

Implementation of Infinite impulse Response Filter using Parvartya Yojayet in DSP System

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ABSTRACT

Dsp filter use in the communication and computation. Optimization methods are used to design top-quality digital IIR filter. IIR filter is designed by the Ancient Vedic mathematics. IIR filter design first time with the help of vedic method technique, Which is parvartya method . here paper a fast method for filter designing is depend lying on very old Vedic mathematics, is completed with the help of Paravartya Yojayet method, which is very efficient method for all algebraic method . This method is applicable for all types of DSP operation. This algorithm is implement in MATLAB and all the operation is performed in GUI. The parvartya method is more efficient method. With the help of using this method the result like as the speed of operation and reducing noise in 85-90%. This algorithm operates in the concept of Vedic multiplier.

Keywords

DSP, IIR, FIR, Parvartya, Vedic Mathematics, GUI

1. INTRODUCTION

DSP operation is the compassion of the mobile communication and satellite communication system. The convolution plays a preciously role in Digital Signal Processing and Image Processing.. DSP operation is the compassion of the mobile communication and satellite communication system.. It is used for crafty of dsp Filter obtain clear of the harmful constituent from signal The undesirable component is generate the noise in signal so the process could now not work in correctly and don't cater correct information of the approach. Filter is filtering these noise sign and generate a noise much less signal these noise less signal is provide correct expertise of the system. One other- distinctive digital filters are used in many fields like that conversation, picture[3] field, scientific field and so on. These digital filter furnish the extra accuracy and beef up the in fluence of sign. a couple of approaches to be had for the design of filters. Digital IIR filters are used many utility comparable to excessive-speed and low-energy conversation transceivers, they are additionally used in repeatedly employed as accustom designed digital block. IIR Filters are endless response filter, they have desire retort of limitless duration. IIR filter are recursive filter in which suggestions is present from output part to the input facet. IIR filter response is depend on the previous output samples, gift and previous input samples. It's used in a lot software corresponding to excessive-speed and low-power communication transceivers techniques. implementation is design on Matlab with GUI, which is user friendly and easy to use[6].

2. VEDIC MATHEMATICS

Vedic arithmetic is an ancient quick calculation mathematics technique which is taken from historical ancient book of

wisdom. Vedic mathematics is an ancient Vedic mathematics which provides the unique technique of intellectual calculation with the help of simple rules and principles. Veda rediscover by the holiness Shree Bharti Krishna Tirtha Ji Maharaj 1884 to 1960 all Vedic mathematics is based on 16- Sutra and 16- up-sutra after broadly research in Atharva Veda [4]. It is dwpend on 16 sutras which manage poles apart twigs of arithmetic i.e. [7]. Ekadhikena Purvena – By one more than the previous one.

1. Nikhilam Navatascaramap Dasatah – All from 9 and last from 10.
2. Urdhva-Tiryagbhyam – Vertically and crosswise.
3. Paravartya Yojayet – Transpose and adjust.
4. Shunyam Samyasamuccaye – When the sum is the same that sum is zero.
5. (Anurupye) Shunyamanyat – If one is in ratio, the other is zero.
6. Sankalana-Vyavakalanabhyam – By addition and by subtraction.
7. Puranapuranyam – By the completion or no completion.
8. Calana-Kalanabhyam – Differences and Similarities.
9. Yaavadunam – Whatever the extent of its deficiency.
10. Vyastisamanstih – Part and Whole.
11. Sesanyankena Caramena – The remainders by the last digit.
12. Sopantyadvayamantyam – The ultimate and twice the penultimate.
13. Ekanyunena Purvena – By one less than the previous one.
14. Gunakasamuccayah – The product of the sum is equal to the sum of the product.
15. Gunakasamuccayah – The factors of the sum is equal to the sum of the factors.

3. PROPOSED ALGORITHM

Paravartya Yojayet

Paravartya approach “Transpose and observe”. This sutra can also be higher understood through examples. (The reason beneath can be complicated on the preliminary reading, but after just a few examples the notion will probably be clear and the division of algebraic expressions can be a simple affair[6].

I. Division of an algebraic expression via yet another algebraic expression

Examples: $12x^2-8x-32 / x-2$

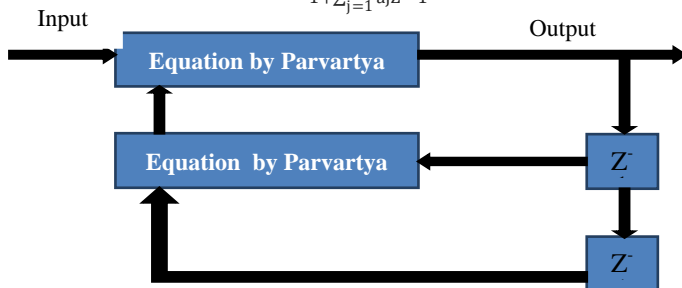
Solution:

| Column 1 | | Column 2 | |
|----------|--------|----------|-----|
| $x-2$ | x | I | r |
| 2 | 12 | -8 | -32 |
| | | +24 | +32 |
| | 12x+16 | 0 | |

Explanation: in the over case $12x^2-8x-32$ is the dividend; $x-2$ is the divisor. In the backside line of column 2 of the solution are the quotient ($12x+16$) and the remainder (zero). The highest row in italics represents the powers of x within the quotient. From the divisor $x-2$ take the term (-2) , next to the first term (x), change its sign (transpose) and make contact with it the ‘Dividend Multiplier’ position it under the divisor in column 1 under Column 2, in the prime row write the power of x , which might be one, not up to the powers of x in the dividend. For instance the powers of x within the

dividend are x^2 , x and constant. In Column 2 write them as x , 1 (steady) and r (representing remainder). Then write the Coefficients of x^2 , x and the consistent time period in column 2 under x , 1 and r . For that reason write the coefficients 12, -eight and -32, in the dividend beneath x , 1 and r in column 2

$$H(z) = \frac{\sum_{i=1}^p b_i z^{-i}}{1 + \sum_{j=1}^q a_j z^{-j}}$$



4. IMPLEMENTATION & RESULT

The pole-zero plot used to determine the stability and visualize the connection between the Frequency domain and Z-domain. The set of value is called the region of convergence. The frequency response $H(e^{j\omega})$ is acquired from the switch operate $H(z)$, by using evaluating the transfer function at precise values of $z = e^{j\omega}$. Considering, the frequency response is periodic with period 2π , we need to review it over one period, equivalent to $-\pi < \omega < \pi$. If we substitute these values of ω in $z = e^{j\omega}$, values of z lie on the unit circle and range from $z = -1$ all the method around and back to the point $z = -1$.

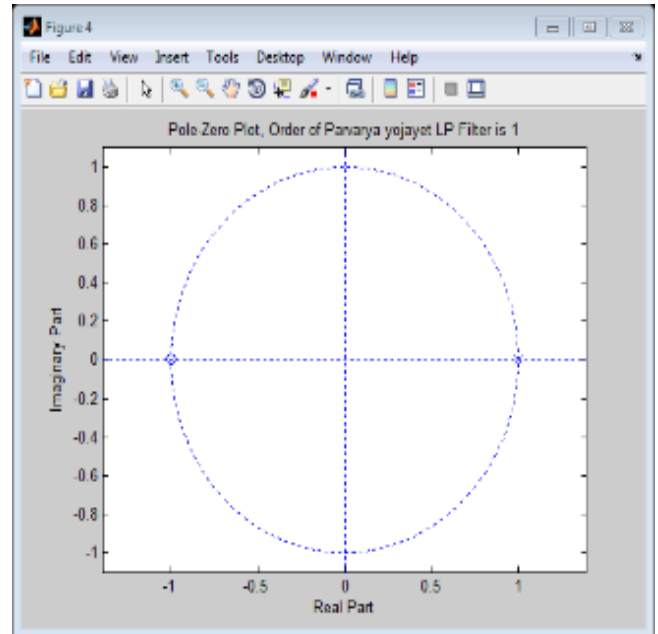


Fig.1 proposed in zero & poles graph by Parvartya Method

Infinite impulse response (IIR) filters are another class of traditional filters. A k th order IIR filter is described by the difference equation,

$$y(n) = - \sum_{k=1}^K a_k y(n-k) + \sum_{k=0}^K b_k x(n-k)$$

where each output sample depends on present and past input samples, as well as output samples.

This is because the impulse response is directly specified in the design process. Making the filter boast right-left balance is all is necessary This is not the case with IIR filters, since the recursion coefficient are what is specified, not the impulse response. The impulse response of a recursive filter is not symmetrical between the right and left, and therefore has a nonlinear phase.

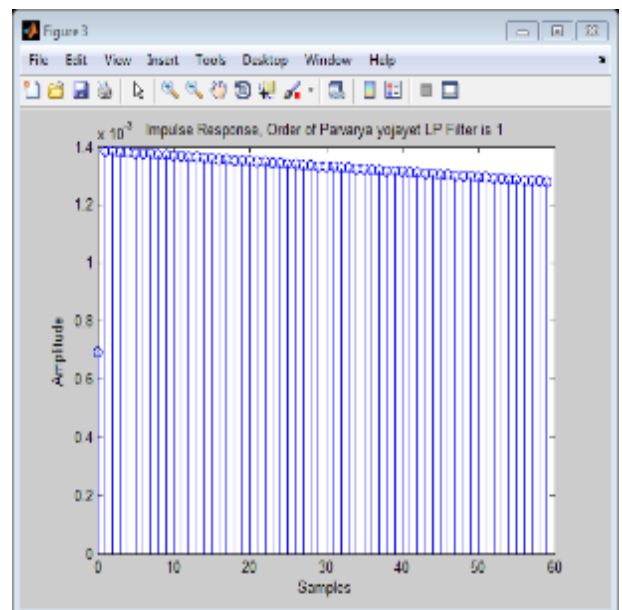


Fig.2 proposed in impulse response by Parvartya Method

For linear phase, it is sufficient to have that property only in the passband(s) of the filter, where $|A(\omega)|$ has relatively large values. Therefore both magnitude and phase graphs (Bode plots) are customarily used to examine a filter's linearity. A "linear" phase grid might have discontinuities of π and/or 2π radians. The smaller ones happen where $A(\omega)$ changes sign. Since $|A(\omega)|$ cannot be negative, the changes are reflected in the phase plot. The 2π discontinuities happen because of plotting the principal value of ωt instead of the actual value

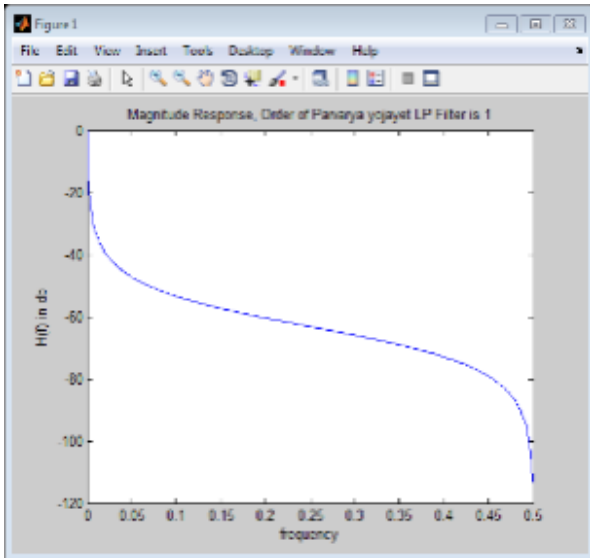


Fig.3 In phase response by Parvartya Method

Magnitude response of low pass filter in terms of frequency $f(H)$ with respect of time. The results obtained for elliptic, Butterworth, sankalana and parvartya filters are summarized and are compared with pole zero, magnitude response, impulse response and phase response

- When compared with optimization algorithm Vedic mathematic provides better magnitude response for all types of digital IIR filters
- In addition to above the parvartya was tested for the robustness by performing 100 independent runs with random variation. The minimum, maximum, average and standard deviation values obtained have been summarized in a robust and effective algorithm for the design of digital IIR filters of better responses and lower order

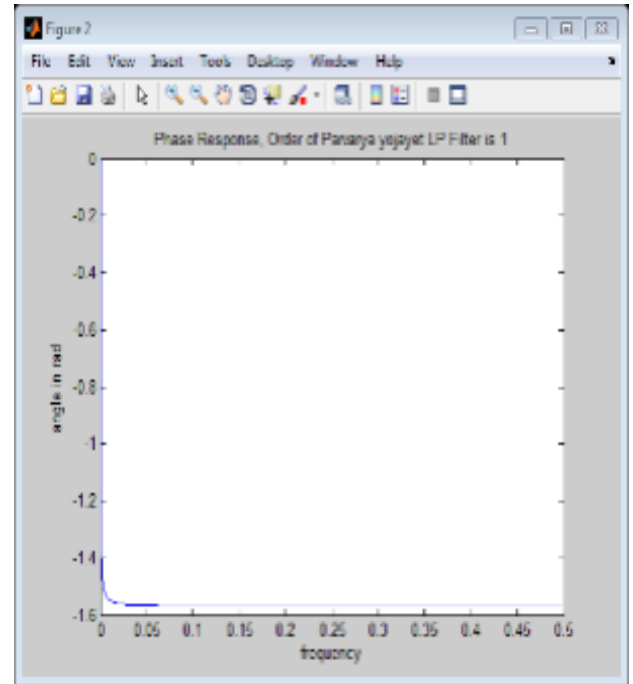


Fig.4 proposed in Phase response by Parvartya Method

Table 1 shows numerator and denominator coefficient

| METHOD | NUMERATOR COEFFICIENT | DENOMINATOR COEFFICIENT |
|-------------|------------------------|-------------------------|
| SANKALANA | 0.031623 0.031623 | 1 -0.93675 |
| BUTTERWORTH | 08286 08286 | 1-0.83428 |
| PARVARTYA | .00069409 .00069409 | 1 0.99861 |
| ELLIPTIC | 0.99982 -0.99982 | 1 -0.99964 |

5. CONCLUSION

This paper proposes a vedic approach for optimization of digital IIR filters considering multiple conflicting objectives. lying on the root of results obtained for the design of digital IIR filter, it can be concluded that vedic is a robust algorithm and possesses the capacity for the local tuning of the solutions. parvarta can design a digital filter of any type, while the lowest order of the filter is achieved; it can design the IIR filter with better magnitude and phase performance. Concluding, Simulation studies show that the proposed method is accurate, robust and an efficient optimizer for digital IIR filter design during the comparison the parvartya method is best as comparative to the sankalana method by using this method the result like as speed of operation and reduce noise.

6. REFERENCES

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