



Semi-automated extraction of  
expressive performance information from  
acoustic recordings of piano music

---

**Andrew Earis**



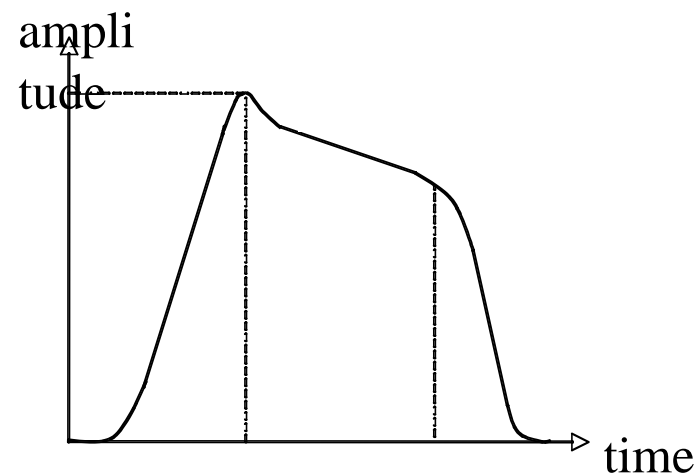
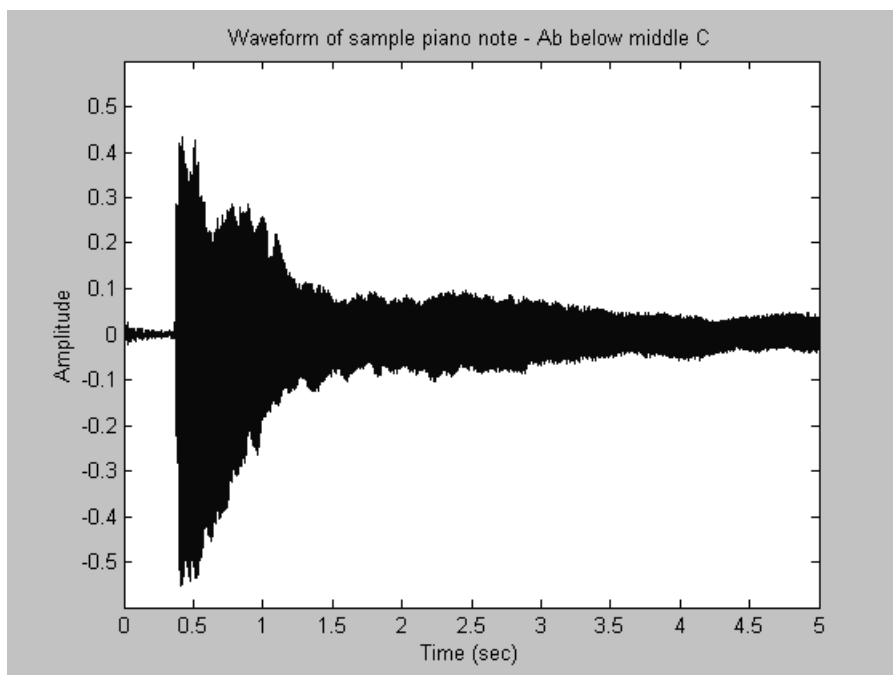
# Outline

---

- Parameters of expressive piano performance
- Scientific techniques: Fourier transform & wavelet transform
- CHARM: The semi-automated system

# Measurable features of expressive performance

- *Timing (note onset and offset)*
- *Dynamics (variations in intensity of notes and chords)*
- *Pitch*





# Timing parameterisation

---

- Note onset time. The time at which the note is played.
- Note offset time. The time at which the note is released.
- Attack duration. The time from the initial sound until its amplitude is at a maximum.
- Sustain duration. The time from maximum sound amplitude until the note is released.
- Decay time. The time from when the note is released until the sound ceases.



# Dynamics parameterisation

---

- Power of note
- Maximum amplitude (amplitude of attack)
- Sustain amplitude (amplitude, or change of amplitude, of the sustain)



# Pitch parameterisation

---

- Fundamental frequency of note
- Frequency of individual partials (and how they deviate from the precise harmonic frequencies)
- Vibrato
  - Frequency of vibrato
  - Amplitude of vibrato
- Glissando



# Aims

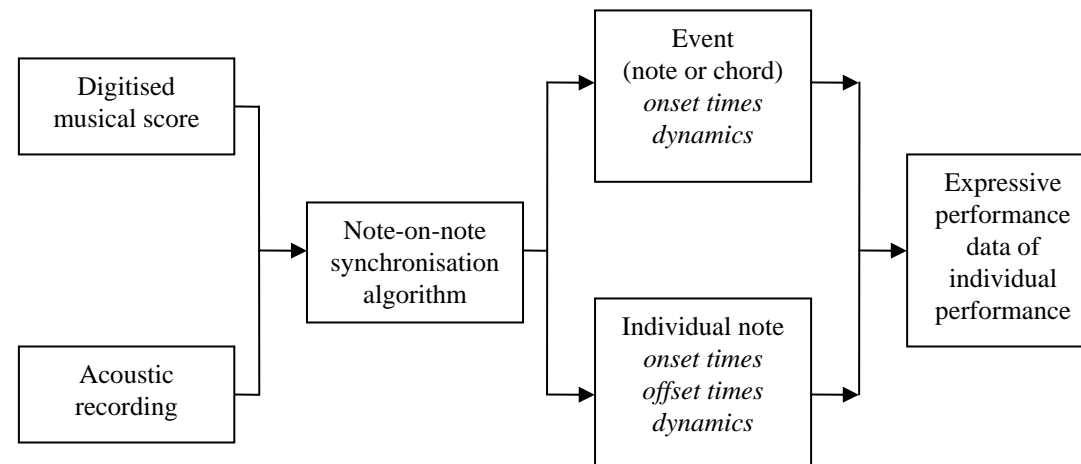
---

- to extract the expressive performance information from acoustic recordings of piano music with reference to a digitised version of the musical score of the work being performed.



# Simple flowchart

---







# Methods

---

The location in time of individual notes and chords within an acoustic recording must be identified and related to a digitised score of the work being performed.

This is a classical time-frequency analysis problem



# Fundamental research issues

---

- A number of constraints on the time-frequency resolution required.
  - Required frequency resolution
    - determined by the interval between adjacent semitones
    - proportional to the frequency of the lower semitone
    - Spacing varies from about 1.6Hz at the bottom of the piano keyboard to over 120Hz at the top
    - the fundamental frequencies ranging from 27.5Hz to just over 4kHz.
  - Required timing resolution
    - The shortest note duration in rapid note passages is around 50ms.
    - The human ear is capable of perceiving differences of as little as 10ms.

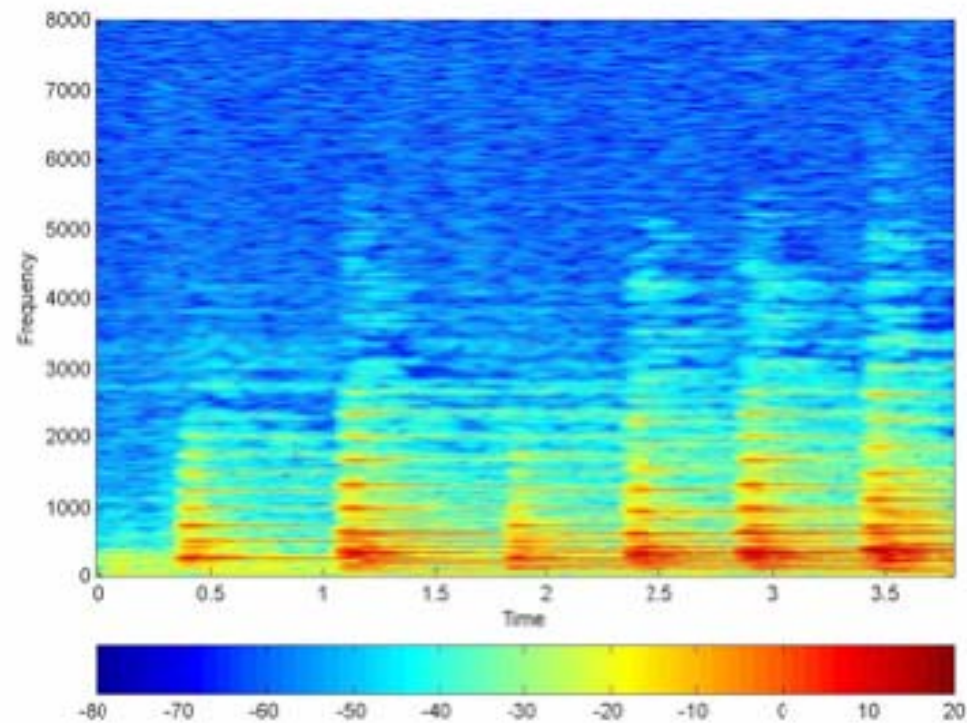


# Possible approaches

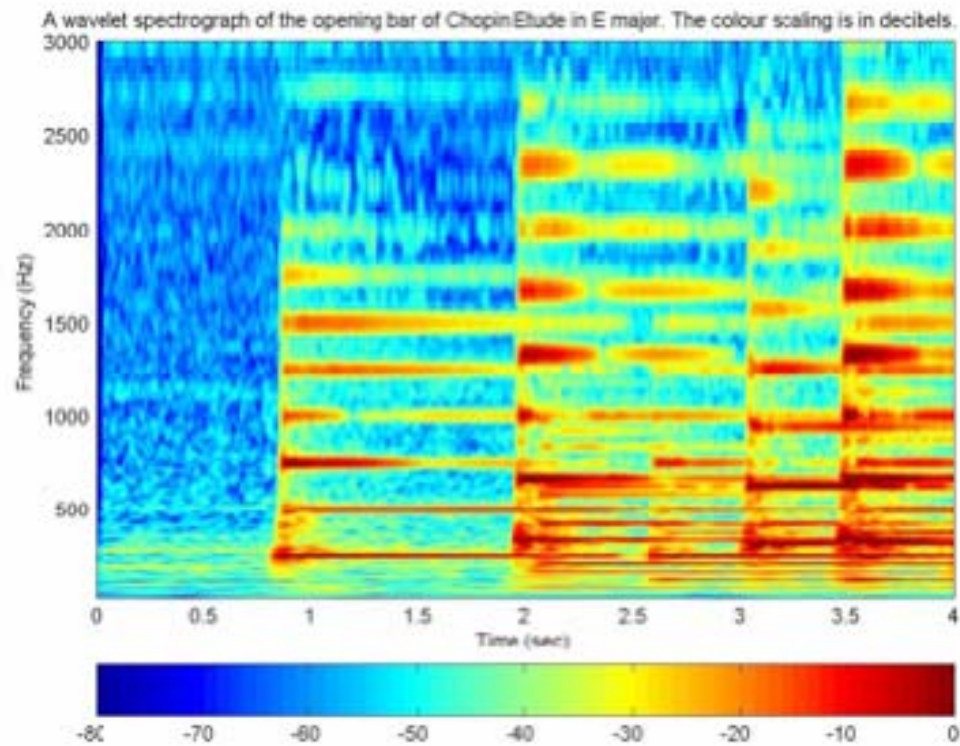
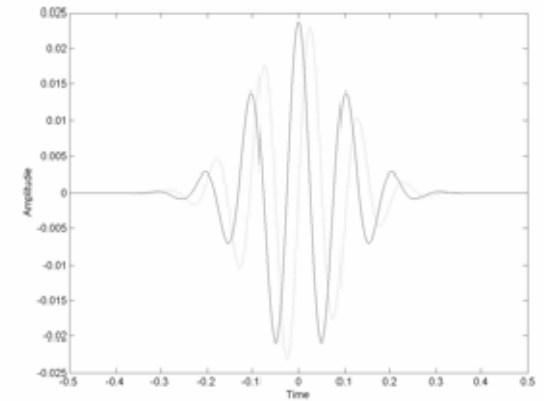
---

- Filterbank techniques
- Time-frequency analysis (using Fourier Transforms, wavelets and spectrographs)
- Modelling

# Short Time Fourier analysis



# Wavelet analysis





# Problems with 'simple approach'

---

- Synchronization between score and acoustic recording relies on stepping through each event in the score in turn
- If a particular note or chord in the score is not played, the whole system can break down
- Other problems
  - too much noise in the recording
  - a high degree of rubato
  - chords are played with too much asynchrony
- The longer the passage of music to be analysed, the more likely it is that such a system will break down.
- Time-consuming manual checking of the data would also be required to ensure accuracy.



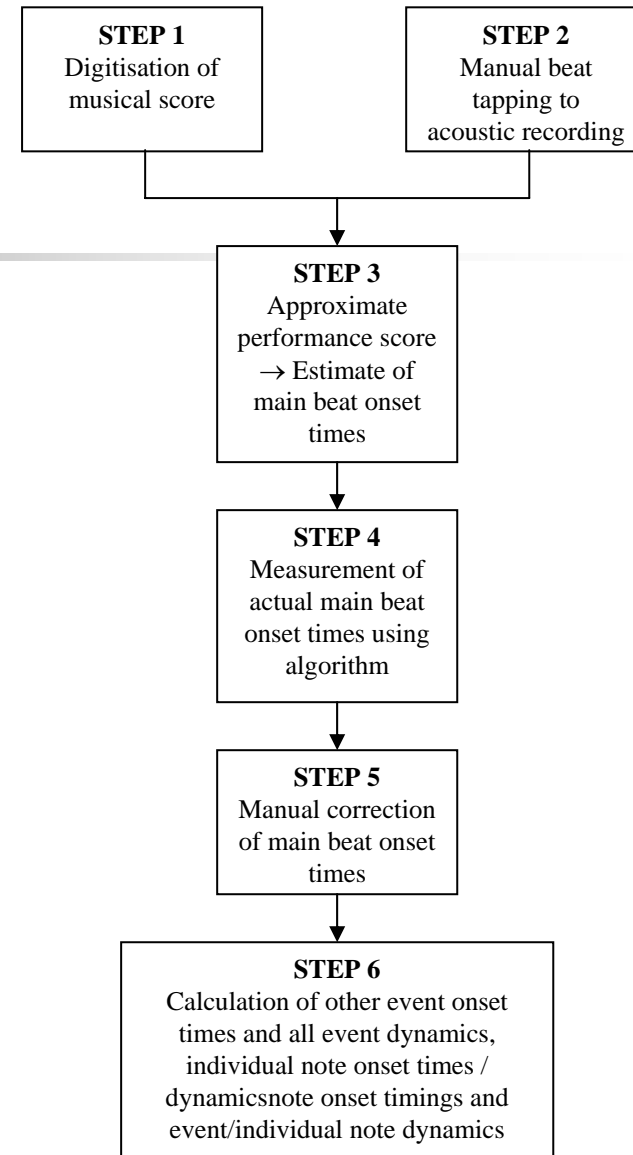
# CHARM: Semi-automated system

---

- A semi-automated expression extractor, requiring user input for checking and verifying results as the algorithm proceeds i.e. note-on-note.
- Input to system: MIDI score, acoustic waveform
- Technology incorporated: filterbank/wavelet-based note onset detector
- Note/event parameter measurements: onset time, dynamics



# Algorithm plan







# Advantages

---

- Quick to use, results can be verified as they go along – know they are correct!
- Very easy to use
- Wide range of scores can be used
- Quick and easy to develop and automate further
- Results are verified by the user as measurements are made, ensuring that all results are correct and can be used reliably for musicological analysis
- Further enhancements can be made (e.g. increase level of automation, sophistication of parameters)



# STEP 1

---

- Digitisation of musical score
  - Score is scanned using a standard flatbed scanner
  - Converted to symbolic data with Sharpeye
  - Errors are corrected manually
  - Symbolic data is then converted to the Humdrum data format
  - This is outputted as ASCII tabular data



## STEP 2

---

- Manual beat tapping to acoustic recording
  - Standard beat tapping procedures
  - backbeat.exe
  - average of 20 complete taps initially!



## STEP 3

---

- Averaged manual beat tapped data combined with the digitized musical score
  - -> Approximate performance score

Column	Description	
1	abstime	Average absolute time in ms of beat tapped timings
2	duration	Expected duration of note in ms based on score duration
3	note	MIDI number of note pitch
4	metric level	1 = downbeat 0 = beat -1 = offbeat
5	bar number	
6	absolute beat	measured from beginning of work



# Approximate performance score: Mazurka in A minor op 7 no 2

---

1912	646	76	1	0	0
2558	463	77	0	1	1
3021	154	76	-1	1	1.75
3175	603	57	0	1	2
3175	603	62	0	1	2
3175	603	65	0	1	2
3175	603	74	0	1	2
3778	652	57	1	1	3
3778	652	62	1	1	3
3778	652	65	1	1	3
3778	652	77	1	1	3
4430	1111	77	0	2	4
4914	627	60	0	2	5
4914	627	65	0	2	5



## STEP 4

---

- Correction of 'main beat' onset times (from beat tapped times) using wavelet-based algorithm, programmed in MATLAB

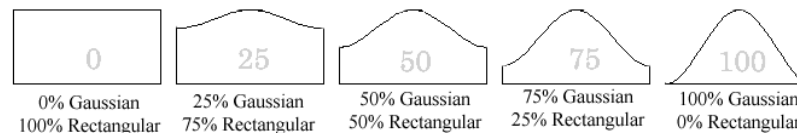
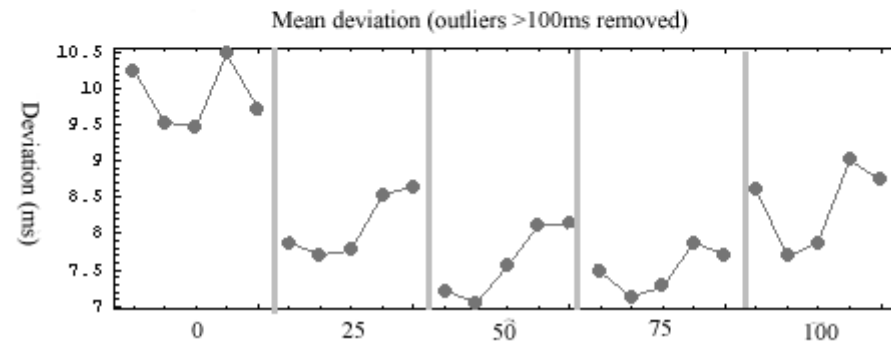
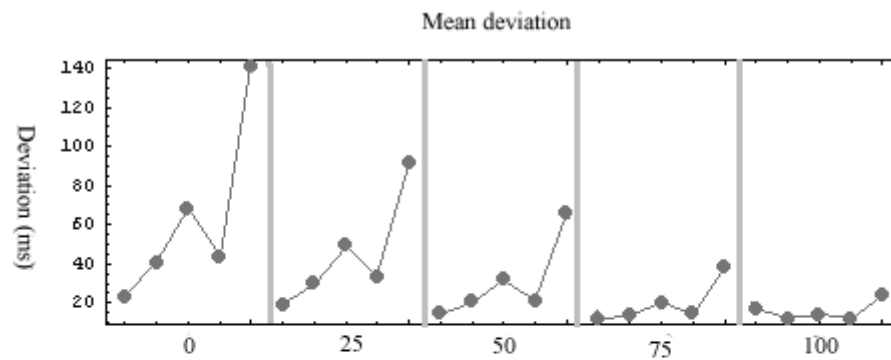


# Parameters

---

- Search window parameters
- Wavelet parameter -  $k$
- Wavelet parameter -  $LPF$
- Wavelet tuning
- Feedback analysis

# Search window parameters



a = [-0.33 IOI : 0.33 IOI]  
 b = [-0.50 IOI : 0.50 IOI]  
 c = [-0.67 IOI : 0.67 IOI]

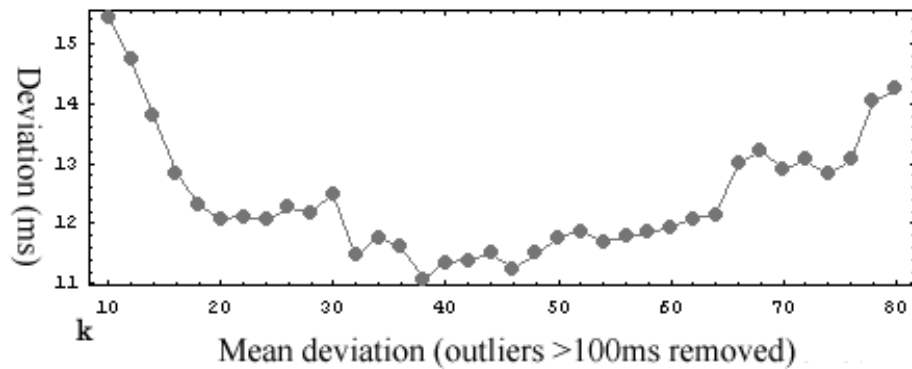
d = [-0.33 IOI : 0.67 IOI]  
 e = [-1.00 IOI : 1.00 IOI]



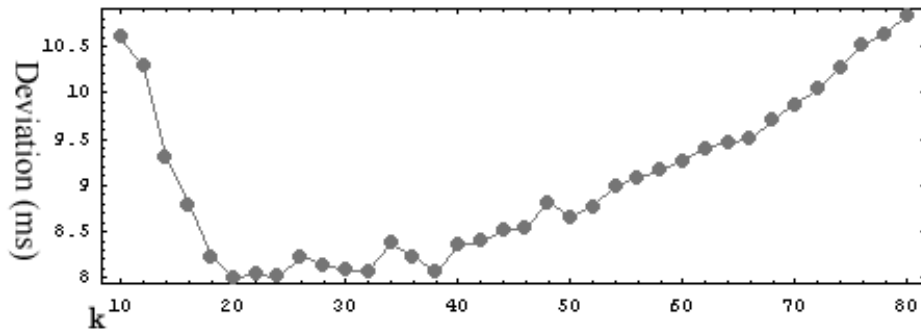
# Wavelet parameters - $k$

Chiu

Mean deviation

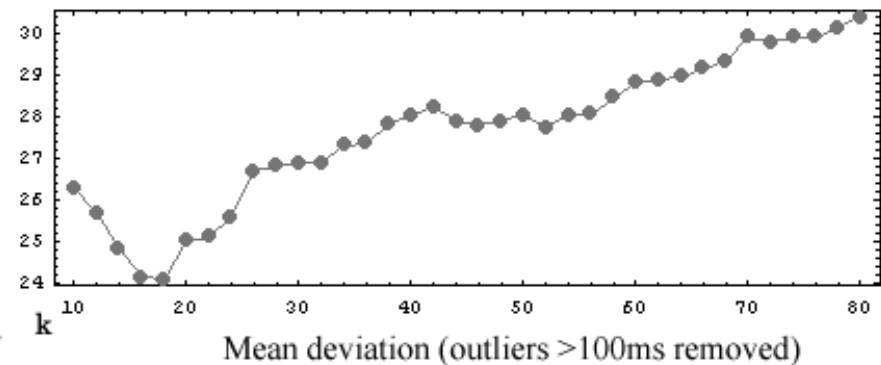


Mean deviation (outliers >100ms removed)

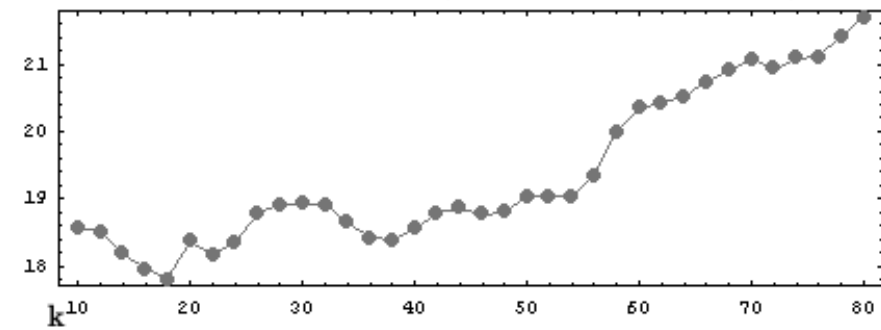


Friedman

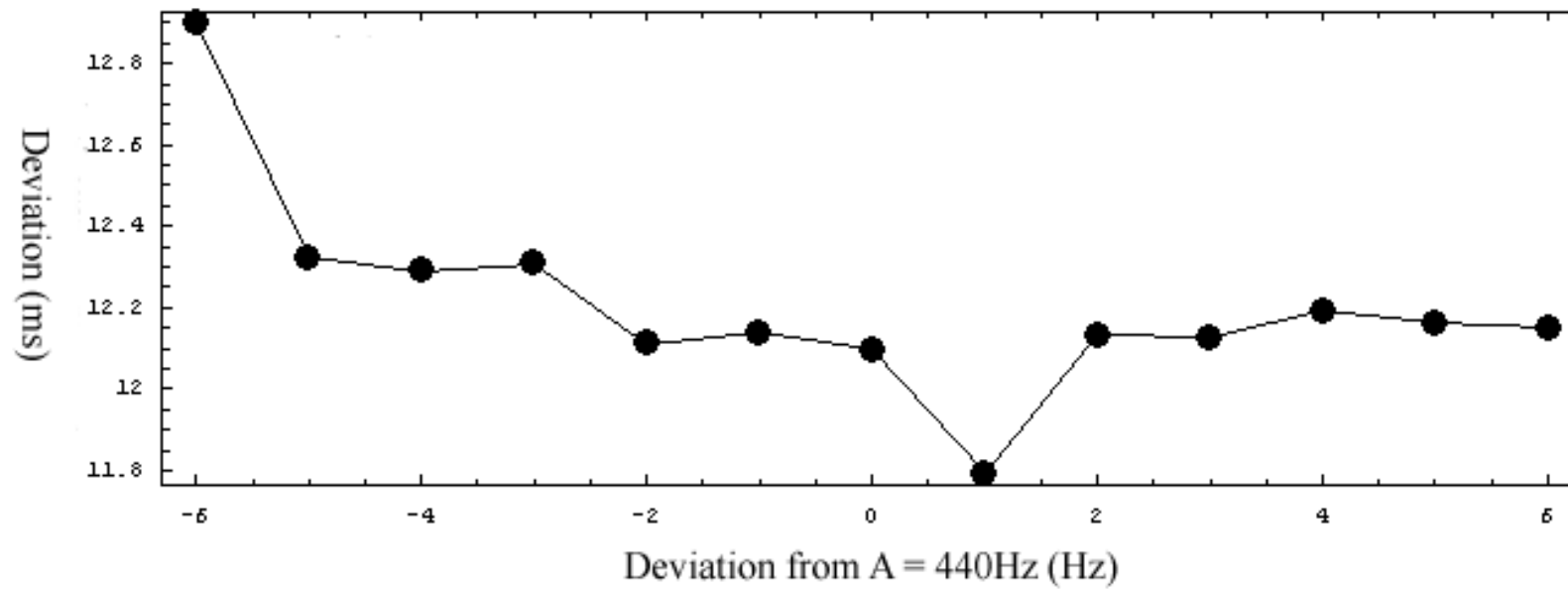
Mean deviation



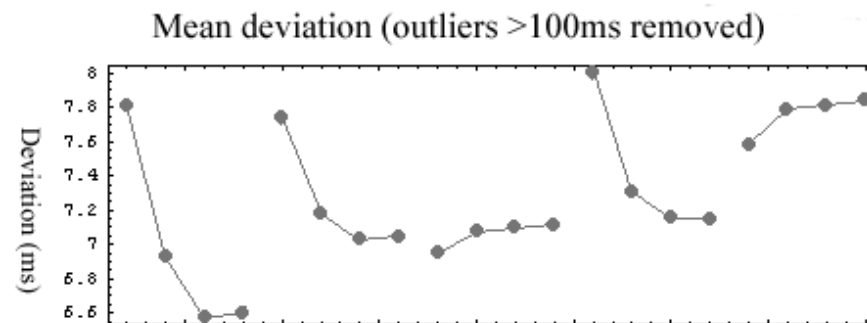
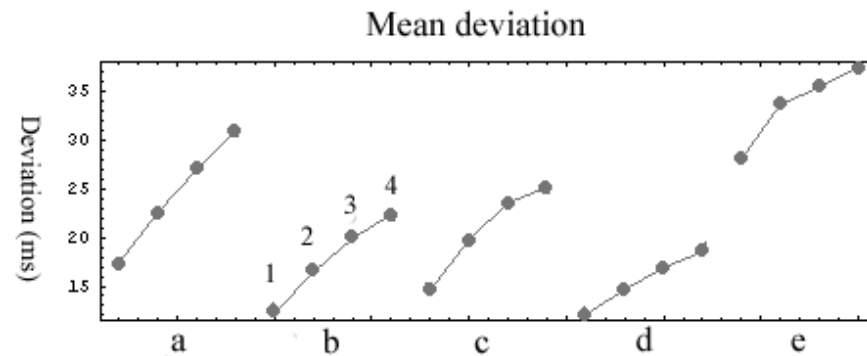
Mean deviation (outliers >100ms removed)



# Wavelet tuning



# Feedback analysis



a = [-0.33 IOI : 0.33 IOI]

b = [-0.50 IOI : 0.50 IOI]

c = [-0.67 IOI : 0.67 IOI]

d = [-0.33 IOI : 0.67 IOI]

e = [-1.00 IOI : 1.00 IOI]



## STEP 5

---

- Manual correction of main beat onset times
  - Sonic Visualiser
- Output – table of corrected main beat onset times



## STEP 6

---

- Estimation of subbeat onset times from main beat onset times
- Correction of estimated subbeat onset times using algorithm
- Estimation of main beat and subbeat dynamics
- Estimation of individual note onset times



# End product

---

- MATLAB code converted to standard .exe files
- Input data required
  - Excel control spreadsheet - for batch processing of data
    - Input/output file names
    - Analysis parameters
  - 'Main beat' beat tapped input data spreadsheet
  - .wav sound recording



# End product

---

- Three different algorithms
  - mainbeatcorrection.exe
    - correction of 'main beat' tapped onset times
    - measurement of main beat dynamics (in dB)
  - subbeatmeasurement.exe
    - measurement of sub beat onset times
    - measurement of sub beat dynamics
  - individualnotemeasurement.exe
    - measurement of individual note onset timings and dynamics
- Will be available (or a variation of this) on CHARM website as downloadable resource



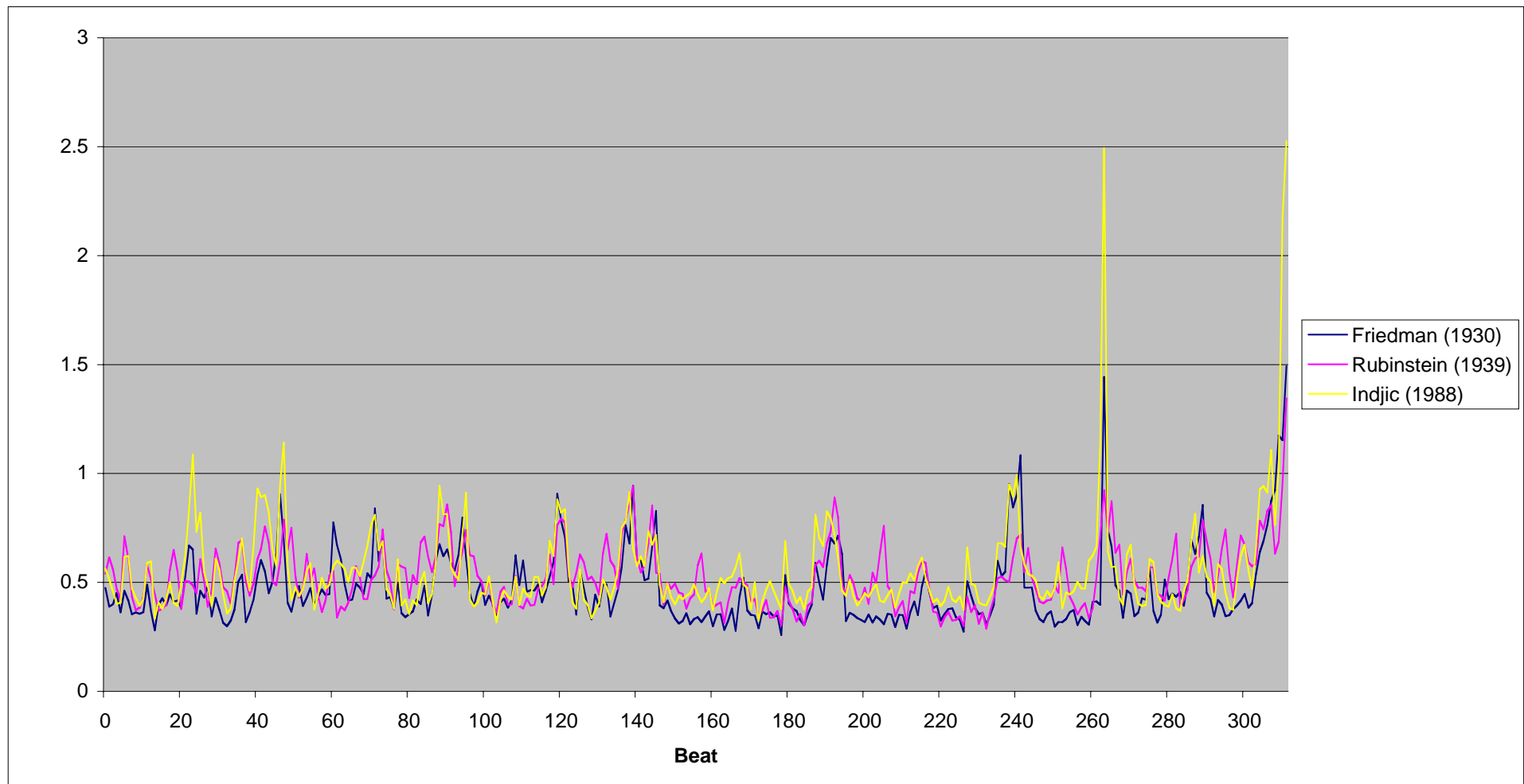
# End product

---

- For example...
  - 3 recordings of Chopin's Mazurka in A minor op 7 no 2
    - Friedman (1930)
    - Rubinstein (1939)
    - Indjic (1988)
  - Graphs of
    - inter-onset interval for each event in score
    - dynamics of each event (scaled in dBs)

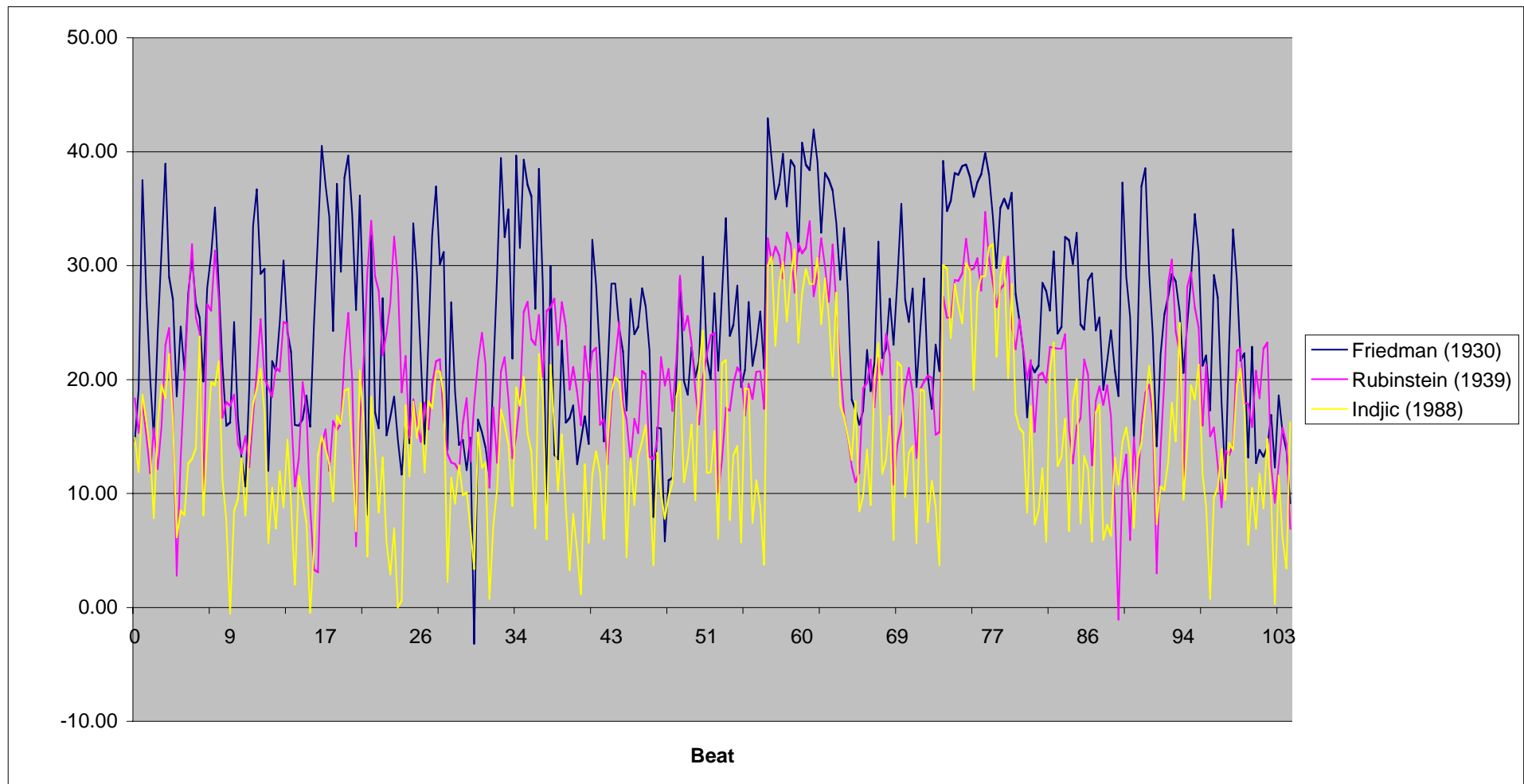


# Inter-onset interval





# Event dynamics





## Conclusions and future work

---

- Further scientific development (wavelet parameters, shape, dynamics measurement)
- Extension to note offset times
- Testing and refinement of user interface