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.

Contents

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 $(c_1 \Rightarrow r_1 | c_2 \Rightarrow r_2 | ... | c_n \Rightarrow r_n)$.

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 $(-\infty, \infty)$
array/list	x[]	Ordered list of type x.
negate	$\neg$	Logical math NOT.
and	$\wedge$	Logical math AND
or	$\vee$	Logical math OR
implies	$\implies$	Logical math implies
assignment	:=	For A := B. B is assigned to A.
for all	$\forall$	Referencing all items. For example: $(\forall\ variable\ |\ condition:\ statement)$

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	
Behaviour-Hiding	Input Parameters Output Format Output Verification 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 int16

16 bits signed integer ($\mathbb{Z}$).

6.1.2 int32

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

6.1.3 int64

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

6.1.4 uint16

16 bits unsigned integer ($\mathbb{N}$).

6.1.5 uint32

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

6.1.6 uint64

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

6.1.7 float32

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

6.1.8 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.

subsubsectionAudio360Status

1. NoFault: No fault state.

2. MicrophoneXFault: Fault with microphone X.

3. AudioClassificationFault: Fault with audio classification.

4. directionalAnalsysiFault: Fault with the directional analysis.

6.3 Data Structures

6.3.1 FrequencyDomain

1. N [uint16]: The number of data points.

2. frequency [float32[]]: The frequency represented in Hz.

3. real [float32[]]: The real component of thre frequency contribution.

4. img [float32[]]: The imaginary component of thre frequency contribution.

5. magnitude [float32[]]: The magnitude of the frequency component.

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 LaTeX for hypperlinks 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. Classification Module

2. DOA Processor Module

3. Fault Manager Module

4.

5. Logging 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.

Audio360 Engine State Machine
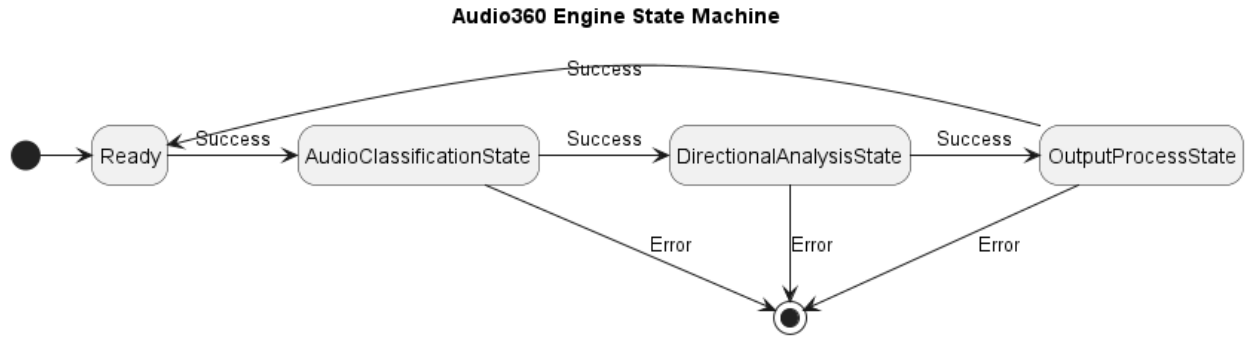
Figure 1: Internal state machine of Audio360 Engine module.

- output: None

- exception: ProgramStartFailure, ProgramRunTimeFailure

8.4.5 Local Functions

None

9 MIS of Audio Sampling Module - M??

9.1 Module

Provides sampling of audio data from a given source while preserving the order of individual samples.

9.2 Uses

- Microphone Input Module

- Microphone Diagnostic Module

9.3 Syntax

9.3.1 Exported Constants

- sampleWindowSize [uint16]: The number of samples in a window. The value is a power of 2.

9.3.2 Exported Access Programs

Name	In	Out	Exceptions
sample	inputSource	-	InputSouceError
getSampledData	-	sampledData	-

9.4 Semantics

9.4.1 State Variables

- sampledData [uint32[]]: A cyclic array of size sampleWindowSize that stored the sampled data.

- windowPosition [uint32]: The current reference index position of the array storing sampled data. This denotes the starting point of the cyclic array.

9.4.2 Environment Variables

None

9.4.3 Assumptions

None

9.4.4 Access Routine Semantics

sample():

- transition:

  1. $sampleData[windowPosition] := inputSource.newData$
  2. $((windowPosition + 1) \geq sampleWindowSize) \implies windowPosition := 0 \lor \neg((windowPosition + 1) \geq sampleWindowSize) \implies windowPosition := windowPosition + 1$

- output: None

- exception: InputSouceError

getSampledData():

- transition: None

- output: sampleData

- exception: None

9.4.5 Local Functions

None

10 MIS of Audio Spectral Leakage Prevention Module - M??

10.1 Module

Applying windowing on audio signal to reduce spectral leakage before frequency domain processing.

10.2 Uses

None

10.3 Syntax

10.3.1 Exported Constants

None

10.3.2 Exported Access Programs

Name	In	Out	Exceptions
applyFilter	sampleWindow	filteredWindow	-

10.4 Semantics

10.4.1 State Variables

None

10.4.2 Environment Variables

None

10.4.3 Assumptions

None

10.4.4 Access Routine Semantics

applyFilter():

- transition: None

- output:

1. $(\forall\ i\ |\ 0 \leq i \leq N - 1 :\ filteredWindow[i] := sampleWindow[i] * \sin(\frac{\pi * i}{N - 1})^2)$
   such that $N = $ size of sampleWindow.

- exception: None

10.4.5   Local Functions

None

11   MIS of Audio Normalizer Module - M??

11.1   Module

Processes audio signals to maintain consistent amplitude across different sources.

11.2   Uses

None

11.3   Syntax

11.3.1   Exported Constants

None

11.3.2   Exported Access Programs

Name	In	Out	Exceptions
normalize	sampleWindow, source	filteredWindow	-

11.4   Semantics

11.4.1   State Variables

None

11.4.2   Environment Variables

None

11.4.3   Assumptions

Normalization factor is pre-deteremined and known for all given sources.

11.4.4   Access Routine Semantics

normalize():

- transition: None

- output:

  1. $(\forall\ i\ |\ 0 \le i \le N - 1 :\ filteredWindow[i] := \frac{sampleWindow[i]}{F_{source}})$
     such that N = size of sampleWindow and F_{source} is normalization factor of a the sampleWindow source.

- exception: None

11.4.5   Local Functions

None

12 MIS of Fast Fourier Transform - M??

12.1 Module

Computes the discrete Fourier transform of the input signal to obtain its frequency domain

12.2 Uses

- Audio Spectral Leakage Prevention Module

12.3 Syntax

12.3.1 Exported Constants

None

12.3.2 Exported Access Programs

Name	In	Out	Exceptions
signalToFrequency	signal	frequency	-

12.4 Semantics

12.4.1 State Variables

None

12.4.2 Environment Variables

- inputsize [uint32]: The size of the signal input, as in number of samples. This is required to interface with hardware acceleration methods on the microcontroller.

12.4.3 Assumptions

None

12.4.4 Access Routine Semantics

signalToFrequency():

- transition: None

- output:

  1. $(\forall\, k \mid 0 \leq k \leq N - 1 :\ frequency[k] := \sum_{n=0}^{N-1} x_n * e^{-2i\pi n \frac{k}{N}})$
     such that N = size of input signal.

- exception: None

12.4.5 Local Functions

- createOutput(): returns a FrequencyDomain. This function extract features from the FFT and stores it in the data structure.

13 MIS of Logging Module - M??

13.1 Module

Provides centralized logging of data streams to designated output destinations for debugging and monitoring.

13.2 Uses

- USART Communication Module

- SD Card Interface Module

13.3 Syntax

13.3.1 Exported Constants

None

13.3.2 Exported Access Programs

Name	In	Out	Exceptions
log	text	-	-

13.4 Semantics

13.4.1 State Variables

None

13.4.2 Environment Variables

None

13.4.3 Assumptions

None

13.4.4 Access Routine Semantics

log():

- transition: None

- output: input text is logged to output source (SD card or USART port)

- exception: None

13.4.5 Local Functions

None

14 MIS of Mel Filter Module - M??

14.1 Module

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

14.2 Uses

None

14.3 Syntax

14.3.1 Exported Constants

None

14.3.2 Exported Access Programs

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

14.4 Semantics

14.4.1 State Variables

None

14.4.2 Environment Variables

None

14.4.3 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.

14.4.4 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.

14.4.5 Local Functions

None

15 MIS of Discrete Cosine Transform Module - M??

15.1 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.

15.2 Uses

None

15.3 Syntax

15.3.1 Exported Constants

None

15.3.2 Exported Access Programs

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

15.4 Semantics

15.4.1 State Variables

None

15.4.2 Environment Variables

None

15.4.3 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.

15.4.4 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.

15.4.5 Local Functions

None

16 MIS of Principle Component Analysis Module - M??

16.1 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.

16.2 Uses

Non

16.3 Syntax

16.3.1 Exported Constants

None

16.3.2 Exported Access Programs

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

16.4 Semantics

16.4.1 State Variables

None

16.4.2 Environment Variables

None

16.4.3 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.

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

16.4.4 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 eignevectors). 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.

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

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 happens 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.

16.4.5 Local Functions

None

17 MIS of Linear Discriminant Analysis Module - M??

17.1 Module

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

17.2 Uses

None

17.3 Syntax

17.3.1 Exported Constants

None

17.3.2 Exported Access Programs

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

17.4 Semantics

17.4.1 State Variables

None

17.4.2 Environment Variables

None

17.4.3 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 seperation is maximized.

17.4.4 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 matricies 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 eignvectors in the precomputed LDA matrix.

17.4.5 Local Functions

None

18 MIS of Classification Module - M??

18.1 Module

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

18.2 Uses

1. Audio Sampling Module

2. Audio Normalizer Module

3. Fast Fourier Transform Module

4. Mel Filter Module

5. Discrete Cosine Transform Module

6. Principle Component Analysis Module

7. Linear Discrete Analysis Module

8.

18.3 Syntax

18.3.1 Exported Constants

None

18.3.2 Exported Access Programs

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

18.4 Semantics

18.4.1 State Variables

- state[classificationState]: Stores the current classification. Will continuously be updated at runtime after recieving continuous audio signals.

- state[classificationStatus]: Stores the confidence of the classification that was done. Values for status include high, medium or low confidence.

18.4.2   Environment Variables

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

18.4.3   Assumptions

None.

18.4.4   Access Routine Semantics

applyLDA():

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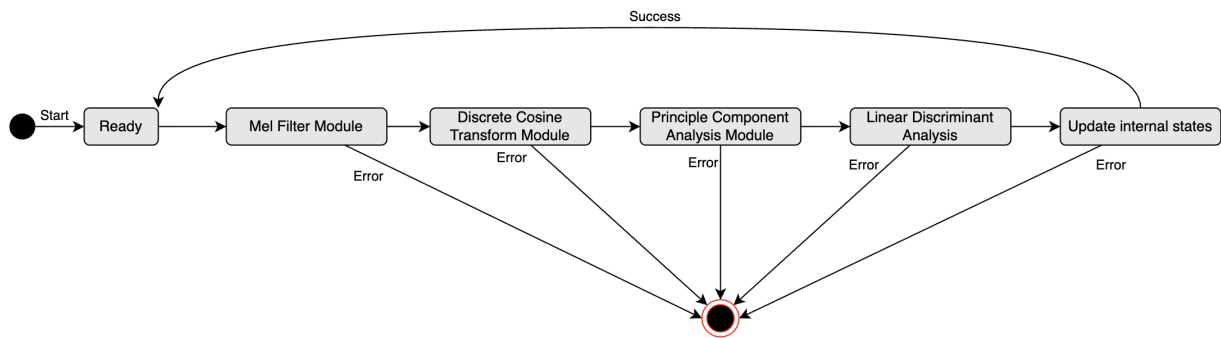

Figure 2: Internal state machine of Classification module.

- **output:** None

- **exception:** ClassificationStartFailure, ClassificationRunTimeFailure

18.4.5   Local Functions

None

19 MIS of Fault Manager Module - M??

19.1 Module

Monitors system health and manages critical faults to maintain core functionality.

19.2 Uses

1. Microphone Diagnostic Module

2. Classification Module

3. DOA Processor Module

4. Audio Anomaly Detection Module

19.3 Syntax

19.3.1 Exported Constants

None

19.3.2 Exported Access Programs

Name	In	Out	Exceptions
runFaultAnalysis	-	-	-

19.4 Semantics

19.4.1 State Variables

- faultState [FaultState]: The current fault state of the overall software system.

19.4.2 Environment Variables

- faultState: This variable is shared with the display, to notifiy user of the internal status of the software system.

19.4.3 Assumptions

Fault Manager module itself does not run into any critical faults.

19.4.4 Access Routine Semantics

faultState():

- transition: update the value of faultState depending on the overal health of the software system.

- output: None

- exception: None

19.4.5 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.

20 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 fornately 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 mentionned 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), it 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.