AI Term Project Report

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Abstract—This class project aims to implement a machine learning based method of designing electrical analog filters using an unbalanced-ladder topology. The AI method developed is compared to standard filter prototypes using G-coefficients and element scaling.

Keywords—AI in Analog Design, Filter Synthesis, G-coefficients, Filter prototypes, Element Scaling

I. DESCRIPTION OF PROBLEM

Analog filters are used in many electronic devices to block unwanted power at specific frequencies from entering the system. Cell phones, radios and radar systems all use analog filters in the front end to prevent out of band frequencies from compressing sensitive receive electronics (leading to degraded performance). There are some closed form methods for computing filter component values, BUT they are limited to very basic filter shapes (like low-pass, high-pass, symmetric bandpass) AND almost always require a skilled human designer to tune the values afterward. Additionally, the closed form filter synthesis methods cannot easily account for higher-order real-world effects which have a big impact on component value selection. These higher order effects include uninteded coupling between circuit elements, component parasitics, and propagation delays for high-frequency distrubuted circuits.

The problem: Existing closed form filter synthesis methods are limited to very basic filter shapes, do not account for impactful higher order effects, and require significant time for human designers to tweak values by trial and error.

Overall Goal of Paper: Make progress towards a machine learning (ML) model that can:

- 1) generate filters with much more arbitrary frequency responses (not just simple low-pass, symmetric bandpass, etc.)
- 2) accurately account for the higher-order effects which closed-form solutions ignore,
- 3) significantly decrease the amount of human tuning required by generating more accurate component values.

Specific Project Scope: This project specifically tackles the design of 5th-order unbalanced ladder filters with series element first. The user gives model the desired filter frequency

response (can be arbitrary shape), and the model designs a working filter.

Figure 3 shows a 3rd order filter topology. For this design, each element of the ladder is made up of a capacitor and inductor. The series elements are made up of a series LC and the shunt elements are composed of a parallel LC. This means there are 10 different electrical component values to select for the filter design.

II. DESCRIPTION OF APPROACH

To achieve the overall project goal the following approach was taken. Each entry has its own sub-section:

- 1) Research existing attempts at ML based circuit designers
- 2) Learn and implement the standard closed form filter synthesis methods *
- Develop method to quickly evaluate circuit performance AND verify against commercial circuit solvers **
- 4) Generate Training Data
- 5) Model Training Details

Importance of criterion selection

Multi-Step Training Program to significantly increase training speed

- * This is critical for publishing a paper in the future. Many of the machine learning papers on circuit design DO NOT compare against standard circuit design methods and lack sufficient electrical engineering expertise. The lack of standard EE knowledge is a large hole in the current literature as of 2025.
- ** generating sufficient data can be expensive in microwave design

A. Research existing attempts at ML based circuit designers

Existing work on ML based circuit designers can be broken into parameter design vs. topological design. Parameter designs have the AI only choose component values for a pre-chosen circuit, whereas in ML topological design the AI generates the entire circuit with very little human guidance.

My class presentation and survey essay go into depth on the recent ML microwave design literature published in 2023 2025. In those submissions the basics of the RF design process are explained, and the potential for limited AI assistance at each step is briefly discussed [1]. Next, the distinction between parameter optimization and topological design is made. [2] was used to demonstrate AI's ability to balance the many parameters required in an RF design whose topology is fixed. Then, the topological AI design algorithms described in [3], [4], [5] are described and methods compared. Particular attention is paid to the wavelet transform used in [3] and how it aids in inverse mapping. Additionally, the two-step learning method and fully connected layer dropout operations in [4] was discussed in depth.

Some key items to note are:

- Collecting the required amount of training data is very difficult for microwave designs because the most accurate electromagnetic simulators are extremely compute intensive.
- 2) Many of the AI topological design papers involve training a model which quickly evaluates circuit performance with very little compute, this model is then used to guide an evolutionary algorithm which explores the design space.
- 3) This project takes the reverse approach of generating a model which directly generates the filter design, instead of using an evolutionary algorithm to search through many different designs every time the user asks for a new filter synthesis.

B. Learn and implement the standard closed form filter synthesis methods

One of the standard methods used for filter synthesis involves calculating a table of values called G-coefficients which define a prototype filter. It is called a prototype filter because its cuttoff frequency is 1 rad/sec. The G-coefficients must be specifically scaled and altered to achieve a different cuttoff frequency and/or to switch filter response shape (e.g. low pass, high pass, bandpass). Different families of filters can be synthesized by using a wide range of complicated recursive formulas. The Butterworth and Chebyshev filter styles are common place in the filter synthesis world.

Why this matters: Many of the existing ML microwave design papers discuss performance of circuits the AI generates, but the authors do not analyze the results against standard methods. For example, in [4] they discuss how amazing it is the AI designed an electrically small patch antenna, but they don't mention antenna efficiency or that electrically small antennas is an entire field of study which has defined upper bound limits of antenna performance for a specific size. [4] is one of the best ML microwave papers as of 2025, but it severely lacks sufficient EE background.

Below the method of computing Butterworth and Chebyshev filter coefficients is described.

For standard filter synthesis there are several commonly used designs which dictate where the poles and zeros of the filter lie. Butterworth and Chebyshev filters are common and are synthesized as follows.

- Butterworth Filter Synthesis

$$g_r = 2\sin\left(\frac{(2r-1)\pi}{2n}\right), \quad r = 1,\dots,n$$
 (1)

— Chebyshev Filter Synthesis

- Insertion Loss Ripple

$$L_{A_dB} = -10 \cdot \log \left(1 - \left(10^{\frac{-L_{T_dB}}{10}} \right) \right)$$
 (2)

- Ripple Factor

$$\epsilon = \sqrt{10^{\frac{L_A dB}{10}} - 1} \tag{3}$$

— coefficient for input port is always 1

$$g_0 = 1 \tag{4}$$

- First Circuit Element

$$g_1 = \frac{2}{\eta} \sin\left(\frac{\pi}{2n}\right) \tag{5}$$

$$\eta = \sinh\left[\frac{1}{n}\sinh^{-1}\left(\frac{1}{\epsilon}\right)\right]$$
(6)

- Subsequent Elements

$$g_r g_{r+1} = \frac{4 \sin\left(\frac{(2r-1)\pi}{2n}\right) \sin\left(\frac{(2r+1)\pi}{2n}\right)}{\eta^2 + \sin^2\left(\frac{r\pi}{n}\right)}, \quad r = 1, 2, \dots, n-1$$
(7)

- Load for Odd Order Filters

$$g_{\text{load}} = 1 \tag{8}$$

— Load for Even Order Filters From:

$$|S_{21}(0)|^2 = \frac{4g_{n+1}}{(g_n+1)^2} = \frac{1}{1+\epsilon^2}$$
 (9)

Solve for g_{n+1} :

- For $S_{11}(0) \ge 0$:

$$g_{n+1} = \frac{\left(\epsilon + \sqrt{1 + \epsilon^2}\right)^2}{1} \tag{10}$$

- For $S_{11}(0) \le 0$:

$$g_{n+1} = \frac{1}{\left(\epsilon + \sqrt{1 + \epsilon^2}\right)^2} \tag{11}$$

C. Develop method to quickly evaluate circuit performance AND verify against commercial circuit solvers

A significant amount of time went into writing the code to compute the G-coefficients and to scale the values to get a standard filter design.

I wrote a program that calculates the G-coefficients for Butterworth and Chebyshev filters using the recursive equations above. I also wrote the scripts that scale those G-coefficients to get electrical component values for inductors and capacitors. Finally, I wrote a circuit simulator that has been checked against actual circuit simulator to verify accuracy. Figure 4 shows the G-coefficients calculated for Chebyshev filter prototypes of different orders.

I also wrote scripts which simulates the filter circuit to measure its frequency response. The results of my circuit

Chebyshev Lowpass Prototype Coef. (0.05dB ripple)												
												g_load
N	g_1	g_2	g_3	g_4	g_5	g_6	g_7	g_8	g_9	g_10	g_11	shunt first
1	0.2152	1.0000										
2	0.6923	0.5585	1.2396									0.8067
3	0.8794	1.1132	0.8794	1.0000								
4	0.9588	1.2970	1.6078	0.7734	1.2396							0.8067
5	0.9984	1.3745	1.8283	1.3745	0.9984	1.0000						
6	1.0208	1.4141	1.9183	1.5475	1.7529	0.8235	1.2396					0.8067
7	1.0346	1.4369	1.9637	1.6162	1.9637	1.4369	1.0346	1.0000				
8	1.0436	1.4514	1.9899	1.6503	2.0457	1.6053	1.7992	0.8419	1.2396			0.8067
9	1.0499	1.4611	2.0065	1.6698	2.0858	1.6698	2.0065	1.4611	1.0499	1.0000		
10	1.0544	1.4679	2.0177	1.6820	2.1085	1.7009	2.0851	1.6277	1.8197	0.8506	1.2396	0.8067
			_									
	Chebyshev Lowpass Prototype Coef. (10 dB ripple)											
				nebysn	ev Lowp	oass Pro	totype	Coef.	10 dB r	ipple)		
				•							. 11	g_load
N	g_1	g_2	g_3	g_4	ev Lowp	g_6	g_7	Coef. (8_9	g_10	g_11	g_load shunt first
1	0.6667	1.0000	g_3	•							g_11	shunt first
1 2	0.6667 1.3601	1.0000 0.7066	g_3 1.9250	g_4							g_11	
1 2 3	0.6667 1.3601 1.5529	1.0000 0.7066 1.1058	g_3 1.9250 1.5529	g_4 1.0000	g_5						g_11	shunt first
1 2 3 4	0.6667 1.3601 1.5529 1.6269	1.0000 0.7066 1.1058 1.2051	g_3 1.9250 1.5529 2.3198	g_4 1.0000 0.8452	g_5	g_6					g_11	shunt first
1 2 3	0.6667 1.3601 1.5529 1.6269 1.6625	1.0000 0.7066 1.1058 1.2051 1.2436	g_3 1.9250 1.5529 2.3198 2.4956	1.0000 0.8452 1.2436	g_5 1.9250 1.6625	g_6	g_7				g_11	0.5195 0.5195
1 2 3 4	0.6667 1.3601 1.5529 1.6269	1.0000 0.7066 1.1058 1.2051	g_3 1.9250 1.5529 2.3198	g_4 1.0000 0.8452	g_5	g_6					g_11	shunt first
1 2 3 4 5	0.6667 1.3601 1.5529 1.6269 1.6625	1.0000 0.7066 1.1058 1.2051 1.2436	g_3 1.9250 1.5529 2.3198 2.4956	1.0000 0.8452 1.2436	g_5 1.9250 1.6625	g_6	g_7		g_9		g_11	0.5195 0.5195
1 2 3 4 5	0.6667 1.3601 1.5529 1.6269 1.6625 1.6821	1.0000 0.7066 1.1058 1.2051 1.2436 1.2626	g_3 1.9250 1.5529 2.3198 2.4956 2.5617	1.0000 0.8452 1.2436 1.3308	g_5 1.9250 1.6625 2.4304	g_6 1.0000 0.8738	8_7 1.9250	g_8	g_9	g_10	g_11	0.5195 0.5195
1 2 3 4 5 6	0.6667 1.3601 1.5529 1.6269 1.6625 1.6821 1.6940	1.0000 0.7066 1.1058 1.2051 1.2436 1.2626 1.2733	1.9250 1.5529 2.3198 2.4956 2.5617 2.5940	1.0000 0.8452 1.2436 1.3308 1.3627	1.9250 1.6625 2.4304 2.5940	1.0000 0.8738 1.2733	8_7 1.9250 1.6940	g_8	g_9	g_10		0.5195 0.5195 0.5195
1 2 3 4 5 6 7	0.6667 1.3601 1.5529 1.6269 1.6625 1.6821 1.6940 1.7019	1.0000 0.7066 1.1058 1.2051 1.2436 1.2626 1.2733 1.2801	1.9250 1.5529 2.3198 2.4956 2.5617 2.5940 2.6123	1.0000 0.8452 1.2436 1.3308 1.3627	1.9250 1.6625 2.4304 2.5940 2.6527	1.0000 0.8738 1.2733 1.3571	1.9250 1.6940 2.4641	1.0000 0.8841	g_9 1.9250	g_10		0.5195 0.5195 0.5195

Fig. 1. Computed Chebyshev G-coefficients. These values have been verified against other sources.

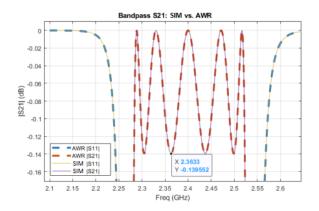


Fig. 2. S21 of a bandpass filter simulated with AWR and my own script. Notice the very zoomed in vertical axis to accurately measure the filter passband ripple. My circuit simulator lines up exactly with the commercial solver.

simulator were compared against a commercial microwave circuit simulator (Microwave Office AWR). Figure 5 shows that my simulated filter performance exactly aligns with the commercial solver. Notice the precise alignment between the commercial solver (AWR) and my own circuit simulator script. It was critical to write the circuit solver so that it could run on the GPU for fast simulation time and so that an ML model training on the GPU could easily differentiate through the weights of the circuit solver. This allows the training ML model to have its loss calculated off of the error of circuit element values OR the difference in the input frequency response versus the generated circuit's simulated frequency response. The difference between these methods is further discussed below.

D. Generate Training Data

I wrote a full set of functions which allowed me to generate filters with different parameters. I used Latin Hyper Cube sampling to generate a random set of training data, validation data, and testing data. I also added the ability to introduce varying levels of component variation in the generated filter designs.

I used this to generate three categories of training data that is used in my training (more about training steps in next section.)

- Perfect Chebyshev Filter Designs (no random component variation)
- 2) Chebyshev filters with varying level of random component variation
- 3) Completely randomly generated component values

For the Chebyshev family of filters there are three main parameters: Center Frequency, Passband Ripple (dB), and Fractional Bandwidth. The Hyper Cube Sampling ensures that a random sample is taken in each section of this 3-dimensional space. LHCS has the advantage of better coverage for fewer samples than using the Monte-Carlo sampling approach.

TABLE I DESCRIPTION OF TRAINING DATA STYLES

Function	Description
gen_cheby_data()	Ideal Chebyshev filters with no component variation; serves as baseline.
<pre>gen_cheby_data(, comp_var=0.1)</pre>	Chebyshev filters with $\pm 10\%$ component variation for moderate fabrication noise.
<pre>gen_cheby_data(, comp_var=0.2)</pre>	Chebyshev filters with ±20% variation to represent higher manufacturing variation.
<pre>gen_cheby_data(, comp_var=0.5)</pre>	Chebyshev filters with ±50% variation to simulate extreme uncertainty.
gen_rand_data()	Filters with fully random component values $(10^{-14}-10^{-6})$, unconstrained by a prototype.

This data was saved to files using the pickle python package for quick loading into a workspace for training.

III. EVALUATION OF METHOD

This paper demostrates that MLP based models can perform extremely well at circuit design of analog filters with arbitrarly defined frequency responsess.

IV. PROBLEM STATEMENT

My class presentation covered the recent developments in RF and microwave circuit design methods using AI. The basics of the RF design process were explained, and the potential for limited AI assistance at each step was briefly discussed [1]. Next, the distinction between parameter optimization and topological design was made. [2] was used to demonstrate AI's ability to balance the many parameters required in an RF design whose topology is fixed. Then, the topological AI design algorithms described in [3], [4], [5] were described and methods compared. Particular attention was paid to the wavelet transform used in [3] and how it aids in inverse mapping.

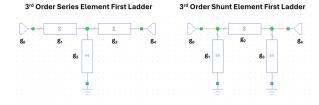


Fig. 3. 3rd order ladder representation: comes in series element first or shunt element first.

Additionally, the two-step learning method and fully connected layer dropout operations in [4] was discussed in depth.

My class project on the other hand involves training my own AI model which takes in circuit performance as an input (i.e. frequency vs. magnitude/phase values) and generates the circuit element values in a ladder topology to reach the desired performance. With a ladder topology circuit design (of high enough order) almost any circuit response can be generated, but choosing each circuit element's type and location in the network is complex. This is a great application of AI because it is easy to calculate the performance of a design, but difficult to determine which values are needed to achieve the circuit characteristics that are desired. The values that are determined by the AI can be compared with standard human design methods such as filter prototypes and scaling using G-coefficients.

V. BACKGROUND

Understanding the standard methods designers use for filter synthesis is critical to evaluating whether using AI for filter design is actually an advantage. The basics of filter synthesis is described below.

In analog filter design the unbalanced-ladder topology for circuit design is often used. Figure 3 shows two versions of a 3rd order ladder circuit.

VI. PROJECT PROGRESS

I wrote a program that calculates the G-coefficients for Butterworth and Chebyshev filters using the recursive equations above. I also wrote the scripts that scale those G-coefficients to get electrical component values for inductors and capacitors. Finally, I wrote a circuit simulator that has been checked against actual circuit simulator to verify accuracy. Figure 4 shows the G-coefficients calculated for Chebyshev filter prototypes of different orders.

I also wrote scripts which simulates the filter circuit to measure its frequency response. The results of my circuit simulator were compared against a commercial microwave circuit simulator (Microwave Office AWR). Figure 5 shows that my simulated filter performance exactly aligns with the commercial solver.

These scripts will be used to generate the training data and the circuit simulator will also be used to evaluate model performance.

	Chebyshev Lowpass Prototype Coef. (0.05dB ripple)											
N	g_1	g_2	g 3	g_4	g_5	g_6	g_7	g 8	g_9	g_10	g_11	g_load shunt first
1	0.2152	1.0000		02	0_					0_		
2	0.6923	0.5585	1.2396									0.8067
3	0.8794	1.1132	0.8794	1.0000								
4	0.9588	1.2970	1.6078	0.7734	1.2396							0.8067
5	0.9984	1.3745	1.8283	1.3745	0.9984	1.0000						
6	1.0208	1.4141	1.9183	1.5475	1.7529	0.8235	1.2396					0.8067
7	1.0346	1.4369	1.9637	1.6162	1.9637	1.4369	1.0346	1.0000				
8	1.0436	1.4514	1.9899	1.6503	2.0457	1.6053	1.7992	0.8419	1.2396			0.8067
9	1.0499	1.4611	2.0065	1.6698	2.0858	1.6698	2.0065	1.4611	1.0499	1.0000		
10	1.0544	1.4679	2.0177	1.6820	2.1085	1.7009	2.0851	1.6277	1.8197	0.8506	1.2396	0.8067
			_									
	Chebyshev Lowpass Prototype Coef. (10 dB ripple)											
N	g_1	g_2	g_3	g_4	g_5	g_6	g_7	g_8	g_9	g_10	g_11	g_load shunt first
1	0.6667	1.0000	8_3	8_4	8_5	8_0	8_/	8_o	8_9	g_10	g_11	shunt hist
2	1.3601	0.7066	1.9250									0.5195
3	1.5529	1.1058	1.5529	1.0000								0.5195
4	1.6269	1.2051	2.3198	0.8452	1.9250							0.5195
5	1.6625	1.2436	2.4956	1.2436	1.6625	1.0000						0.5195
	1.6821	1.2436	2.4956	1.3308	2.4304	0.8738	1.9250					0.5195
6	1.6940	1.2626	2.5940	1.3308	2.4304	1.2733	1.6940	1.0000				0.5195
7	1.7019	1.2733	2.5940	1.3627	2.6527	1.3571	2.4641	0.8841				0.5195
8												0.5195
9	1.7072	1.2846	2.6238	1.3867	2.6805	1.3867	2.6238	1.2846				
10	1.7111	1.2877	2.6315	1.3921	2.6960	1.4006	2.6797	1.3671	2.4789	0.8889	1.9250	0.5195

Fig. 4. Computed Chebyshev G-coefficients. These values have been verified against other sources.

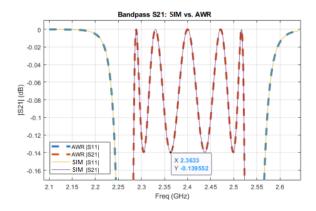


Fig. 5. S21 of a bandpass filter simulated with AWR and my own script. Notice the very zoomed in vertical axis to accurately measure the filter passband ripple. My circuit simulator lines up exactly with the commercial solver.

VII. PROJECT SCOPE

For this project a ladder topology of 6 circuit elements is chosen over a 100x frequency bandwidth. Each circuit element has the following options: open, short, resistor, capacitor, inductor, series LC, or shunt LC. Many randomly generated circuit designs will be generated and performance calculated over the frequency range. This data will be used to train the model. The performance of the model will be based on how well it can design a circuit to match desired performance.

VIII. CONCLUSION

A program has been written that can calculate the G-coefficients for Butterworth and Chebyshev filters. More scripts have been written that can scale those G-coefficients to get electrical component values for inductors and capacitors. Finally, I wrote a circuit simulator that has been checked against actual circuit simulator to verify accuracy. These

scripts will be used to generate the training data and the circuit simulator will also be used to evaluate model performance.

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