

Department of Information Engineering and Computer Science

Bachelor’s Degree in

COMPUTER SCIENCE

final dissertation

a web graphical user interface for the

packet-loss-concealment testbench tool

|  |  |
| --- | --- |
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Academic year 22/23

**Acknowledgements**

*To* ***my wife*** *and to* ***my daughters***

*who constantly supported me during this journey*

*Thanks to* ***Luca Vignati***

*who provided guidance and precious feedback*

*during the entire project development*

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**Abstract**

*The purpose of this thesis was implementing a graphical user interface for the PLC (Packet Loss Concealment) Testbench Tool developed by the Ph.D student Luca Vignati, thus making the tool easier to use while at the same time empowering the interpretation and analysis of the tool’s outputs.*

*Before the implementation of this user interface, the only possible way to interact with the tool was by a Jupyter Notebook, where inputs must be given directly by modifying the code. This required the user to have a fully functional development environment, with all the complexity implied and it also required the user to be familiar with Python programming. Moreover, there was no aid in the interpretation of results, so the only option was to consult the raw output audio files and images. This type of interaction is suboptimal, especially in terms of consulting the results, as one is forced to consider one element at a time, precluding the possibility of increasing the quality of the analysis by considering multiple aspects at once.*

*PLC algorithms are typically needed in real-time data transmission contexts, and their role is to mask (“conceal”) any loss that may occur to one or more packets in a manner that is transparent to the application consuming the data stream.*

*The PLC Testbench Tool aims at providing a modular framework to support a qualitative and quantitative comparison of different PLC algorithms’ performance, by applying two or more of them to a set of audio files provided as an input to the tool. Packet loss is simulated by applying stochastic**mathematical models with configurable parametrization. The software package includes an implementation of the most frequently used packet loss simulation algorithms such as the Binomial PLS, based on binomial loss probability distribution and the Gilbert-Elliott PLS, based on a two-states Markov’s chain.*

*The qualitative analysis is supported by several visual representations of the audio signals’ features (lost packets masks, waveforms, spectrograms) and by the possibility to play any subset of each produced audio track to evaluate its perceived quality.*

*The quantitative analysis of the PLC algorithm outcome is supported by multiple metrics produced by the tool for each of the reconstructed audio files like the Mean Squared Error (MSE), the Mean Absolute Error (MAE), the Spectral Energy, the PEAQ (Perceptual Evaluation of Audio Quality).*

*As a framework, the tool can be easily extended by providing additional implementations of the different object types (PLS algorithm, PLC algorithm, output analysis metric).*

*The GUI can handle these extensions transparently exposing the new features without the need of any code modifications. This is achieved by exploiting the underlying programming language’s introspection capabilities and by scanning the PLC testbench library to discover the custom features.*

*Each elaboration of the tool produces a result dataset, conventionally called a “Run”, which contains a snapshot of all the input parameters and of all the output produced (lost packets masks, reconstructed audio, metrics’ file).*

*The elaboration can be represented as a forest of trees whose root node is associated to one of the audio file given as input. Each input file originates multiple loss simulations, one for each instance of a packet loss algorithm configured. Loss simulations can be mapped to nodes at depth one in the trees and the corresponding audio tracks are in turn reconstructed by applying each one of the PLC algorithms configured at Run level.*

*Reconstructed audio files are represented by nodes at depth two in the trees. Finally, each reconstructed algorithm is evaluated by applying each metric configured in the Run to the corresponding reconstructed audio track.*

*Multiple instances of each configuration object can be set on a Run, as long as the corresponding settings are unique within the Run.*

*During the design phase different types of GUI were evaluated and their pros and cons were carefully weighted up. Finally, the decision was made to go for the development of a web application because of the many advantages offered by the web technology, like a wide range of deployment modes (ranging from a local standalone environment to big, distributed infrastructures), the portability of the application on different platforms, the ease of installation. All of this comes at the price of a slightly increased effort in the implementation.*

*In order to make the distribution of the software as easy as possible, despite the very large number of dependencies in terms of required libraries, it was decided to leverage containers technology.*

*The design of the GUI was made with a particular focus on modularity to make maintenance and future enhancements easier. Modularity was achieved by building graphical widgets that can be reused across the application.*

*The development of the application presented several challenges, especially because of the large amount of data inherently involved by the audio processing. Even short audio tracks are composed of millions of samples that needed to be represented visually as waveforms and/or streamed as sound in real-time. In order to reach the desired performance subsampling techniques and multiple optimizations had to be adopted.*

*Another challenging aspect was the wide spectrum of technologies and programming languages to be learnt in a relatively short period of time.*

1. **Introduction**

Nowadays an increasing number of applications exploit the public Internet to implement remote interactions with users or with other systems and those dealing with digital audio make no exception. They range from streaming applications, to audio editing programs, to sophisticated solutions in the domain of the so-called Internet of Sounds. In most cases the Internet is perceived by the users as a quite reliable infrastructure because transmission protocols like TCP build transport reliability on top of IP protocol, which by itself does not provide any guarantee on transport. The drawback of this reliability is however an increased latency caused by retransmission of lost packets. While this can be acceptable for most applications, it becomes a severe problem in applications working with video and audio media, which do not tolerate very well high latency. Actually, most of these applications prefer to live with a not reliable protocol like UDP in exchange for reduced latency, but this implies that packets can, and typically will be lost. The rate of this loss is strictly related to the congestion or the presence of malfunctioning or broken devices in the network between the two communication endpoints. In video streams packet loss causes “freezing” or image quality degradation, while in audio they produce annoying pops and crackles.

As the acronym suggests, PLC (Packet Loss Concealing) algorithms aim at reducing this problem by reconstructing the lost parts of the audio signal using different techniques which all try to “predict” the missing signal based on previously received packets and mitigate the discontinuity created by the impulsive zeroing created by the frame’s losses.

The domain where PLC techniques were applied throughout this work it that of non-speech audio signals, as speech reconstruction requires peculiar techniques that were out of scope for the research from which this thesis originated.

Currently only WAV audio format is supported but extension to other formats could easily be achieved by pre-processing the input files and normalizing them into WAV format.

* 1. **Digital audio fundamental concepts**

Sound transmission is a phisical phenomenon consisting of a pressure wave propagating across gaseous, liquid or solid medium. The frequency, amplitude and phase of this wave can change over time, originating a frequency spectrum. Sound can be perceived by the human ear only if any of the frequencies in the spectrum falls within a certain range, called human auditory field.

Digital audio is a representation of sound in a digital form. Typically the pressure wave amplitude is sampled at regular intervals and its value is mapped to a discrete range of values, thus originating a stream of numbers.

According to the Nyquist–Shannon sampling theorem, in order to sample all frequencies included in the human auditory field an audio signal must be sampled at a rate of 44.1 kHz .

|  |  |
| --- | --- |
|  |  |
| Figure 1 – a) Wave pressure affecting air density [1] | b) example of wave sampling [2] |

* 1. **Packet loss simulation**

The packet loss can be simulated by randomly dropping some packets from the original audio stream based on a given probability distribution. The random variable represents a packet transmission whose possible outcomes are successful transmission or packet loss.

### Binomial probability distribution

In this model the probability of the random variable to take the value corresponding to a lost packet follows a binomial probability distribution, which represents the sum of a series of multiple independent and identically distributed Bernoulli trials. A Bernoulli trial is a random experiment with exactly two possible outcomes (success/failure), whose probabilities are p [0, 1] and q = 1 – p [0, 1].

The resulting probability to get **k** success/failure events in **n** trials is described by the following formula based on the binomial coefficients.

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Figura 1 – Probability to get k success/failure events in n trials

### Gilbert-Elliot loss model

Compared to models based on a single probability distribution, Gilbert-Elliot model has the big advantage of keeping into account the correlation between loss events. While in the binomial distribution loss model each loss event is unrelated to the previous one and therefore its probability is independent, in the Gilbert-Elliot model the probability of each loss event is conditioned by the previous one.

This behaviour is in line with the assumptions of the Markov’s chains and in fact Gilbert-Elliot model can be considered a two-state Markov’s chain whose graph is the one depicted below.

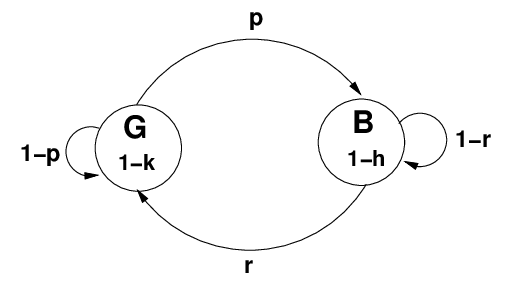


Figure 2 – Gilbert-Elliot’s model of bursty packet loss as a Markov chain [3]

This model is much more accurate compared to single distribution models, because in the real-world packet loss is usually caused by failures or congestion in the transmission network and therefore tends to happen in bursts.

So, when a first loss is experienced, it is more likely that subsequent packets are lost compared to a transmission success scenario, because the system has usually transitioned from a working to a non-working state. In the same way, when the last packet has been transmitted successfully this is an indication that the system is working properly and therefore the probability that the next packet will not be lost is higher than when the system is in a non-working state.

* 1. **Packet Loss Concealment (PLC) algorithms**

PLC algorithms are typically needed in real-time data transmission contexts, and their job is to mask (“conceal”) any loss that may occur to one or more packets in a manner that is transparent to the application consuming the data stream. Any PLC algorithm needs to satisfy the following requirements:

* Its execution time needs to be smaller than the period between the consumption of packets.
* Its output doesn’t need to match exactly the content of the corresponding packet. However, when put back in place of the lost packet, it needs to be perceptually indistinguishable from the original audio stream.

There are many ways to perform PLC on audio signals. From simpler waveform-based solutions such as waveform repetition and PSOLA to more advanced autoregressive models. Recently, Deep Learning based PLC has begun to arise in the literature with some noticeable examples both in speech and music.

Regardless of the type, they all use the audio data received immediately before the lost packet as the input to the algorithm, which will output the number of samples contained in the lost packet.

### Zeros PLC

The lost samples are replaced by zeros. This algorithm is used to generate the audio file representing the result of the loss simulation. It can be useful as a benchmark for the files reconstructed using other PLC algorithms.

A screen shot of a graph

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Figure 3 – Example of Zeros PLC algorithm. The algorithm simply produces an audio file corresponding to the loss simulation where for any lost packet the signal is zeroed.

### Last Packet PLC

The lost samples are replaced by the last received packet. This is the easiest PLC technique.

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Figure 4 – Example of the Last Packet PLC algorithm. As clearly shown by this picture (green waveform), the algorithm uses the signal from the last packets preceding the loss to reconstruct the missing portion of the file. This is particularly evident in the range of samples going from sample 70 to sample 140 of the zoomed region.

### Low Cost PLC

This module is an implementation of the algorithm proposed in [3].

A graph on a screen

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Figure – Example of

### Burg PLC

Python bindings for the C++ implementation of the Burg method [4].



Figure 6 – Example of a very good quality reconstruction of the singla obrained by Burg’s algorithm with train\_siz=1024 and order=512; in red the original file waveform, in green the one generated by Burg’s algorithm, in blue the signal with the lost packets.

### Deep Learning PLC

Implementation of the algorithm proposed in [4]. Compared to the other algorithms it is more expensive due to the complex processing performed.

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Figure 7 – Example of Deep Learning PLC reconstruction.

### External PLC

Python bindings for C++ to simplify the integration of existing algorithms.

* 1. **Output analysis metrics**

The metrics implemented by the last version PLC Testbench tool at the time of writing this thesis are:

**Mean Squared Error (MSE):** is a measure representing the mean value of the squared differences between paired observations of the same phenomenon. The observation pairs can be represented by an estimatated value versus an actual value or by measures of the same parameter collected at different times, or any other pair of values that make sense in the context of the statistical analysis being performed. It is computed as:

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**Mean Absolute Error (MAE):**

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**Spectral Energy: ???**

**Perceptual Evaluation of Audio Quality:** is a standardized algorithm for objectively measuring perceived [audio quality](https://en.wikipedia.org/wiki/Audio_quality). Among all the metrics implemented out-of-the-box in the PLC Testbench tool this is certainly the most specialized and most effective one for the domain of audio signals processing because it is based on a [psychoacoustic model](https://en.wikipedia.org/wiki/Psychoacoustics) of the human ear and brain. The model takes into consideration human auditory field and its limits and compares a reference signal with a test signal, ignoring differences that are not considered as perceptible according to the model’s rules. PEAQ transforms the inputs by applying filter banks to the Discrete Fourier Transform (DFT) of the signals, producing an intermediate output in the form of a set of variables, each one capturing a different psychoacoustic dimension.

Finally the output variables are used to feed a neural network that simulates the cognitive processes happening inside the human brain which translates the input variables into an overall quality score named Objective Difference Grade (ODG).

The algorithm comes in two different versions: the basic one, less computationally expensive, and the advanced one, more accurate but computationally heavier and therefore not always applicable to real-time signals.

A diagram of a process

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Figure 8 – Perceptual Evaluation of Audio Quality. [5]

1. **PLC Testbench UI**
   1. **Web GUI’s main functions**

The purpose of the user interface for the PLC Testbench is to make the tool easier to use, reduce the time required to consult results and increase the quantity and quality of information obtained from the analysis of results.

Before the implementation of this user interface, the only available interaction took place directly in a Jupyter Notebook, where inputs had to be provided directly by modifying the code. Moreover, there was no aid in the interpretation of results, so the only way was to consult the raw output audio files and images. This type of interaction is suboptimal, especially in terms of consulting the results, as one is forced to consider one element at a time, precluding the possibility of increasing the quality of the analysis by considering multiple aspects at once.

The user interface exposes all the functions provided by the underlying PLC Testbench and can be conceptually divided into four modules:

1. Configuration of the input data
2. Progress monitoring
3. Results analysis
4. Elaborations search and inquiry
   * 1. **Landing page**

The landing page of the application is divided into two sections: one contains a document providing a general overview of the PLC Testbench tool, the other contains a list of the main tasks supported by the application.

In the context of the PLC Testbench UI, the purpose of the landing page is helping the user to develop a mental model of the application as accurate as possible and putting the user in condition to operate on the application as quickly as possible. The help section of the page addresses the first objective, while the tasks list addresses the second one.

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**Figura 2** – Landing page of the application. On the left side an overview of the PLC Testbench tool is presented. On the right side the user can find an entrypoint to the main use cases supported by the application.

* + 1. **Configuration of the input data**

The purpose of this module of the GUI is to allow the selection of the files to be processed by a Run of the PLC Testbench tool and to allow the configuration of the sets of packet loss, reconstruction and evaluation algorithms to be applied to each input file with the corresponding parametrizations.

Given the multi-step nature of the task and the high number of selections and choices involved in the process supported by this part of the interface, while designing it the focus was put on making the interaction as simple and intuitive as possible.

In order to achieve this goal, it was decided model the main steps of the process through a wizard based interaction where each step has a clear and small purpose and the user goes the sequence of steps in a logical manner. Splitting a complex task into multiple simpler tasks has several advantages like decreasing the likelihood of mistakes, giving a clearer context for error reporting, making it easier for the user to develop a more accurate mental model of the software, giving the users an update on their progress in the procedure, reducing the cognitive load.

At each step the coherence of the data is enforced reporting any error to the user before the next step is undertaken. This way any wrong or missing information is pointed out in a limited scope and therefore the problem is easier to understand and fix.

From a technical side, this was identified from the very beginning as the part of the GUI which is more likely to be impacted by extensions or future developments, so it looked crucial to make it as flexible and auto-adaptive as possible.

This meant that the interface had to be able to deal with the addition of new algorithms with arbitrary parameters in a transparent way, without requiring any code adjustment.

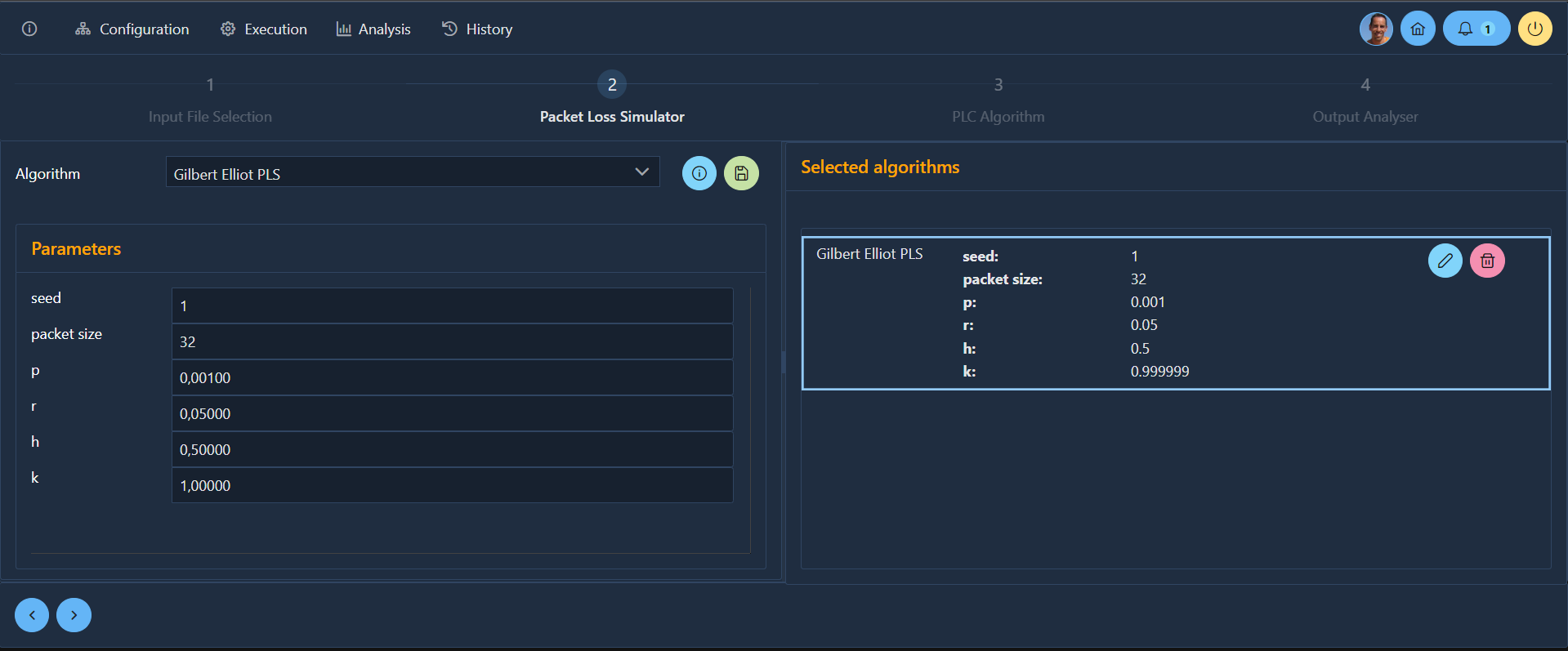
The fulfillment of this requirement was obtained by exploiting the powerful introspective functions of python programming language, that allow to analyze code structure at runtime. Through these APIs it is possible to discover relationships between software components.

In the PLC Testbench UI it is enough for any implementation of a new algorithm to inherit from the base class of if its own category, be it a loss simulation or a packet loss concealing or an output analysis one, for it to get automatically exposed in the corresponding screens with all its properties.

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**Figure 9** – Application screen to configure an elaboration:input file selection phase.



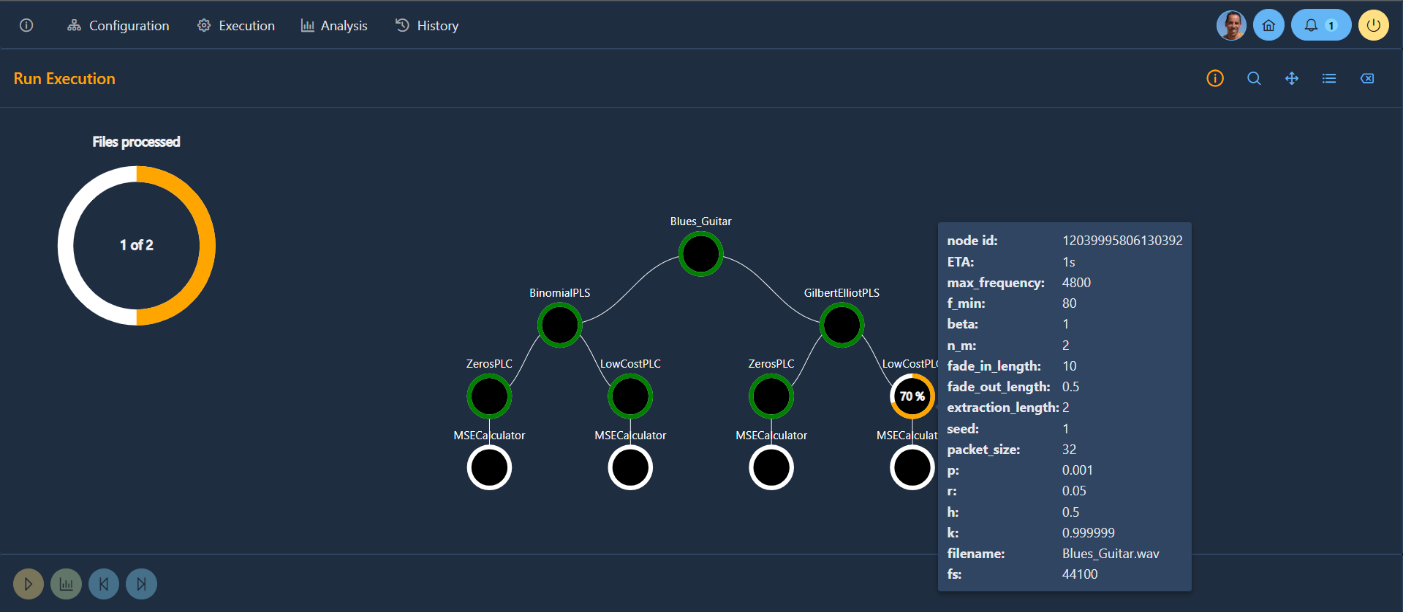
**Figure 10** – Application screen to configure an elaboration: module’s parameters setting. The interface is self-adapting to the underlying PLC Testbench code so that adding new modules or new module parameters can be handled automatically.

* + 1. **Execution management and monitoring**

This module of the GUI is intended to allow the user to monitor the state of an in-progress elaboration, providing some sort of progress bar and information like the ETA (Estimated Time of Arrival) for the sub-steps.

Each elaboration of the PLC Testbench tool is named a “Run” and it contains a snapshot of all the input data and all the generated output results, like the loss simulation masks, the reconstructed files and the metrics evaluated on each reconstructed file. Each of the elaboration’s steps can be mapped to a tree node where the root node represents an input file. The overall Run is therefore a forest of trees.

Since the elaboration is identical for all the input files in terms of the algorithms that are executed on the original track and it is sequential, displaying the entire structure of a Run all the time would imply cluttering the interface with redundant information. It was therefore decided to represent progress at two different levels, that is within each input file tree and globally.



**Figure 11** – Application screen to manage and monitor elaborations

* + 1. **Analysis of the execution’s results**

This module was considered from the very beginning as the most important one, so a great part of the time spent in the analysis and development of the PLC Testbench UI was dedicated to it.

In this case the focus was on providing an easy way to analyze the results from both qualitative and quantitative perspectives.

Qualitative analysis is mainly represented by browsing and visually comparing the waveforms in the regions where packets have been lost and by listening to the audio files generated from the PLC algorithms.

Quantitative analysis is mainly represented by the numerical comparison of the performance achieved by the PLC algorithms according to a given metric like the Mean Squared Error, the Mean Absolute Error, the Spectral Energy, the Perceptual Evaluation of Audio Quality (PEAQ) with respect to the specified input parameters.

Another extremely important aspect was providing the capability to integrate and correlate as much as possible the different types of information to get deeper insight into the behavior of the algorithms being tested.

The main requirements in this case were:

* clearly show the lost portions of the original file
* allow easy comparison of the waveforms ranging from very high level to sample level detail
* give the possibility to display every combination of waveforms in the comparison
* give the possibility to play any audio track, original or reconstructed setting the playhead at any position by simply clicking on the waveform
* support a very wide range of zoom levels including a sample level one
* display visually the metrics calculated on the reconstructed audio files
* display a spectrogram of the audiofile, be it the original or a reconstructed one

The need to display multiple information at the same time made a dashboard interface more appropriate and appealing in this case. The dashboard was organized in 4 sections where the first one is focused on representing the waveforms of all the audio versions of an input file and providing navigation across the files and zoom functions. Lost samples region are displayed together with the waveform and when clicked the corresponding detailed data are loaded in the section below.

The second section provides the ability to inspect the waveform at sample level in any region were some samples were lost. This is extremely useful to evaluate from a qualitative point of view the behaviour of the algorithm and to cross check the correctness of the algorithm’s implementation as visual representation makes any misbehaviour much more apparent.

The third section is intended to display the metrics calculated on the reconstructed versions of the analyzed file and is made by two different charts, one containing the figures from metrics producing time series, the other representing the figures from metrics producing single values.

Finally the fourth section displays the spectrogram of the current audio file, representing the distribution of the frequencies composing it and their relative and absolute intensity in the form of a color map.

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**Figure 3** – Waveforms and packet loss visualization.

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**Figure 4** – Zoomed waveforms and packet loss regions.

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**Figure 5** – Sample visualizer: when a specific packet loss is selected by clicking on it, the sample visualizer displays a view of the signal that can be zoomed at single sample level for each version of the audio file, original or reconstructed.

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**Figure 12** – Metrics visualizer: linear metric.

A screenshot of a computer screen

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**Figura 6** – Spectrogram: a view presenting a spectrogram of the selected portion of the file in the form of a heat map

* + 1. **Elaboration search and inquiry**

After the concept of “Run” emerged as a way to organize input and output of a specific elaboration, the need to design some search function became more and more clear.

The purpose of the inquiry function was not only to allow recalling previous elaborations, but also to allow finding out similar elaborations for comparison purposes.

In order to make searching as easy as possible but at the same time keep it very powerful, flexible and resilient to code changes or extensions, a query builder component has been exploited.

The component allows to graphically create a filter by adding multiple conditions and connecting them with the logical and/or operators. It is also possible to parenthesize the conditions by nesting them in groups.

All the relevant fields, both at run level and at configuration level can be referenced in each condition and they can be matched against specific values by using a wide set of operators, depending on the type of the field.

Given the high number of parameters involved in the configuration of a PLC Testbench run, queries can quickly become quite complex and writing them from scratch each time could become frustrating and time-consuming. To address this problem filter saving functions have been provided, allowing to store each filter persistently in the application database, while associating a meaningful name to them.

Finally the results are presented in the bottom section of the page.

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**Figura 7** – Run History page. In the search section custom queries can be built with the help of a query composer. In the Run list section the results of the search operation are displayed.

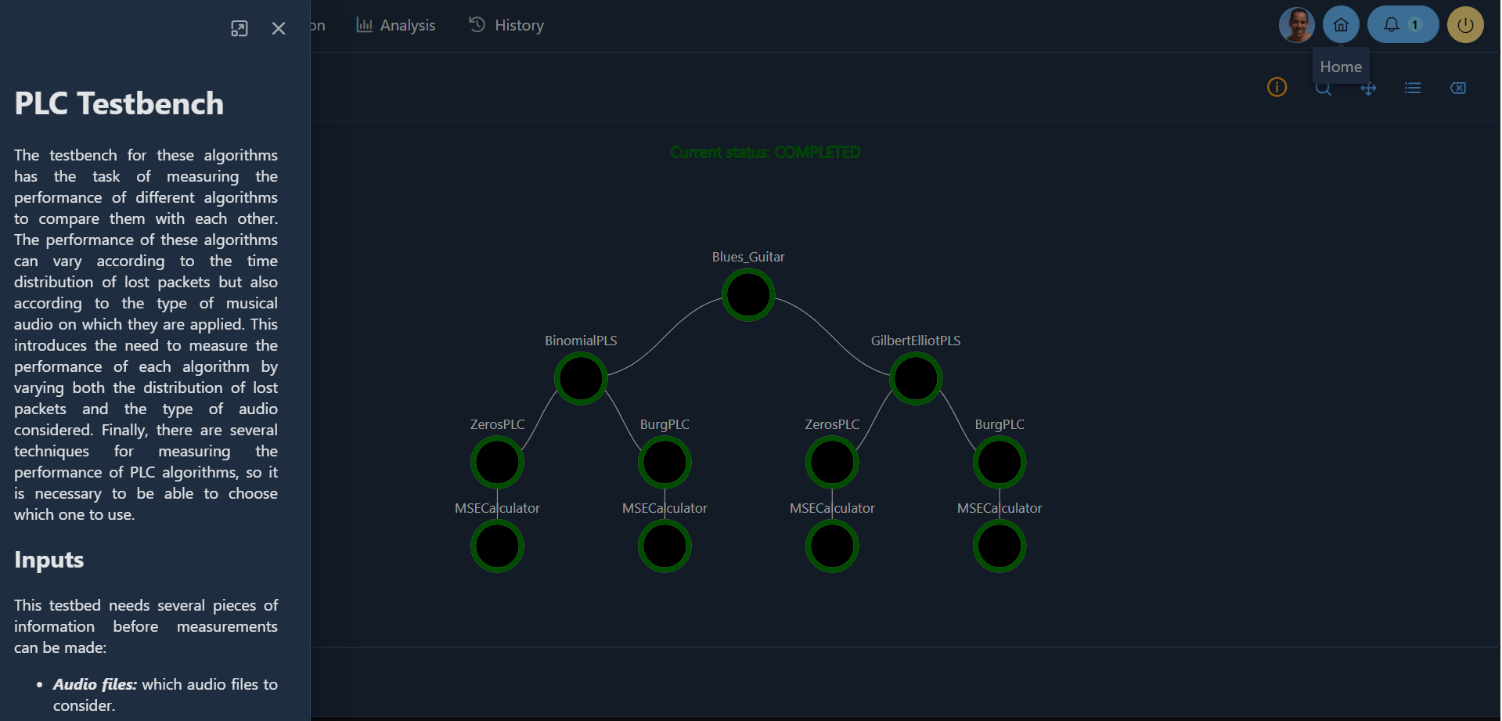
* + 1. **Utility functionalities**

Several utility functionalities have been developed to keep the user constantly informed about relevant information. For example by clicking on the info button in the top menu bar it is possible to display an help page about the currently selected functions.

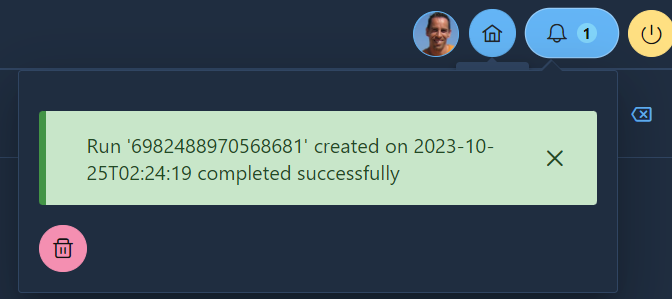
As each elaboration can take a significant amount of time, depending on the number of input files, the duration of each audio track and the parameters of the simulated loss, reconstruction algorithm and output metric evaluation, a notification menu has been added to the top menu bar.

When an elaboration is complete a notification is dropped in the user “mailbox” and the number of unread notification is displayed on the component.

This is particularly helpful because the results of an elaboration can be analyzed only once the elaboration is completed, therefore a push notification mechanism can free the users from having to check regularly the elaboration status while carrying out other activities in the application.



**Figure 13** – In each context a relevant help page is displayed in the side bar



**Figure 14** – Push notifications: when relevant events happen (e.g., an elaboration is completed) a notification is displayed in this area

* + 1. **Responsive layout and PWA (Progressive Web Application)**

The application has a responsive layout that can adapt gracefully to most screen dimensions, ranging from big workstation monitors to tablets and smartphones screens.

The application can be installed as a Progressive Web Application giving the user a “desktop-like” experience, where the application can be run by just clicking on a desktop icon and screen space is not wasted by the browser toolbars.

A screenshot of a computer

Description automatically generatedA screen shot of a device

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**Figure 15** – a) Tablet layout b) Smartphone layout

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**Figure 16** – Progressive Web App layout



**Figure 17** – Progressive Web App launch shortcut

* 1. **Technical choices**

During the design phase different types of GUI were evaluated and their pros and cons were carefully weighted up. Finally, the decision was made to go for the development of a web application because of the many advantages offered by the web technology, like a wide range of deployment modes (ranging from a local standalone environment to big, distributed infrastructures), the portability of the application on different platforms, the ease of installation.

* + 1. **Project architecture**

The user interface application is made of a Web application composed of two layers: one managing the user interface representation and the interaction with the user, the other providing the backend services to be consumed by the graphical components.

The application therefore can be used both as a standalone application by deploying it on the user machine or in a multi-user environment where the application is deployed on one or more remote servers.

The two layers exchange information via a RESTFul API which, if needed, can be exploited also by third-party services. The API is secured by using OAuth2 protocol.

In the presentation layer, modularity has been achieved through the creation of specialized but highly customizable components that can easily be reused across different pages.

In the backend layer, modularity is provided by the extensibility of the RESTFul API where new functions can be “plugged in” by simply adding new endpoints. This layer is also hiding the details of the interaction with the PLC Testbench from the presentation layer, providing an abstract and stable view of the library functions.

Persistence is based on a NoSQL database, where the data are stored in JSON format in order to provide higher flexibility to accommodate future evolution of the schema. Both on-premise and cloud databases are supported.

The GUI adapts automatically to the PLC Testbench library by using introspection to retrieve the list of the algorithms used for packet loss, PLC, and output analysis. This way any extension made to the underlying tool does not require any manual change to the UI code.

The application architecture has been designed to be as stateless as possible and thus supports very well horizontal scalability, which means it is possible to split the workload among as many instances of the application as it is need by providing proper load balancing in front of them.

Scalability together with containerized distribution makes the application suitable for deployment in an orchestrated environment.

A diagram of a computer program

Description automatically generated

**Figure 18** – Application’s layered architecture and layer mapping onto corresponding frameworks and languages

* + 1. **Adopted programming languages**

The criteria that drove the selection of the programming languages to be used in the project were the following in order of importance:

1. More portable language are to be preferred over less portable language.
2. More largely adopted and supported language are to be preferred over less widespread language.
3. The languages already used by the PLC Testbench are to be preferred in order to reduce the project maintenance complexity.
4. Languages with better support for scalability are to be preferred over less scalable languages.

Based on this criteria Javascript popped up as the best choice for the core UI layer because of its out-of-the-box integration with all browsers, its being the de-facto standard for the web and the huge number of GUI widgets libraries freely available. On the other side Python seemed like the most natural choice for the Rest API and services’ layer, given its tighter integration with the PLC Testbench tool and the availability of several libraries and frameworks inherently designed for that purpose.

Adopting the same language used by the PLC Testbench tool helped a lot in keeping the project complexity as low as possible, both in terms of languages and in terms of final artifact’s distribution.

* + 1. **Frameworks**

Although frameworks in some cases tend to increase the complexity of software projects, in this case it was judged to be worth adopting some. In fact, leveraging available solutions to common problems was seen in this case not only as a way to avoid “reinventing the wheel” and speed up the project delivery, but also as a mean to standardize the code organization and incorporate some important best practices and design-patterns.

The frameworks used in the PLC Testbench UI codebase are the following:

|  |  |
| --- | --- |
| **Framework** | **Description** |
| **React JS** | probably the most important one; the library makes declarative, component base development of GUIs much easier, by providing support for components’ lifecycle management, state management, UI rendering management and optimization. |
| **PrimeReact** | a rich library of powerful react graphical widgets with a very nice look and feel which allowed fast and reliable implementation of the interface and supported easy refactoring and layout rearrangement. |
| **Node js** | used essentially to manage javascript dependencies through the npm package manager. |
| **D3js** | for the highly specialized tasks related to visual representation of data in most cases D3js library was used behind the scene by wrapping the d3js code into brand new custom React components. |
| **Wavejs** | a flexible graphical library to represent layered audio waveforms by SVG (Scalable Vector Graphics). |
| **Flask** | a lightweight framework providing all the functionalities needed by a REST API based web application. |
| **MongoDB** | persistence of the data was achieved by means of MongoDB, which is one of the most widespread opensource no-SQL databases. The choice of a no SQL database was mainly influenced by the fact that this technology does not impose a rigid schema on the data and therefore looked more flexible and adaptable in case of future extensions of the software |
| **PyMongo** | a library for Python programming language that makes interacting with MongoDB much more convenient by providing higher level abstractions over the data layer. |
| **Wavesurfer** | library to display audio waveforms, supporting a rich set of plugins, one of which was exploited to support spectrogram visualization for PLC Testbench UI. |
| **Waves-UI** | library to display custom audio visualizations based on stacking specialized layers one on top of the other, like cursor layer, waveform layers, segment layers, datapoints layers. |

One of the most complex parts of the graphical user interface both from design and implementation perspective is certainly the component displaying audio files’ waveform. Despite the existence of several libraries accomplishing a similar function, none of them matched perfectly the specific needs of the project, as none of them could be natively integrated into React and each one did not cover all the identified use case scenarios out of the box.

In some cases the mismatch was mainly concerning the visual functions, like not being able to present multiple graphs on the same timeline, in other cases it was more related to data handling.

Peaks.js for example is a very nice looking framework developed at BBC, mainly intended to support audio files annotation and editing, but in PLC testbench’s context it presented some severe limitations in zoom management. In PLC analysis being able to span from full track to extremely detailed waveform visualization is an important requirement, but the approach to zoom taken from the library is discrete, supporting only a set of precalculated zoom levels. With this regard the choice made in PLC Testbench UI project was dramatically different as it was decided to manage zoom in a continuous way rather than in a discrete one. In order to reach adequate performance several optimization had to be applied like subsetting and downsampling techniques. This way the system can deal only with the portion of the data actually displayed at the minimal resolution actually needed.

Nevertheless, after a preliminary evaluation, one of the library seemed to be flexible enough to work as good starting point for the implementation of the needed set of React components. The waves-ui library, thanks to its approach based on composable layers, could support one of the main requirements for the waveform visualization components, which was displaying multiple overlapped waveforms and lost segment in the same view.

* + 1. **Packaging and distribution**

With regard to packaging and distribution, the objective was a zero-setup installation of the tool and the proper technology to achieve it was considered to be Docker containers.

The PLC Testbench UI is available as a docker public image built from the master branch of the publicly available GitHub repository of the project. The only prerequisite to get the tool running in standalone mode is having a Docker installation on the target machine.

In order to start the application two containers have to be created, one to host a MongoDB instance and one to host the Flask web application. The template for command lines to launch each container are provided in the README.md file of the PLC Testbench UI’s GitHub project and must be customized in the parts which are necessarily specific to each user’s environment, like volume or SSL certificates setup.

This means that the PLC Testbench UI can run on any platform where Docker can run, like all Linux distributions, Windows and MacOS.

The Docker containerization, together with the packages installer tool from Python also helped in managing the huge amount of transitive dependencies brought into the project by specialized libraries for audio manipulation like gstreamer, librosa or by others for more general purposes like tensorflow.

* + 1. **Software Development Process**

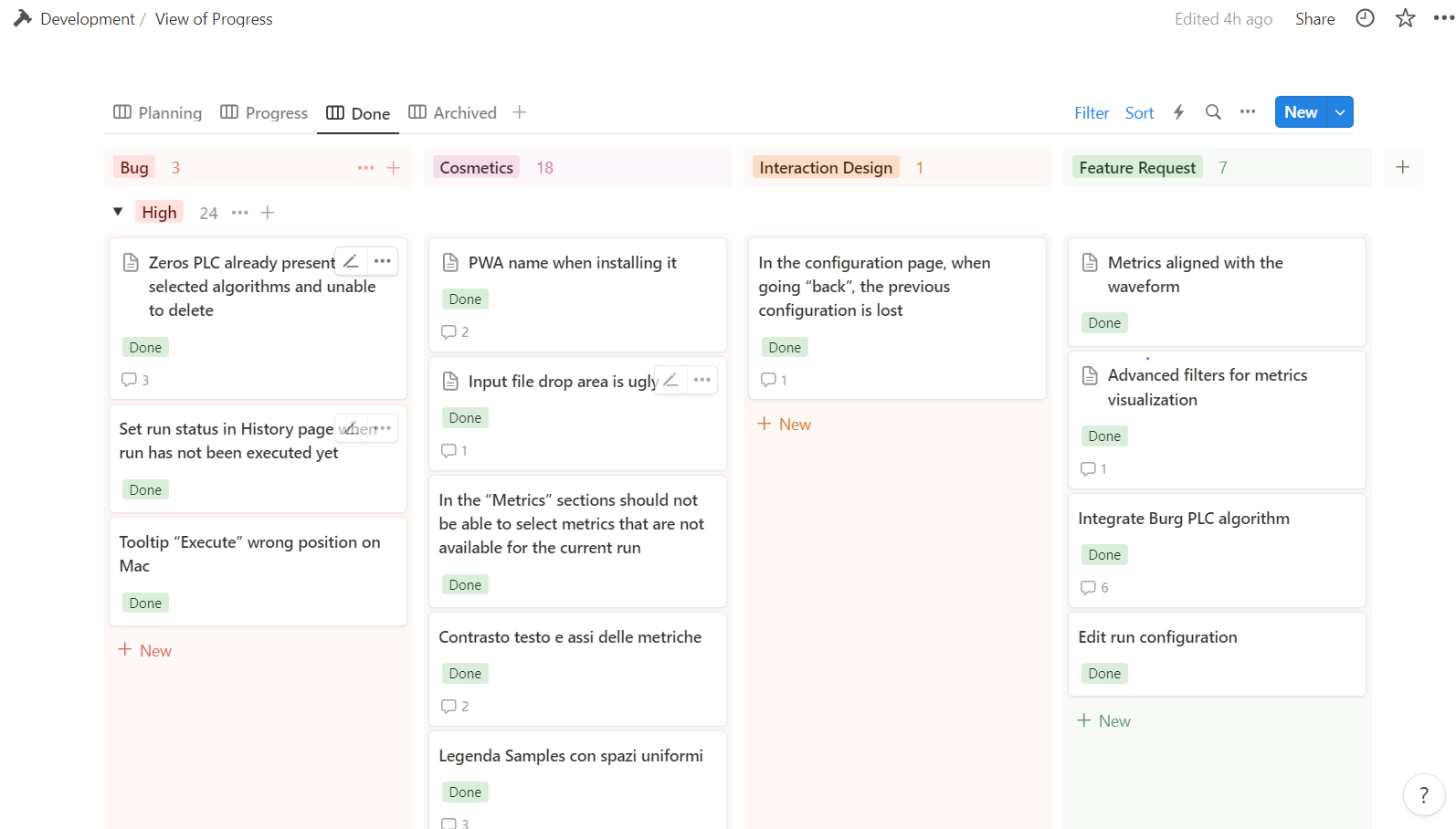
At the beginning of the thesis work my knowledge about the domain of digital audio signals processing was extremely limited and my knowledge on the specific subdomain of packet loss concealing was close to zero. Therefore, a major concern was keeping software delivery iterations as short as possible in order to avoid wasting time on useless features and in order to allow fast feedback on the developed features.

A software development process based on Kanban methodology seemed to be a perfect match for this agile way of working because of its simplicity compared to other agile models like Scrum, which made it more appropriate and effective for a project managed by a single developer. Enforcing the strict role separation prescribed by Scrum or applying advanced Scrum techniques like velocity estimation and burn down charts would probably represent an overkill in this case.

The process was enforced through a combined use of Notion and GitHub to manage issues backlog.

Notion proved to be an excellent tool to quickly capture any type of request about the application, be it a bug fixing request or a new requirement to be addressed or a new feature to be developed.

On the other side GitHub proved to be a very powerful tool to track the correlation between the change requests and their implementation in the code base, combining at the same time simplicity, expressiveness, and tight integration with the source code management system.



**Figure 19** – Notion board organizing issues according to their priority and type

A screenshot of a computer

Description automatically generated

**Figure 20** – GitHub’s issue management

The current status of the GUI is the result of an iterative development process during which initial ideas where prototyped and reviewed in close collaboration with prof. Luca Turchet and Luca Vignati. Their precious feedback was crucial in recognizing new emerging use case, which lead to the expansion of the application functionalities, and made it possible to gradually improve both the appearance and the ease of use of the application.

A thorough evaluation of the GUI usability was not carried out as it was far beyond the purpose of this thesis, given the considerable effort already implied by the software development activity and the bachelor thesis’s nature of the present work, but it could represent an opportunity for a future dedicated work.

Nevertheless, a lot of attention was paid to making the interface as clear as possible by adopting a minimalistic design, standard conventions, and consistency across the entire application, making interaction flow as smooth as possible, providing the user with all and only the relevant information in the current context, detecting errors as early as possible and providing meaningful error description and instructions on how to solve or avoid the problem.

The need to release code frequently dictated by the agile development process adopted, also led to the decision to setup a Continuous Integration/Continuous Delivery infrastructure to automate the release of new features in the test environment. CI/CD pipelines are a very powerful tool as they allow describing the build and deploy process in the shape of declarative files that can be automatically parsed and executed by CI/CD platforms. The tools can also execute different test suites to address unit testing, regression testing, integration and load/stress testing before actually deploying the artifacts, thus ensuring a proper quality of the deployed releases. On top of all the usual advantages coming from automation, like increased reliability, higher standardization, increased traceability of the process, being simple text files, pipelines descriptors can be managed like any other source file in the project. This means they can be stored in the source code repository, thus getting the benefits of versioning and more reliable management in collaborative environments where multiple developers work on the same project.

Automated deployment can be automatically triggered when source code is committed to the repository, so that any code commit that is successfully built and tested can be immediately propagated to the test environment.

Circle CI cloud was selected as the CI/CD platform for the PLC Testbench UI project because it is very powerful and it is freely available for small teams and non-commercial use.

Having a CI/CD infrastructure in place helped a lot in shortening the release cycles’s duration and in preventing deployment mistakes related to the differences between the local development environment located on a Windows laptop and the test environment running on a Mac OS server.

A screenshot of a computer

Description automatically generated

**Figure 21** – Circle CI pipeline automatically building and pushing a new Docker image to Docker Hub whenever any change is committed to the master branch of the application’s GitHub repository

1. **Results**

The most important results achieved through the PLC Testbench UI development can be summarized as follows:

* Support to the validation and debugging of the PLC Testbench tool
* Highlighting improvement areas for the PLC Testbench tool
* Extension of the tool usability
* Empowering of PLC Testbench tool’s result analysis
* Installation of the PLC Testbench made easier and more stable through containerization
  1. **Support to the validation and debugging of the PLC-TestBench tool**

During the development of the Graphical User Interface, it has become evident how the GUI could support and speed-up the debugging of the PLC algorithm, thus acting also as a validation tool for the PLC Testbench itself. The possibility to compare the waveforms of the original files with those of the reconstructed files at different level of details and focusing only on the regions affected by packets loss made it much easier to confirm the compliance of the PLC algorithms with some essential invariants.

One of these invariants is for example the fact that signals must being identical to the original one in every region where packets have not been lost except for some deviations in the area immediately following a packet loss where there can be some border effects caused by a gradual fading of the PLC reconstruction to make transition smoother.

The detailed visual representation of the signals also made it easier to discover any weird behavior in the PLC algorithms. Without a visual representation more sophisticated than the raw images generated natively by the PLC Testbench tool, finding out such situations would probably have been much more time consuming or even not possible.

* 1. **Highlighting improvement areas for the PLC Testbench tool**

Before the development of the GUI the PLC Testbench tool had only a minimal persistence layer built around filesystem storage for audio file output and .pickle files for configurations and packet loss masks. Pickle is a serialization format natively supported by python which allows to persist memory objects into a file and rebuild them when needed.

The inputs and outputs of each elaboration were grouped by means of filesystem directories and to avoid collisions in the folder names naming conventions had been adopted to encode the parameters. Naming conventions, together with the elaboration logs also represented the main criteria to find out the location where the outputs had been stored.

Every elaboration was unrelated from the others and the inquiry functions were completely unsupported.

While developing the GUI slightly the “Run” concept emerged as a container for the metadata related to an elaboration and together with it the need to move to a more sophisticated persistence layer became clear.

At the beginning a raw set of inquiry functions was built on top of the existing persistence but they were extremely limited due to the poor information available about the “Run”. Finally the decision was made to exploit a database to collect the relevant metadata of the elaborations.

Database based persistence paved the way for the development of a much more powerful inquiry module, exploiting a query engine to dynamically build queries. It also led to further insights, like the possibility to store into persistence the hashes of the files and configuration parameters used by each elaboration and use them to enable performance optimization in the PLC Testbench by completely removing redundant processing across different elaboration. Thanks to the introduction of the hashing feature it was also possible to build a validation layer to detect and prevent data inconsistencies.

* 1. **Extension of the tool usage**

Providing a GUI for the Testbench Tool improves the overall user experience, makes the learning curve of the tool less steep and extends its usability to users not having a programming background.

* 1. **Empowering of PLC Testbench tool’s result analysis**

Compared to the out of the box representations provided by the PLC Testbench tool which are essentially static images, the PLC Testbench UI allows real-time exploration of the data at different level of detail. The detail level can be selected interactively, depending on the needs and the data are more integrated by being displayed, when comparable, as multiple time-series on the same graph.

In order to improve the user experience the GUI has been made responsive so that it can automatically adjust to different devices (PC, tablet, mobile).

The interface has been designed to be user-friendly, intuitive, and as fast as possible, considering the large dataset inherently involved in audio processing applications and the overhead related to charting functions. Whenever possible subsampling or caching techniques have been applied to minimize latency and network bandwidth waste.

* 1. **Comparison between native PLC Testbench’s I/O functions and the Web GUI**

Below a visual comparison is performed between the input/output formats consumed/produced out of the box by the PLC Testbench and the corresponding Web GUI functions. The comparison is structured by functionality.

|  |  |
| --- | --- |
| **Elaboration configuration** | |
| PLC Testbench native functions | Web GUI |
|  |  |
| **Figure 22** – The configuration must be edited directly in the Integrated Development Environment (IDE), by changing the list of the objects passed as an input to the main PLC Testbench class instance. No validation is performed before the application is actually run. | **Figure 23** – The configuration is performed in a visual way, going through a series of guided steps and getting immediate feedback on the its validity. In the upper screen input files can easily be uploaded and selected. In the lower screen the settings for a Gilbert-Elliot packet loss simulator instance are configured. |

|  |  |
| --- | --- |
| **Progress monitoring** | |
| PLC Testbench native functions | Web GUI |
|  | A screenshot of a computer  Description automatically generated |
| **Figure 24** – The native progress monitor does not show the hierarchical nature of the elaboration and the only information available for each step are the progress and the time to complete | **Figure 25** – The Web UI shows both the overall progress (on the left side) and the progress within the current file elaboration (on the right side). The progress information is presented in a hierarchical fashion, and enriched with the specific parameters of each node (see tooltip in the right-bottom corner of the picture). |

|  |  |  |  |
| --- | --- | --- | --- |
| **Results analysis** | | | |
| PLC Testbench native functions | Web GUI | | |
|  |  | | |
| **Figure 26** – Visual plots are saved as bitmap images and the entire track is represented, thus fixing resolution at creation time and making overlapping multiple waveforms unmanageable. No sample level view is available. | **Figure 27** – In the Web UI the same informations are displayed but in a more integrated and comparable way. Packet loss regions and waveforms are overlapped, a sample level zoom function and a spectrogram are available. All the views allow the selection of the dataset to be displayed and share the same zoom capability. Finally they are scroll together with waveforms when playback is activated in zoomed mode. | | |
| **Audio files playback** | | | | |
|  | | Web GUI | | |
|  | |  | | |
| **Figure 28** – In order to play a specific track’s version the user has to navigate across the output folder of the application in the Integrated Development Environment (IDE, Visual Studio Code in this case) and find the proper audio file by inspecting the parent folders’ and files’ names. | | **Figure 29** – The user can easily select what track and what version of the track to play. | | |
| **Search** | | | | | |
| PLC Testbench native functions | | | Web GUI | | |
|  | | |  | | |
| **Figure 30** – Only available through MongoDB clients or Visual Studio Mongo extension, by directly executing an handwritten query. The output is in JSON format. | | | **Figure 31** – Queries can easily be composed through a builder by adding groups of conditions connected by logical operators. | | |

* 1. **Future work**

Despite providing an implementation for all the basic recognized use cases and being fully operational, the PLC Testbench UI has obviously still a lot of room for improvements. Future enhancements of the GUI could be dictated by the evolution of the PLC Testbench itself or could be more strictly connected to the user interaction.

The following list mentions a few features that could be developed in future releases of the application:

* extending support to additional audio file formats, like mp3, ogg, flac, etc.
* providing a function to expose the PLC Testbench logs
* implementing search functions on output results
* implement different user interface themes
* internationalize the UI by adding support for additional languages
* implementing user collaboration use cases
* extending supported identity providers for login (currently only Google)
* developing export functions for the most widely used formats (Excel, CSV, etc.)

1. **Conclusions**

PLC Testbench provided natively some basic visual outputs but they were not sufficient to perform an effective analysis of the results produced by the different PLC algorithms. Also, the configuration phase was complex due to the need to install a fully working development environment to be able to modify the parametrization in the Jupiter notebook source code and run it.

Distribution was made difficult by the high number of dependencies involved in the project setup.

The main deliverable of this thesis is a Web GUI for the PLC Testbench that addresses all the mentioned pain points by:

* providing a set of interactive functions to analyze and compare the result of the different PLC algorithms at any level of detail
* guiding the user through the configuration of each PLC Testbench module
* making Docker the only prerequisite to run the application
* providing a containerized bundle with all the needed dependencies

On top of this, the development of the Web GUI based on a multi-layer architecture led to these additional benefits:

* possibility to deploy the application in a highly scalable environment;
* possibility to expose the PLC Testbench functions as services via a RESTful API;
* thorough debug of the PLC Testbench.

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|  |  |
| --- | --- |
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**Attachment A Source code repository**

**PLC Testbench UI’s GitHub source code repository**

The source code of the project is available on the following GitHub public repository:

<https://github.com/stefano-dallona/plc-testbench-ui>

**Attachment B Docker image**

**PLC Testbench UI’s Docker image**

A pre-built image of the application is publicly available at the following URL:

<https://hub.docker.com/r/stdallona/plc-testbench-ui>

**Running pre-built image and building custom images from source code**

If the pre-built image does not fit your needs you can build a suitable image by customizing the Dockerfile located in the root directory of the source code project on published on GitHub.

**Attachment C CI/CD Pipelines**

In order to make release cycles faster and more reliable a Continuous Integration/Continuous Deployment pipeline has been setup on the free cloud-based tool Circle CI.

The pipeline has been developed exploiting the pipeline as code feature of Circle CI, based on yaml language and is stored in the ***.circleci/config.yml*** file in the GitHub source code repository of the PLC Testbench UI project.

**CircleCI CI/CD Pipelines**

# Use the latest 2.1 version of CircleCI pipeline process engine.

# See: https://circleci.com/docs/configuration-reference

version: 2.1

# Define a common Docker container and environment for jobs

executors:

  docker-publisher:

    # Define the image tag

    environment:

      IMAGE\_TAG: stdallona/plc-testbench-ui:1.0.3

    # Use `docker:stable` as the Docker image for this executor

    docker:

      - image: docker:stable

# Define a job to be invoked later in a workflow.

# See: https://circleci.com/docs/configuration-reference/#jobs

jobs:

  build:

    # Use docker-publisher from above as the Docker container to run this job in

    executor: docker-publisher

    # Add steps to the job

    # See: https://circleci.com/docs/configuration-reference/#steps

    steps:

      - checkout

      # Set up a separate Docker environment to run `docker` commands in

      - setup\_remote\_docker

      - run:

          name: Build Docker image

          command: docker build --tag "${IMAGE\_TAG}" .

      # Archive and persist the Docker image

      - run:

          name: Archive Docker image

          command: docker save --output image.tar "${IMAGE\_TAG}"

      - persist\_to\_workspace:

          root: .

          paths:

            - ./image.tar

  push:

    # Use docker-publisher from above as the Docker container to run this job in

    executor: docker-publisher

    steps:

      # Set up a separate Docker environment to run `docker` commands in

      - setup\_remote\_docker

      # Load and un-archive the Docker image

      - attach\_workspace:

          at: /tmp/workspace

      - run:

          name: Load Docker image

          command: docker load --input /tmp/workspace/image.tar

      # Log in to Docker Hub and push the image

      - run:

          name: Publish Docker image

          command: |

            echo "${DOCKERHUB\_PASS}" | docker login --username "${DOCKERHUB\_USERNAME}" --password-stdin

            docker push "${IMAGE\_TAG}"

# Run the two different jobs as a sequenced workflow

workflows:

  version: 2

  build-push:

    jobs:

      # Run the build first

      - build

      # Push the image second

      - push:

          # Build needs to finish first

          requires:

            - build

          # Only push images from the master branch

          filters:

            branches:

              only: master