

Module Interface Specification for Audio360

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1 Revision History

Date	Version	Notes
2025-11-13	1.0	Initial write-up

2 Symbols, Abbreviations and Acronyms

symbol	description
Audio360	360 Audio analysis system on smart glasses
FR	Functional Requirement
M	Module
MG	Module Guide
MIS	Module Interface Specification
NFR	Non-functional Requirement
R	Requirement
SPI	Serial Peripheral Interface
SRS	Software Requirements Specification
USART	Universal Synchronous/Asynchronous Receiver/Transmitter

Table 1: Symbols, abbreviations and acronyms used in the MIS document.

See SRS Documentation at Symbols, Abbreviations, and Acronyms for a complete table used in Audio360.

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3 Introduction

The following document details the Module Interface Specifications for [Fill in your project name and description —SS]

Complementary documents include the System Requirement Specifications and Module Guide. The full documentation and implementation can be found at [provide the url for your repo —SS]

4 Notation

[You should describe your notation. You can use what is below as a starting point. —SS]

The structure of the MIS for modules comes from [1], with the addition that template modules have been adapted from [2]. The mathematical notation comes from Chapter 3 of [1]. For instance, the symbol $::=$ is used for a multiple assignment statement and conditional rules follow the form $(c_1 \Rightarrow r_1 | c_2 \Rightarrow r_2 | \dots | c_n \Rightarrow r_n)$.

The following table summarizes the primitive data types used by Audio360.

Data Type	Notation	Description
character	char	a single symbol or digit
integer	\mathbb{Z}	a number without a fractional component in $(-\infty, \infty)$
natural number	\mathbb{N}	a number without a fractional component in $[1, \infty)$
real	\mathbb{R}	any number in $(-\infty, \infty)$

The specification of Audio360 uses some derived data types: sequences, strings, and tuples. Sequences are lists filled with elements of the same data type. Strings are sequences of characters. Tuples contain a list of values, potentially of different types. In addition, Audio360 uses functions, which are defined by the data types of their inputs and outputs. Local functions are described by giving their type signature followed by their specification.

5 Module Decomposition

The following table is taken directly from the Module Guide document for this project.

Level 1	Level 2
Hardware-Hiding	
	Input Parameters
	Output Format
	Output Verification
Behaviour-Hiding	Temperature ODEs
	Energy Equations
	Control Module
	Specification Parameters Module
Software Decision	Sequence Data Structure
	ODE Solver
	Plotting

Table 2: Module Hierarchy

6 Data Types

6.1 Generics

6.1.1 int32

32 bits signed integer (\mathbb{Z}).

6.1.2 int64

64 bits signed integer (\mathbb{Z}).

6.1.3 uint32

32 bits unsigned integer (\mathbb{N}).

6.1.4 uint64

64 bits unsigned integer (\mathbb{N}).

6.1.5 float32

32 bits floating point (\mathbb{R}).

6.1.6 float64

64 bits floating point (\mathbb{R}).

6.2 Enums

6.2.1 Audio360State

1. AudioClassificationProcess: State when audio classification is running.
2. DirectionalAnalysisProcess: State when directional analysis is running.
3. OutputProcess: State when output processing is running.

6.2.2 Audio360Status

1. Uninitialized: Audio360 Engine is not initialized.
2. Initialized: Audio360 Engine is initialized, but not ready.
3. Ready: Audio360 Engine ready for requests.
4. Running: Audio360 Engine is running. Can not accept new requests.
5. Error: Audio360 Engine is stuck at an unhandled error.

6.3 Data Structures

7 MIS of [Module Name —SS]

[Use labels for cross-referencing —SS]

[You can reference SRS labels, such as R. —SS]

[It is also possible to use L^AT_EX for hyperlinks to external documents. —SS]

7.1 Module

[Short name for the module —SS]

7.2 Uses

7.3 Syntax

7.3.1 Exported Constants

7.3.2 Exported Access Programs

Name	In	Out	Exceptions
[accessProg —SS]	-	-	-

7.4 Semantics

7.4.1 State Variables

[Not all modules will have state variables. State variables give the module a memory. —SS]

7.4.2 Environment Variables

[This section is not necessary for all modules. Its purpose is to capture when the module has external interaction with the environment, such as for a device driver, screen interface, keyboard, file, etc. —SS]

7.4.3 Assumptions

[Try to minimize assumptions and anticipate programmer errors via exceptions, but for practical purposes assumptions are sometimes appropriate. —SS]

7.4.4 Access Routine Semantics

[accessProg —SS]():

- transition: [if appropriate —SS]
- output: [if appropriate —SS]
- exception: [if appropriate —SS]

[A module without environment variables or state variables is unlikely to have a state transition. In this case a state transition can only occur if the module is changing the state of another module. —SS]

[Modules rarely have both a transition and an output. In most cases you will have one or the other. —SS]

7.4.5 Local Functions

[As appropriate —SS] [These functions are for the purpose of specification. They are not necessarily something that is going to be implemented explicitly. Even if they are implemented, they are not exported; they only have local scope. —SS]

8 MIS of Audio360 Engine - M??

8.1 Module

Orchestrates the overall audio processing by receiving raw input data and managing [module](#) communication.

8.2 Uses

1. Audio classification
2. Directional Analysis
3. Output
4. Audio Sampling Module

8.3 Syntax

8.3.1 Exported Constants

None

8.3.2 Exported Access Programs

Name	In	Out	Exceptions
runProgram	-	-	ProgramStartFailure, ProgramRunTimeFailure

8.4 Semantics

8.4.1 State Variables

- state [[Audio360State](#)]: Determines the current state of the the Audio360 engine. Each state will run a specific module in [used modules](#).

8.4.2 Environment Variables

- status [[Audio360Status](#)]: Status of the module. This encapsulates initialization, running, ready, or errored.

8.4.3 Assumptions

None

8.4.4 Access Routine Semantics

run():

- transition: The state machine in figure 1 outlines the transition of this module.

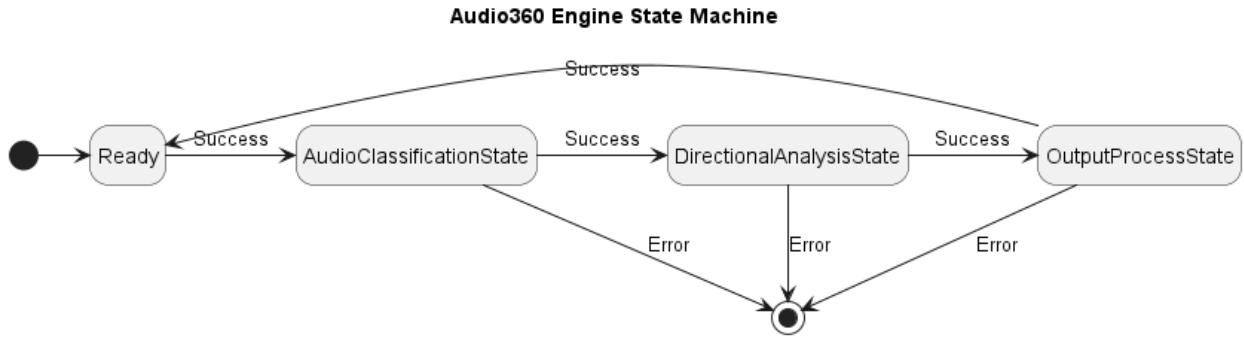


Figure 1: Internal state machine of Audio360 Engine module.

- output: None
- exception: ProgramStartFailure, ProgramRunTimeFailure

8.4.5 Local Functions

None

9 MIS of Mel Filter Module - M??

9.1 Module

Converts a Spectrogram into the equivalent Mel-Spectrogram by applying a bank of n mel-scaled triangular filters to each frequency frame.

9.2 Uses

1. Audio Classification

9.3 Syntax

9.3.1 Exported Constants

None

9.3.2 Exported Access Programs

Name	In	Out	Exceptions
applyMelFilterBank	spectrogram: real array	2D	mel-spectrogram: 2D real array

9.4 Semantics

9.4.1 State Variables

None

9.4.2 Environment Variables

None

9.4.3 Assumptions

- The input spectrogram is the output of the FFT Module across a constant buffer time. This is arranged as a 2D matrix where the rows represent time frames and columns represent frequency bins. This matrix can be represented as $S(t, k)$ where k is the number of frequency bins, and t is the number of time frames.
- The mel filterbank matrix is precomputed offline and stored as a read only constant within the system. It is computed offline using pre-determined information like Sampling Rate, FFT Size, maximum and minimum frequencies and number of mel filters. This matrix can be represented as $H(m, k)$, where m is the number of mel-filters, k is the number of frequency bins and $H(m, k)$ is a weight that tells you how much frequency bin k contributes to mel band m .

9.4.4 Access Routine Semantics

applyMelFilterBank():

- **transition:** None (function doesn't have states)
- **output:** Returns the equivalent mel-spectrogram, where each time frame is converted from linear-frequency bin representation to mel-scaled frequency bins. This is done by:
 1. Multiplying each spectrogram frame by the mel filterbank matrix.
 2. Summing weighted energy contributions per mel filter.

$$E(t, m) = \sum_k S(t, k) \cdot H(m, k)$$

- **exception:** Invalid spectrogram dimension. If the dimension of the input spectrogram is not $S(t, k)$ and perhaps something like $S(t, p)$ where $p \neq k$, then the matrix multiplication would throw an invalid dimension error, which must be exposed from the module as invalid spectrogram dimension as well. This would only happens if the number of frequency bins in the computted FFT is not equal to the number of frequency bins in the pre-computed mel filterbank matrix.

9.4.5 Local Functions

None

10 MIS of Discrete Cosine Transform Module - M??

10.1 Module

Converts a mel-spectrogram into a 2D array containing the sequence of Mel Frequency Cepstral Coefficients (MFCC) vectors by applying the Discrete Cosine Transform (DCT) to each time frame. This reduces overlap between frequency bands that carry similar information and ensures each coefficient is uncorrelated to the other one. Pre-processing step that makes it easier to perform Principle Component Analysis later on.

10.2 Uses

1. Audio Classification

10.3 Syntax

10.3.1 Exported Constants

None

10.3.2 Exported Access Programs

Name	In	Out	Exceptions
applyDCT	mel-spectrogram: real array	2D mfcc: 2D real array	invalidDimensions

10.4 Semantics

10.4.1 State Variables

None

10.4.2 Environment Variables

None

10.4.3 Assumptions

- The input mel-spectrogram is the output of the Mel Filter Module. This input will be of size $E(t, m)$, where t is the number of time frames, and m is the number of mel-frequency bins.
- The DCT transform matrix $D(c, m)$ will be computed offline, where c is the desired number of MFCC coefficients for each time frame, and m is the number of mel-frequency bands. $D(c, m)$ represents the unique cosine wave that is associated with the

c^{th} coefficient and m^{th} band. A combination of these waves will be used to uniquely represent the spectrogram at time frame t .

$$D(c, m) = \cos\left(\frac{\pi c}{M}(m - 0.5)\right)$$

- Only 13 MFCC coefficients will be calculated for each time frame t .

10.4.4 Access Routine Semantics

applyDCT():

- **transition:** None (function doesn't have states)
- **output:** Returns a matrix $mfcc(t, c)$, such that for each time frame t and each MFCC coefficient index c , the value of the matrix at $mfcc(t, c)$ can be computed as

$$mfcc(t, c) = \sum_{m=1}^M E(t, m) \cdot DCT(c, m)$$

Where $E(t, m)$ represents the mel-spectrogram at time frame t and mel-frequency m . And $DCT(c, m)$ represents the DCT transform matrix that was computed offline.

- **exception:** Invalid spectrogram dimension. If the dimension of the input spectrogram is not $E(c, m)$ and perhaps something like $E(c, p)$ where $m \neq p$ then the matrix multiplication would throw an invalid dimension error, which must be exposed from the module as invalid spectrogram dimension as well. This would only happens if for some reason the number of mel-frequency bands returned by the Mel Filter Module is not the same as the number of mel-frequency bands in the DCT matrix computation.

10.4.5 Local Functions

None

11 MIS of Principle Component Analysis Module - M??

11.1 Module

Principle Component Analysis (PCA) highlights the features with the greatest variance from an input feature matrix. In this case, uses the MFCC feature vector for each time frame to find the direction of greatest variance. This helps with eliminating noise and unimportant information.

11.2 Uses

1. Audio Classification

11.3 Syntax

11.3.1 Exported Constants

None

11.3.2 Exported Access Programs

Name	In	Out	Exceptions
applyPCA	mfcc: 2D real array	pcaFeatures: 2D real array	invalidDimensions

11.4 Semantics

11.4.1 State Variables

None

11.4.2 Environment Variables

None

11.4.3 Assumptions

- The input MFCC matrix is the output of the Discrete Cosine Transform Module. This input will be of size $mfcc(t, c)$, where t is the number of time frames, and c is the number of coefficients.
- The PCA mean vector and PCA projection matrix will be derived offline during the system training and stored as read-only constants. The PCA mean vector will be of size

$(1, c)$, and will be computed as the average MFCC vectors across all training samples.

$$\bar{x} = \frac{1}{T} \sum_{t=1}^T x(t)$$

The PCA projection matrix will be of size (K, c) , where K is the number of eigenvectors with the maximum variance and c is the number of MFCC coefficients. These eigenvectors are extracted from the following covariance matrix.

$$C = \frac{1}{T} \sum_{t=1}^T (x(t) - \bar{x})(x(t) - \bar{x})^T$$

11.4.4 Access Routine Semantics

applyPCA():

- **transition:** None (function doesn't have states)
- **output:** Returns a matrix $pca(t, K)$, where t represents the number of time frames and K represents the number of PCA components retained (proportional to the number of eigenvectors). Essentially, each input mfcc matrix is projected onto the vectors of maximum variance, therefore extracting only the most important features of the MFCC based on training. This projection can be computed using the following matrix multiplication.

$$pca = (mfcc - \mathbf{1}\bar{x}) P^T$$

Where P represents the projection matrix, \bar{x} represents the PCA mean vector, and $mfcc$ represents the full mfcc matrix.

- **exception:** Invalid mfcc dimension. If the dimension of the input mfcc matrix is not (t, c) and perhaps something like (t, p) where $c \neq p$ then the matrix multiplication would throw an invalid dimension error, which must be exposed from the module as invalid mfcc dimension as well. This would only happen if for some reason the number of coefficients returned by the Discrete Cosine Transform is not the same as the number of coefficients in the projection matrix.

11.4.5 Local Functions

None

12 MIS of Linear Discriminant Analysis Module - M??

12.1 Module

Classifies given feature vector into one of the predefined set of sound classes. This is module is optimized for class separation. The feature vector in this case is already narrowed down to the features with the most variance using the PCA module.

12.2 Uses

1. Audio Classification

12.3 Syntax

12.3.1 Exported Constants

None

12.3.2 Exported Access Programs

Name	In	Out	Exceptions
applyLDA	pcaFeatures: 2D real array	classLabels: 1D real array	invalidDimensions

12.4 Semantics

12.4.1 State Variables

None

12.4.2 Environment Variables

None

12.4.3 Assumptions

- The input pcaFeatures is the output of the Principle Component Analysis Module. This input will be of size (t, K) , where t is the number of time frames, and K is number of features selected with the highest variances from PCA.
- The LDA projection matrix, and class weight parameters are computed offline using labelled training data. They are stored as read-only constants in the system. The LDA projection matrix is of size $(K, C - 1)$, where C is the number of classes, and K is the number of variance eigenvectors. In essence, this matrix maps the input feature space into the discriminant subspace where class separation is maximized.

12.4.4 Access Routine Semantics

applyLDA():

- **transition:** None (function doesn't have states)

- **output:**

Returns a vector of class labels, one per time frame.

Each feature vector from PCA is projected onto the discriminant space using the LDA projection matrix using the following multiplication.

$$z(t, :) = pca(t, :) \cdot W_{\text{LDA}}$$

Where $pca(t, :)$ represents the PCA eigenvectors at time t , and W_{LDA} represents the LDA projection matrix.

Classification is then performed using linear discriminant functions for each class j , where w_j and b_j is the weight and bias matrices that were pre-computed offline during training for class j .

$$g_j(t) = w_j^T z(t, :) + b_j$$

The predicted class for that time frame is then assigned as the class with the maximum value.

$$\text{classLabel}(t) = \arg \max_j g_j(t)$$

The output of the module is then a vector of length t , representing the best classification for all time frames.

- **exception:** Invalid feature space dimension. If the dimension of the input feature matrix is not (t, K) and perhaps something like (t, p) where $K \neq p$ then the matrix multiplication would throw an invalid dimension error, which must be exposed from the module as invalid feature dimension as well. This would only happens if for some reason the number of eigenvectors returned by the PCA module is not the same as the number of eigenvectors in the precomputed LDA matrix.

12.4.5 Local Functions

None

13 MIS of Classification Module - M??

13.1 Module

Orchestrates the overall audio classification flow by receiving signals in the frequency domain and encapsulating the complexity of the various modules involved.

13.2 Uses

1. Audio Classification

13.3 Syntax

13.3.1 Exported Constants

None

13.3.2 Exported Access Programs

Name	In	Out	Exceptions
classify	spectrogram: 2D real array	-	ClassificationStartFailure, ClassificationRunTimeFailure

13.4 Semantics

13.4.1 State Variables

- state[classificationState]: Stores the current classification. Will continuously be updated at runtime after receiving continuous audio signals.
- state[classificationStatus]: Stores the confidence of the classification that was done. Values for status include high, medium or low confidence.

13.4.2 Environment Variables

- state[status]: Encapsulates initialization, running, ready or errored status of the module.

13.4.3 Assumptions

None.

13.4.4 Access Routine Semantics

applyLDA():

- **transition:** State machine below outlines the transition of this module.

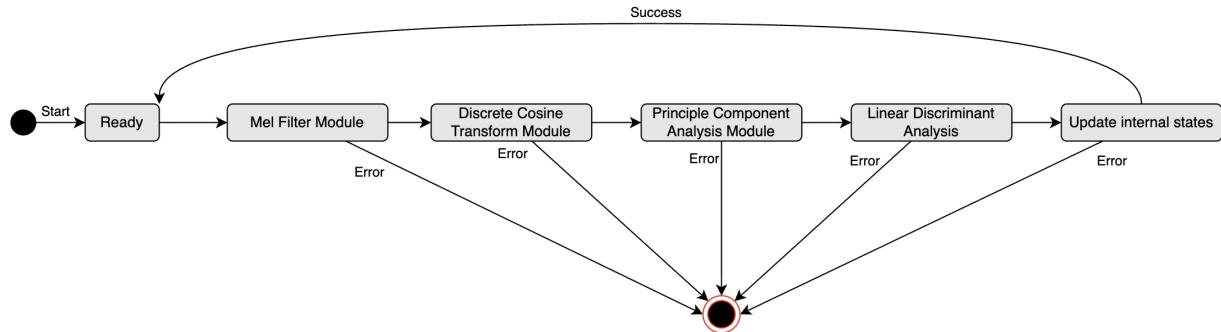


Figure 2: Internal state machine of Classification module.

- **output:** None
- **exception:** ClassificationStartFailure, ClassificationRunTimeFailure

13.4.5 Local Functions

None

References

- [1] D. M. Hoffman and P. A. Strooper, *Software Design, Automated Testing, and Maintenance: A Practical Approach*. New York, NY, USA: International Thomson Computer Press, 1995. [Online]. Available: <http://citeseer.ist.psu.edu/428727.html>
- [2] C. Ghezzi, M. Jazayeri, and D. Mandrioli, *Fundamentals of Software Engineering*, 2nd ed. Upper Saddle River, NJ, USA: Prentice Hall, 2003.

14 Appendix

[Extra information if required —SS]

Appendix — Reflection

1. What went well while writing this deliverable?

Sathurshan: The system was decomposed into modules that was small enough. This allowed the team to easily split up the work without having much dependencies on each other. Furthermore, writing the MIS has helped the team further align with design decisions and formally document.

2. What pain points did you experience during this deliverable, and how did you resolve them?

Sathurshan: The main pain point was the deliverable deadline as the team was asked to finish the first revision in a week while preparing for the proof of concept of demo. The team expressed the concerns with the professor, and fortunately received an extension allowing us to put more effort into this document.

3. Which of your design decisions stemmed from speaking to your client(s) or a proxy (e.g. your peers, stakeholders, potential users)? For those that were not, why, and where did they come from?

Sathurshan: The supervisor mentioned the difficulty of getting access to hardware that will meet our software specification. As a result, we designed the system such that the software is portable and can take input from different sources. This ensures that the capstone project can still be successful in the case we are not able to find the right hardware within our budget.

4. While creating the design doc, what parts of your other documents (e.g. requirements, hazard analysis, etc), if any, needed to be changed, and why?

Sathurshan: SRS needed to be updated. It was updated because part of writing this document required reviewing the SRS. From this, I have found parts of the requirements that can be improved based on recent better understanding of the system.

5. What are the limitations of your solution? Put another way, given unlimited resources, what could you do to make the project better? (LO_ProbSolutions)

Sathurshan: The limitation of the solution is compute and memory power. Given further time, the team would be able to design a system that is more low level and is optimized in accomplishing the tasks that are required.

6. Give a brief overview of other design solutions you considered. What are the benefits and tradeoffs of those other designs compared with the chosen design? From all the potential options, why did you select the documented design? (LO_Explores)

Sathurshan: Personally, there was not a lot of time to think about other designs. The team has been implementing the software before this document was created since the proof of concept is one week after this document is due. Thus, the team already

considered and analyzed high level designs months ago and is too far back to document them.