm, Buzz x1 x2 x3 x4) =
    hPutStr h (nm ++ " buzz ") » writeArgs [x1,x2,x3,x4]
writeOrcExp (nm, Pluck x1 x2 x3 x4 x5 x6) =
    hPutStr h (nm ++ " pluck ") » writeArgs [x1,x2,x2,x3,x4,x5,x6]
writeOrcExp (nm, Delay x1 x2) = hPutStr h (nm ++ " delayr ") » writeArgs [x1]
writeOrcExp (nm, DelayW x) = hPutStr h (" delayw ") » writeArgs [x]
writeOrcExp (nm, De1Tap x1 x2) = hPutStr h (nm ++ " deltap ") > writeArgs [x1]
> writeOrcExp (nm, DelTapI x1 x2) = hPutStr h (nm ++ " deltapi ") > writeArgs [x1]
> writeOrcExp (nm, Comb x1 x2 x3) = hPutStr h (nm ++ " comb ") ") writeArgs [x1,x2,x3]
> writeOrcExp (nm, AlPass x1 x2 x3) = hPutStr h (nm ++ " alpass ") > writeArgs [x1,x2,x3]
writeOrcExp (nm, Reverb x1 x2 x3) = hPutStr h (nm ++ " reverb ") > writeArgs [x1,x2,x3]
writeOrcExps _ = error "writeOrcExp: unknown constructor\n"
> writeArgs :: [OrcExp] -> IO ()
> writeArgs [x] = hPutStrLn h (showExp noes x)
> writeArgs (x:xs) = hPutStr h (showExp noes x ++ ", ") > writeArgs xs
```

Figure 12: The Function writeOrcExp

```
> showExp :: [(Name, OrcExp)] -> OrcExp -> String
> showExp _ (Const x) = show x
> showExp _ (Pfield p) = "p" ++ show p
> showExp xs (x1 'Plus` x2) = showBin xs " + " x1 x2
> showExp xs (x1 `Minus` x2) = showBin xs " - " x1 x2
> showExp xs (x1 'Times` x2) = showBin xs " * " x1 x2
> showExp xs (x1 `Divide` x2) = showBin xs " / " x1 x2
> showExp xs (x1 `Power` x2) = showBin xs " ^ " x1 x2
> showExp xs (x1 `Modu7o` x2) = showBin xs " % " x1 x2
> showExp xs (Int x) = "int(" ++ showExp xs x ++ ")"
> showExp xs (Frac x) = "frac(" ++ showExp xs x ++ ")"
> showExp xs (Abs x) = "abs(" ++ showExp xs x ++ ")"
> showExp xs (Neg x) = "-(" ++ showExp xs x ++ ")"
> showExp xs (Sqrt x) = "sqrt(" ++ showExp xs x ++ ")"
> showExp xs (Sin x) = "sin(" ++ showExp xs x ++ ")"
> showExp xs (Cos x) = "cos(" ++ showExp xs x ++ ")"
> showExp xs (Exp x) = "exp(" ++ showExp xs x ++ ")"
> showExp xs (Log x) = "log(" ++ showExp xs x ++ ")"
> showExp xs (AmpToDb x) = "dbamp(" ++ showExp xs x ++ ")"
> showExp xs (DbToAmp x) = "ampdb(" ++ showExp xs x ++ ")"
> showExp xs (PchToHz x) = "cpspch(" ++ showExp xs x ++ ")"
> showExp xs (HzToPch x) = "pchoct (octcps(" ++ showExp xs x ++ "))"
> showExp xs (GreaterThan x1 x2 x3 x4) = showComp xs " > " x1 x2 x3 x4
> showExp xs (LessThan x1 x2 x3 x4) = showComp xs " < " x1 x2 x3 x4
> showExp xs (GreaterOrEqTo x1 x2 x3 x4) = showComp xs " >= " x1 x2 x3 x4
> showExp xs (LessOrEqTo x1 x2 x3 x4) = showComp xs " <= " x1 x2 x3 x4
> showExp xs (Equals x1 x2 x3 x4) = showComp xs " == " x1 x2 x3 x4
> showExp xs (NotEqua1s x1 x2 x3 x4) = showComp xs " != " x1 x2 x3 x4
> showExp xs ox = case find (\(nm,oexp) -> ox==oexp) xs of
> Just (nm,_) -> nm
> Nothing -> error ("showExp: constructor not found\n")
>
> showBin xs s x1 x2 =
> "(" ++ showExp xs x1 ++ s ++ showExp xs x2 ++ ")"
> showComp xs s x1 x2 x3 x4 =
> "(" ++ showExp xs x1 ++ s ++ showExp xs x2 ++ " ? " ++
> showExp xs x3 ++ " : " ++ showExp xs x4 ++ ")"
```

Figure 13: The Function showExp

```
sr = 48000
kr = 24000
ksmps = 2
nchn1s = 2
instr 4
inote = cpspch(p5)
k1 env1px ampdb(p4), .001, p3, .05, 6, -.1, .01
k2 env1px ampdb(p4), .0005, .1, .1, 6, -.05, . 01
k3 env1px ampdb(p4), .001, p3, p3, 6, -.3, .01
a1 oscili k1, inote, 1
a2 oscili k1, inote * 1.004, 1
a3 oscili k2, inote * 16, 1
a4 oscili k3, inote, 5
a5 oscili k3, inote * 1.004, 5
outs (a2 + a3 + a4) * .75, (a1 + a3 + a5) * .75
endin
```

Figure 14: Sample CSound Orchestra File

```
> orc1 :: Orchestra
> orc1 =
> let hdr = (48000, 24000, 2)
> inote = PchToHz p5
> k1 = Env CR (DbToAmp p4) 0.001 p3 0.05 6 (-0.1) 0.01 0
> k2 = Env CR (DbToAmp p4) 0.0005 0.1 0.1 6 (-0.05) 0.01 0
> k3 = Env CR (DbToAmp p4) 0.001 p3 p3 6 (-0.3) 0.01 0
> a1 = OscI AR k1 inote 1
> a2 = OscI AR k1 (inote*1.004) 1
> a3 = OscI AR k2 (inote*16) 1
> a4 = OscI AR k3 inote 5
> a5 = OscI AR k3 (inote*1.004) 5
> out = StereoOut ((a2+a3+a4) * 0.75) ((a1+a3+a5) * 0.75)
> ib = (4,out)
> in (hdr,[ib])
> t1 = processOrcExp (snd (head (snd orc1)))
```

Figure 15: Haskore Orchestra Definition

```
sr = 48000
kr = 24000
ksmps = 2.0
nchn1s = 2
instr 4
k1 env1px ampdb(p4), 0.001, p3, p3, 6.0, -0.3, 0.01, 0.0
a2 osci k1, (cpspch(p5) * 1.004), 5.0
k3 env1px ampdb(p4), 0.0005, 0.1, 0.1, 6.0, -0.05, 0.01, 0.0
a4 osci k3, (cpspch(p5) * 16.0), 1.0
k5 env7px ampdb(p4), 0.001, p3, 0.05, 6.0, -0.1, 0.01, 0.0
a6 osci k5, cpspch(p5), 1.0
a7 osci k1, cpspch(p5), 5.0
a8 osci k5, (cpspch(p5) * 1.004), 1.0
outs (((a8 + a4) + a7) * 0.75), (((a6 + a4) + a2) * 0.75)
endin
```

Figure 16: Result of writeOrchestra orc1
well (for example, in Common Music the user will find a large number of macros which are difficult if not impossible to use in a functional style).

We are not aware of any computer music language that has been shown to exhibit the kinds of algebraic properties that we have demonstrated for Haskore. Indeed, none of the languages that we have investigated make a useful distinction between music and performance, a property that we find especially attractive about the Haskore design. On the other hand, Balaban describes an abstract notion (apparently not yet a programming language) of "music structure," and provides various operators that look similar to ours [Bal92]. In addition, she describes an operation called flatten that resembles our literal interpretation perform. It would be interesting to translate her ideas into Haskell; the match would likely be good.

Perhaps surprisingly, the work that we find most closely related to ours is not about music at all: it is Henderson's functional geometry, a functional language approach to generating computer graphics [Hen82]. There we find a structure that is in spirit very similar to ours: most importantly, a clear distinction between object description and interpretation (which in this paper we have been calling musical objects and their performance). A similar structure can be found in Arya's functional animation work [Ary94].

There are many interesting avenues to pursue with this research. On the theoretical side, we need a deeper investigation of the algebraic structure of music, and would like to express certain modern theories of music in Haskore. The possibility of expressing other scale types instead of the thus far unstated assumption of standard equal temperament scales is another area of investigation. On the practical side, the potential of a graphical interface to Haskore is appealing. We are also interested in extending the methodology to sound synthesis. Our primary goal currently, however, is to continue using Haskore as a vehicle for interesting algorithmic composition (for example, see [HB95]).

## A Convenient Functions for Getting Started With Haskore

```
module TestHaskore where
import Basics
import Performance
import HaskToMidi
import System( system )
import GeneralMidi
> import OutputMidi
> ----------------------------------------------------------------------------
> -- Given a PMap, Context, UserPatchMap, and file name, we can
> -- write a Music value into a midi file:
> ----------------------------------------------------------------------------
> mToMF :: PMap -> Context -> UserPatchMap -> String -> Music -> IO ()
> mToMF pmap c upm fn m =
> let pf = perform pmap c m
> mf = performToMidi pf upm
```

$>$

```
> in outputMidiFile fn mf
>
> ------------------------------------------------------------------------------------
-- Convenient default values and test routines
-----------------------------------------------------------------------------------
-- a default UserPatchMap
-- Note: the PC sound card I'm using is limited to 9 instruments
defUpm :: UserPatchMap
defUpm = [('"piano","Acoustic Grand Piano",1),
("vibes","Vibraphone",2),
("bass","Acoustic Bass",3),
("flute","Flute",4),
    ("sax","Tenor Sax",5),
    ("guitar","Acoustic Guitar (stee1)",6),
    ("violin","Viola",7),
    ("violins","String Ensemble 1",8),
    ("drums","Acoustic Grand Piano",9)]
        -- the GM name for drums is unimportant, on7y channe1 9
    -- a defau7t PMap that makes everything into a fancyPlayer
    defPMap :: String -> Player
defPMap pname =
    MkPlayer pname nf pf sf
    where MkPlayer _ nf pf sf = fancyPlayer
-- a defau7t Context
defCon :: Context
defCon = Context { cTime = 0,
    cPlayer = fancyPlayer,
    cInst = "piano",
    cDur = metro 120 qn,
    cKey = 0,
    cVo1 = 127 }
-- Using the defaults above, from a Music object, we can:
-- a) generate a performance
testPerf :: Music -> Performance
testPerf m = perform defPMap defCon m
testPerfDur :: Music -> (Performance, DurT)
testPerfDur m = perf defPMap defCon m
-- b) generate a midifile datatype
testMidi :: Music -> MidiFile
testMidi m = performToMidi (testPerf m) defUpm
-- c) generate a midifile
test :: Music -> IO ()
test m = outputMidiFile "test.mid" (testMidi m)
-- d) generate and play a midifile on Windows 95, Windows NT, or Linux
testWin95, testNT, testLinux :: Music -> IO ()
```

```
testWin95 m = do
    test m
    system "mplayer test.mid"
    return ()
testNT m = do
    test m
    system "mplay32 test.mid"
    return ()
testLinux m = do
    test m
    system "playmidi -rf test.mid"
    return ()
```

Alternatively, just run "test m" manually, and then invoke the midi player on your system using "play", defined below for NT:

```
play = do
    system "mplay32 test.mid"
    return ()
```

A more general function in the tradition of testMidi, makeMidi also takes a Context and a UserPatchMap.
> makeMidi :: (Music, Context, UserPatchMap) -> MidiFile
$>$ makeMidi ( $\mathrm{m}, \mathrm{c}$, upm) $=$ performToMidi (perform defPMap c m ) upm
> -- Some General Midi test functions (use with caution)

> -- a General Midi user patch map; i.e. one that maps GM instrument names
> -- to themselves, using a channel that is the patch number modulo 16.
> -- This is for use ONLY in the code that follows, o/w channel duplication
> -- is possible, which will screw things up in general.
> gmUpm :: UserPatchMap
$>g m U p m=\operatorname{map}(\backslash(g m n, n)->(g m n, g m n, \bmod n 16+1))$ genMidiMap
$>$
> -- Something to play each "instrument group" of 8 GM instruments;
> -- this function will play a $C$ major arpeggio on each instrument.
gmTest : : Int -> IO()
gmTest $i=1$ et $g M M=$ take 8 (drop ( $i * 8$ ) genMidiMap)
$\mathrm{mu}=$ line (map simple gMM)
simple (inm,_) = Instr inm cMajArp
$>\quad$ simple (inm,_) = Instr inm cMajArp
> in mToMF defPMap defCon gmUpm "test.mid" mu

## B Examples of Haskore in Action

> module HaskoreExamples (module HaskoreExamples, module Haskore, module IO,

```
> modu7e ChildSong6, modu7e Se7fSim)
> where
> import Haskore
> import IO
> import ChildSong6
> import SelfSim
> import Ssf
```

Simple examples of Haskore in action. Note that this module also imports modules ChildSong6, SelfSim, and Ssf.

From the tutorial, try things such as pr12, cMajArp, cMajChd, etc. and try applying inversions, retrogrades, etc. on the same examples. Also try "childSong6" imported from module ChildSong. For example:

```
> t0 = test (Instr "piano" childSong6)
```

C Major scale for use in examples below:
> cMajScale $=$ Tempo 2
$>\quad$ (line [c 4 en [], d 4 en [], e 4 en [], f 4 en [],
g 4 en [], a 4 en [], b 4 en [], c 5 en []])
cms' = line [c 4 en [], d 4 en [], e 4 en [], f 4 en [],
g 4 en [], a 4 en [], b 4 en [], c 5 en []]
$>$
> cms = cMajScale
Test of various articulations and dynamics:

```
> t1 = test (Instr "percussion"
> (Phrase [Art (Staccato 0.1)] cms :+:
> cms :+:
> Phrase [Art (Legato 1.1)] cms ))
temp = Instr "piano" (Phrase [Dyn (Crescendo 4.0)] (c 4 en []))
mu2 = Instr "vibes"
    (Phrase [Dyn (Diminuendo 0.75)] cms :+:
        Phrase [Dyn (Crescendo 4.0), Dyn (Loudness 25)] cms)
t2 = test mu2
> t3 = test (Instr "flute"
> (Phrase [Dyn (Accelerando 0.3)] cms :+:
> Phrase [Dyn (Ritardando 0.6)] cms ))
```

```
A function to recursively apply transformations f (to elements in a
sequence) and g (to accumulated phrases):
> rep :: (Music -> Music) -> (Music -> Music) -> Int -> Music -> Music
> rep f g 0 m = Rest 0
> rep f g n m = m :=: g (rep f g (n-1) (f m))
An example using "rep" three times, recursively, to create a "cascade"
of sounds.
> run = rep (Trans 5) (delay tn) 8 (c 4 tn [])
> cascade = rep (Trans 4) (delay en) }8\mathrm{ run
> cascades = rep id (delay sn) 2 cascade
> t4' x = test (Instr "piano" x)
> t4 = test (Instr "piano"
> (cascades :+: revM cascades))
What happens if we simply reverse the f and g arguments?
> run' = rep (delay tn) (Trans 5) 4 (c 4 tn [])
> cascade' = rep (delay en) (Trans 4) 6 run'
> cascades' = rep (delay sn) id 2 cascade'
> t5 = test (Instr "piano" cascades')
```

Example from the SelfSim module.

```
> t10s = test (rep (delay durss) (Trans 4) 2 ss)
```

Example from the ChildSong6 module.
> cs6 $=$ test childSong6

Example from the Ssf (Stars and Stripes Forever) module.

```
> ssf0 = test ssf
```

Midi percussion test. Plays all "notes" in a range. (Requires adding an instrument for percussion to the UserPatchMap.)
> drums a b = Instr "drums"
$>\quad$ (line (map (\p-> Note (pitch p) sn []) [a..b]))

```
> t11 a b = test (drums a b)
```

Test of cut and shorten.
$>$ t12 $=$ test (cut 4 childSong6)
> t12a = test (cms /=: childSong6)

Tests of the tril1 functions.

```
> t13note = (Note (C,5) qn [])
> t13 = test (tri11 1 sn t13note)
> t13a = test (tri11, 2 dqn t13note)
> t13b = test (tri11n 1 5 t13note)
> t13c = test (tril1n' 3 7 t13note)
> t13d = test (ro11 tn t13note)
> t13e = test (Tempo (2/3) (Trans 2 (Instr "piano" (tri11n' 2 7 t13note))))
```

Tests of drum.

```
> t14 = test (Instr "Drums" (perc AcousticSnare qn []))
```

> -- a "funk groove"
$>$ t14b = let p1 = perc LowTom qn []
p2 = perc AcousticSnare en []
in test (Tempo 3 (Instr "Drums" (cut 8 (repeatM
( (p1 :+: qnr :+: p2 :+: qnr :+: p2 :+:
p1 :+: p1 :+: qnr :+: p2 :+: enr)
:=: roll en (perc ClosedHiHat 2 []))))))
-- a "jazz groove"
t14c = 1et p1 = perc CrashCymbal2 qn []
p2 = perc AcousticSnare en []
p3 = perc LowTom qn []
in test (Tempo 3 (Instr "Drums" (cut 4 (repeatM
((p1 :+: Tempo (3\%2) (p2 :+: enr :+: p2))
:=: (p3 :+: qnr)) ))))
t14d $=$ let $\mathrm{p} 1=$ perc LowTom en []
p2 = perc AcousticSnare hn []
in test (Instr "Drums"
( roll tn p1
:+: p1
:+: p1
:+: Rest en
:+: roll tn p1

```
> :+: p1
> :+: p1
> :+: Rest qn
> :+: roll tn p2
> :+: p1
> :+: p1 ))
```

Tests of the MIDI interface.
Music into a MIDI file.
$>$ tab $m=$ do
> outputMidiFile "test.mid" \$ makeMidi (m, defCon, defUpm)
Music to a MidiFile datatype and back to Music.
> tad m = readMidi (testMidi m)
A MIDI file to a MidiFile datatype and back to a MIDI file.
$>$ tcb file $=$ do
$>\quad x<-$ loadMidiFile file
> outputMidiFile "test.mid" x
MIDI file to MidiFile datatype.
$>$ tc file $=$ do
$>\quad x<-$ loadMidiFile file
> print $x$
MIDI file to Music, a UserPatchMap, and a Context.
> tcd file = do
$>\quad x<-$ loadMidiFile file
$>\quad$ print $\$$ fst3 $\$$ readMidi $x$
$>\quad$ print $\$$ snd3 $\$$ readMidi x
$>\quad$ print $\$$ thd3 $\$$ readMidi x
A MIDI file to Music and back to a MIDI file.
> tcdab file $=$ do
$>\quad x<-$ loadMidiFile file
> outputMidiFile "test.mid" \$ makeMidi \$ readMidi x
> getTracks (MidiFile _ _ trks) = trks
$>$
$>$ fst3 $(a, b, c)=a$
$>$ snd3 $(a, b, c)=b$
$>$ thd3 $(a, b, c)=c$

## C Partial Encoding of Chick Corea's "Children's Song No. 6"

```
> module ChildSong6 where
> import Haskore
>
> -- note updaters for mappings
> fd d n = n d v
> vol n = n v
> v = [Volume 80]
> lmap f 1 = line (map f 1)
>
> -- repeat something n times
times 1 m = m
times (n+1) m = m :+: (times n m)
>
> -- Baseline:
b1 = 1map (fd dqn) [b 3, fs 4, g 4, fs 4]
b2 = 1map (fd dqn) [b 3, es 4, fs 4, es 4]
b3 = 1map (fd dqn) [as 3, fs 4, g 4, fs 4]
bassLine = times 3 b1 :+: times 2 b2 :+: times 4 b3 :+: times 5 b1
>
> -- Main Voice:
v1 = v1a :+: v1b
v1a = 1map (fd en) [a 5, e 5, d 5, fs 5, cs 5, b 4, e 5, b 4]
v1b = 1map vo1 [cs 5 tn, d 5 (qn-tn), cs 5 en, b 4 en]
>
v2 = v2a :+: v2b :+: v2c :+: v2d :+: v2e :+: v2f
v2a = 1map vol [cs 5 (dhn+dhn), d 5 dhn,
                            f 5 hn, gs 5 qn, fs 5 (hn+en), g 5 en]
v2b = 1map (fd en) [fs 5, e 5, cs 5, as 4] :+: a 4 dqn v :+:
    1map (fd en) [as 4, cs 5, fs 5, e 5, fs 5, g 5, as 5]
v2c = 1map vol [cs 6 (hn+en), d 6 en, cs 6 en, e 5 en] :+: enr :+:
lmap vol [as 5 en, a 5 en, g 5 en, d 5 qn, c 5 en, cs 5 en]
v2d = 1map (fd en) [fs 5, cs 5, e 5, cs 5, a 4, as 4, d 5, e 5, fs 5] :+:
7map vol [fs 5 tn, e 5 (qn-tn), d 5 en, e 5 tn, d 5 (qn-tn),
cs 5 en, d 5 tn, cs 5 (qn-tn), b 4 (en+hn)]
v2e = 1map vol [cs 5 en, b 4 en, fs 5 en, a 5 en, b 5 (hn+qn), a 5 en,
                        fs 5 en, e 5 qn, d 5 en, fs 5 en, e 5 hn, d 5 hn, fs 5 qn]
v2f = Tempo (3/2) (7map vol [cs 5 en, d 5 en, cs 5 en]) :+: b 4 (3*dhn+hn) v
mainVoice = times 3 v1 :+: v2
-- Putting it all together:
childSong6 = Instr "piano" (Tempo 3 (Phrase [Dyn SF] bassLine :=: mainVoice))
```


## D Example of Simple Self-Similar (Fractal) Music

```
> module SelfSim where
>
> import Haskore
An example of self-similar, or fractal, music.
> data Cluster = C1 SNote [Cluster] -- this is called a Rose tree
> type Pat = [SNote]
> type SNote = [(AbsPitch,Dur)] -- i.e. a chord
>
> sim :: Pat -> [Cluster]
> sim pat = map mkCluster pat
> where mkCluster notes = Cl notes (map (mkCluster . addmult notes) pat)
>
addmult pds iss = zipWith addmult' pds iss
                                    where addmult' (p,d) (i,s) = (p+i,d*s)
simFringe n pat = fringe n (C1 [(0,0)] (sim pat))
fringe 0 (C1 note cls) = [note]
fringe n (C1 note cls) = concat (map (fringe (n-1)) cls)
>
> -- this just converts the result to Haskore:
> simToHask s = let mkNote (p,d) = Note (pitch p) d []
> in line (map (chord . map mkNote) s)
-- and here are some examples of it being applied:
sim1 n = Instr "bass"
        (Trans 36
                            (Tempo 4 (simToHask (simFringe n pat1))))
t6 = test (sim1 4)
sim2 n = Instr "piano"
(Trans 53
                            (Tempo 4 (simToHask (simFringe n pat2))))
t7 = test (sim2 4)
sim12 n = sim1 n :=: sim2 n
t8 = test (sim12 4)
sim3 n = Instr "vibes"
    (Trans 48
    (Tempo 4 (simToHask (simFringe n pat3))))
t9 = test (sim3 3)
sim4 n = (Trans 60
```

```
> (Tempo 2 (simToHask (simFringe n pat4'))))
sim4s n = let s = sim4 n
            11 = Instr "flute" s
            12 = Instr "bass" (Trans (-36) (revM s))
        in 11 :=: 12
ss = sim4s 3
durss = dur ss
t10 = test ss
pat1,pat2,pat3,pat4,pat4' :: [SNote]
pat1 = [[(0,1.0)],[(4,0.5)],[(7,1.0)],[(5,0.5)]]
pat2 = [[(0,0.5)],[(4,1.0)],[(7,0.5)],[(5,1.0)]]
pat3 = [[(2,0.6)],[(5,1.3)],[(0,1.0)],[(7,0.9)]]
pat4' = [[(3,0.5)],[(4,0.25)],[(0,0.25)],[(6,1.0)]]
pat4 = [[(3,0.5),(8,0.5),(22,0.5)],[(4,0.25),(7,0.25),(21,0.25)],
                        [(0,0.25),(5,0.25),(15,0.25)],[(6,1.0),(9,1.0),(19,1.0)]]
```


## E General Midi

```
module Genera1Midi where
type GenMidiName = String
type GenMidiTable = [(GenMidiName,Int)]
genMidiMap :: GenMidiTable
genMidiMap =[
("Acoustic Grand Piano",0), ("Bright Acoustic Piano",1),
("Electric Grand Piano",2), ("Honky Tonk Piano",3),
("Rhodes Piano",4),
("Harpsichord",6),
("Celesta",8),
    ("Music Box",10),
    ("Marimba",12), ("Xylophone",13),
    ("Tubular Be11s",14), ("Dulcimer",15),
    ("Hammond Organ",16), ("Percussive Organ",17),
    ("Rock Organ",18), ("Church Organ",19),
    ("Reed Organ",20), ("Accordion",21),
    ("Harmonica",22), ("Tango Accordion",23),
    ("Acoustic Guitar (nylon)",24), ("Acoustic Guitar (stee1)",25),
> ("Electric Guitar (jazz)",26), ("Electric Guitar (clean)",27),
> ("Electric Guitar (muted)",28), ("Overdriven Guitar",29),
> ("Distortion Guitar",30), ("Guitar Harmonics",31),
> ("Acoustic Bass",32), ("Electric Bass (fingered)",33),
> ("Electric Bass (picked)",34), ("Fretless Bass",35),
```

```
> ("Slap Bass 1",36),
> ("Synth Bass 1",38),
> ("Violin",40),
> ("Cello",42),
> ("Tremolo Strings",44),
> ("Orchestral Harp",46),
> ("String Ensemble 1",48),
> ("Synth Strings 1",50),
> ("Choir Aahs",52),
> ("Synth Voice",54),
> ("Trumpet",56),
> ("Tuba",58),
> ("French Horn",60),
> ("Synth Brass 1",62),
> ("Soprano Sax",64),
> ("Tenor Sax",66),
> ("Oboe",68),
> ("English Horn",70),
> ("Piccolo",72),
> ("Recorder",74),
> ("Blown Bottle",76),
> ("Whistle",78),
> ("Lead 1 (square)",80),
> ("Lead 3 (calliope)",82),
> ("Lead 5 (charang)",84),
> ("Lead 7 (fifths)",86),
> ("Pad 1 (new age)",88),
> ("Pad 3 (polysynth)",90),
> ("Pad 5 (bowed)",92),
> ("Pad 7 (halo)",94),
> ("FX1 (train)",96),
> ("FX3 (crysta1)",98),
> ("FX5 (brightness)",100),
> ("FX7 (echoes)",102),
> ("Sitar",104),
> ("Shamisen",106),
> ("Kalimba",108),
> ("Fiddle",110),
> ("Tinkle Bel1",112),
> ("Steel Drums",114),
> ('Taiko Drum",116),
> ("Synth Drum",118),
> ("Guitar Fret Noise",120),
> ("Seashore",122),
> ("Telephone Ring",124),
> ("Applause",126),
```

("S1ap Bass 2",37),
("Synth Bass 2",39),
("Viola",41),
("Contrabass",43),
("Pizzicato Strings",45),
("Timpani",47),
("String Ensemble 2",49),
("Synth Strings 2",51),
("Voice Oohs",53),
("Orchestra Hit",55),
("Trombone", 57),
("Muted Trumpet",59),
("Brass Section",61),
("Synth Brass 2",63),
("A1to Sax",65),
("Baritone Sax",67),
("Bassoon", 69),
("Clarinet",71),
("F1ute",73),
("Pan Flute",75),
("Shakuhachi", 77),
("Ocarina", 79),
("Lead 2 (sawtooth)",81),
("Lead 4 (chiff)",83),
("Lead 6 (voice)",85),
("Lead 8 (bass+1ead)",87),
("Pad 2 (warm)",89),
("Pad 4 (choir)",91),
("Pad 6 (metallic)",93),
("Pad 8 (sweep)",95),
("FX2 (soundtrack)",97),
("FX4 (atmosphere)",99),
("FX6 (gob1ins)", 101),
("FX8 (sci-fi)",103),
("Banjo", 105),
("Koto", 107),
("Bagpipe", 109),
("Shanai",111),
("Agogo",113),
("Woodblock",115),
("Melodic Drum",117),
("Reverse Cymbal",119),
("Breath Noise",121),
("Bird Tweet",123),
("Helicopter", 125),
("Gunshot", 127)]

## References

[AK92] D.P. Anderson and R. Kuivila. Formula: A programming language for expressive computer music. In Denis Baggi, editor, Computer Generated Music. IEEE Computer Society Press, 1992.
[Ary94] K. Arya. A functional animation starter-kit. Journal of Functional Programming, 4(1):1-18, 1994.
[Bal92] M. Balaban. Music structures: Interleaving the temporal and hierarchical aspects of music. In M. Balaban, K. Ebcioglu, and O. Laske, editors, Understanding Music With AI, pages 110-139. AAAI Press, 1992.
[BW88] R. Bird and P. Wadler. Introduction to Functional Programming. Prentice Hall, New York, 1988.
[Col84] D. Collinge. Moxie: A languge for computer music performance. In Proc. Int'l Computer Music Conference, pages 217-220. Computer Music Association, 1984.
[CR84] P. Cointe and X. Rodet. Formes: an object and time oriented system for music composition and synthesis. In Proceedings of the 1984 ACM Symposium on Lisp and Functional Programmming, pages 85-95. ACM, 1984.
[Dan89] R.B. Dannenberg. The Canon score language. Computer Music Journal, 13(1):47-56, 1989.
[DFV92] R.B. Dannenberg, C.L. Fraley, and P. Velikonja. A functional language for sound synthesis with behavioral abstraction and lazy evaluation. In Denis Baggi, editor, Computer Generated Music. IEEE Computer Society Press, 1992.
[For73] A. Forte. The Structure of Atonal Music. Yale University Press, New Haven, CT, 1973.
[HB95] P. Hudak and J. Berger. A model of performance, interaction, and improvisation. In Proceedings of International Computer Music Conference. Int'l Computer Music Association, 1995.
[Hen82] P. Henderson. Functional geometry. In Proceedings of the 1982 ACM Symposium on Lisp and Functional Programmming. ACM, 1982.
[HF92] P. Hudak and J. Fasel. A gentle introduction to Haskell. ACM SIGPLAN Notices, 27(5), May 1992.
[HMGW96] P. Hudak, T. Makucevich, S. Gadde, and B. Whong. Haskore music notation - an algebra of music. Journal of Functional Programming, 6(3), June 1996. available via ftp://nebula.systemsz.cs.yale.edu/pub/yale-fp/papers/haskore/hmn-1hs.ps.
[HS92] G. Haus and A. Sametti. Scoresynth: A system for the synthesis of music scores based on petri nets and a music algebra. In Denis Baggi, editor, Computer Generated Music. IEEE Computer Society Press, 1992.
[IMA90] Midi 1.0 detailed specification: Document version 4.1.1, February 1990.
[JB91] D. Jaffe and L. Boynton. An overview of the sound and music kits for the NeXT computer. In S.T. Pope, editor, The Well-Tempered Object, pages 107-118. MIT Press, 1991.
[OFLB94] O. Orlarey, D. Fober, S. Letz, and M. Bilton. Lambda calculus and music calculi. In Proceedings of International Computer Music Conference. Int'l Computer Music Association, 1994.
[Sch83] B. Schottstaedt. Pla: A composer's idea of a language. Computer Music Journal, 7(1):11-20, 1983.
[Ver86] B. Vercoe. Csound: A manual for the audio processing system and supporting programs. Technical report, MIT Media Lab, 1986.


[^0]:    ${ }^{1}$ (1) The NeXT music platform is obsolete. (2) There exists a translation to csound for an earlier version of Haskore that I would happily share with anyone interested. (3) We have abandoned CMN entirely, as there are now better candidates for notation packages into which Haskore could be mapped.

[^1]:    ${ }^{2}$ I prefer to call these "musical objects" rather than "musical values" because the latter may be confused with musical aesthetics.

[^2]:    ${ }^{3}$ For example, Forte prescribes normal forms for chords in an atonal setting [For73].

[^3]:    ${ }^{4}$ This design emulates that of CSound. A more conventional design would have comparison operators that return a Boolean value, and a conditional expression to choose between two values based on a Boolean. One could

[^4]:    ${ }^{5}$ Although this function is widely-used in CSound, the same effect can be accomplished by creating a signal that is a combination of straight line and exponential curve segments, and multiplying it by the signal to be modified.

