

Design and Optimization of Fuzzy-based Digital FIR Kernel Filter for Real-Time Control Applications

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Abstract— 21st century farmers are prone to the dangers of climatic changes resulting from intense temperature and relative humidity in their nursery beds of fruits and vegetables. For optimum growth of those fruits and vegetables without having to visit their farms often, digital FIR filters that control the Low pass frequency threshold at 0.3π rad/time span , blocking the high pass frequency signal at 0.7π rad/time span as well as control the high pass frequency threshold at 0.7π rad/time span , blocking the low pass frequency signal at 0.3π rad/time span were designed, employing optimization methods such as Least square method, Parks McClellan Algorithm and window methods setting the filter length of 200 samples; galvanized with convolution techniques. It is interesting to know that the digital FIR filters with built-in temperature and humidity sensors control fuzzy filtering (using Gaussian membership function with 6 base rules) in the greenhouse nursery bed; hence the name Fuzzy Digital FIR kernel Filter for Real-time control Applications. The temperature and relative humidity of the next-generation greenhouse been controlled are in the range of 15-30°C and 50-90% respectively. All through the design, MATLAB fuzzy logic, DSP, Filter tool boxes and MATLAB File editor (M-File) were utilized for the simulation, modeling and generation of C-programming codes.

Keywords—FIR Filter, MATLAB, Greenhouse, Low pass, high pass, Convolution, Fuzzy logic, Temperature, Humidity.

I. INTRODUCTION

The idea of Fuzzy modeling and control is deployed in the design of Digital FIR Filter for Low pass and high pass Filtering of a stipulated threshold of frequency passband while the unrequired threshold is blocked. Fuzzy control is a type of control that makes use of fuzzy set theory. Fuzzy control has two distinct advantages. Fuzzy control, for starters, provides a novel framework for executing control principles that are frequently dependent on knowledge or linguistic descriptions. Second, fuzzy control [7] provides an alternative paradigm for facilitating the design of non-linear controllers for plants that rely on generally uncertain control, which is difficult to link to traditional non-linear control theory. The possibility of improving the accuracy of climate control in next generation greenhouse cultivation by way of model based filtering [1], where temperature and humidity is taken cognizant of for automatic real-time control of nursery cultivated with fruits and vegetables, it is ideal to set up a required threshold for frequency responses when average daily temperature and humidity are set below or above normal to control its filtering action for optimum growth of plants in the nursery bed by use of sensors(temperature and humidity sensors) aimed at filtering out a high frequency noise when a low frequency signal is passed and a low frequency noise when a high frequency signal is passed without having to visit the nursery often. This can be done by installing greenhouse controllers at specified time interval [2], hence the need to design a fuzzy digital filter for the control. Common filtering objectives are to improve the quality of a signal, to extract information from signal or to separate two or more signals previously combined [8].

The digital filter is a discrete system, and it can do a series of mathematic processing to the input signal, and therefore obtain the desired information from the input signal [6]. Digital filter is a linear time invariant system (LTI) which does not vary with time. Digital filters have high accuracy, easy to simulate and design, flexible than analog filters [3]

II. REVIEW OF RELATED WORKS

Two scenarios indicative of modern greenhouse practice were used to test the filter performance [1] in terms of monitoring accuracy of average temperature and humidity inside a greenhouse compartment. Temperature and absolute humidity were measured with a sensor grid inside a big and small greenhouse compartment with a 5-min sample resolution. Sensor grids that were spatially scattered were used in both circumstances. The instances were recreated by using a single sensor's readout as the filtered signal. The reference data was obtained by taking the sensor grid's spatial average. The root mean squared error (RMSE) between the filtered data and the reference data was used to measure monitoring accuracy. The root mean squared errors (RMSEs) of the unfiltered signals (coming from single sensors) were 0.43°C and 0.48 gm⁻³ for greenhouse A and 0.80°C and 0.64 gm⁻³ for greenhouse B, respectively, when compared to the average over all sensors as reference signal. By altering the time step and the magnitude of state noise covariance, the impact of the filter settings was investigated. The robustness of these results in terms of reference signal accuracy was also verified by continually deleting one sensor from the grid. The significance of noise assumptions was checked by substituting manufactured white noise with the same covariance for the observed noise in state and output. By arbitrarily increasing the amounts of observed model and sensor errors, the relationship between model accuracy and monitoring accuracy was projected.

Temperature and Humidity of automatic-computer controlled greenhouse nursery bed was controlled by overhead ventilators, fan, heater and mister at every 10 seconds interval [2]. The generalized bell-shape (gbell) membership function was employed through the application of 'If then rules' and centroid method of defuzzification. In the design, the



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greenhouse was programmed whenever the fan or overhead ventilator are put on or close when the temperature goes above normal and the mister or heater is powered on when the humidity goes below normal and vice versa. Ten iterated base rules were formulated based on fuzzy decision with the parametric universe of discourse of range -30 to 95°C for temperature and 5 to 80% of humidity with each constituting five degree of membership; though the analysis was based on feedback control using MATLAB fuzzy logic tool box.

A greenhouse was managed effectively using wireless network to control temperature, relative humidity, soil moisture and light intensity for 4-months period (between early winters and early spring), based on fuzzy logic system of the sensorbased data input. In the greenhouse, tomatoes was optimally selected at varied temperature range of $25^{\circ}C - 27^{\circ}C$ and the humidity value is in the range of 50% to 70% [9]. The condition of the tomatoes growth was monitored based on fuzzy control system optimization criterion using Zigbee protocol. From this it was deduced that the mean night values for December were estimated at 37 °C and the relative humidity at 83.1%. Soil humidity was 55.7% and the amount of light was 3512 lux; the corresponding calorific value was 4.58 kW, the cooling value was 5.42 µm, the irrigation value was 8.99 l, the shading value was 53.7 cm and the amount of light was 15400 lux; taking into account the heat loss gain of 7.01 kW. The mean daily value for December also accumulated 7.2 °C for temperature, 82.8% for relative humidity, 50.1% for soil humidity, and the amount of light was 11422 lux; the corresponding calorific value was 4.39 kW, the cooling value was 517 µm with an estimated heat loss of 6.62 kW.

There is a prediction of sinusoidal signal with time-varying frequencies using fuzzy filtering scheme based on finite Impulse Response (FIR) filter bank using interpolation of individual filter outputs [4]. There was derivation of the filter coefficient based on the sum of squared prediction error (SSPE) for filtering performance. Five Gaussian membership function with frequency range of 49-51 Hz was employed. The sinusoidal input signal with time-varying frequencies, x(n) was deployed at 1.67KHz, predicted at every period of 2 seconds and If Then Rules used to evaluate the frequency range of the filters as follow: If f(n) < 49.625Hz Then switch to filter designed at 49.5 Hz,

If 49.625Hz < f(n) < 49.875Hz Then switch to filter designed at 49.75 Hz,

If 49.875Hz < f(n) < 50.125Hz Then switch to filter designed at 50 Hz,

If 50.125Hz < f(n) < 50.375Hz Then switch to filter designed at 50.25 Hz,

If $f(n) \ge 50.375$ Hz Then switch to filter designed at 50.5 Hz.

A closed-form linear phase FIR filter design [5] concurrently possessing high efficiency and excellent transfer characteristic by employing Fourier spectrum of a convolution window was proposed. There was increase in the FIR filter order to 198 (i.e., filter length, L=199 and N=100) and specify passband cutoff frequency $w_p = 0.5\pi rad/s$, transition bandwidth 0.02 π (accordingly, M = 26), $w_c(n)$ is derived by convolving an N-length Hamming window with an N-length rectangular window.

III. METHODOLOGY

There are two phases of this design methodology: Digital FIR Filter design, and fuzzy based modeling. First, in the design methodology of both Low pass and high pass Digital FIR Filter, the filter coefficients were generated using first principle and using filter design tool box and signal processing tool box. This is done by allowing a finite impulse response of length 3 units involving time span (n), input sequence, X (n), and output sequence y (n). The input has been assumed to be a casual sequence with the first non-zero sample occurring at n=0, and for calculating y (0) and y (1), set x (-1) = x (-2) = 0.

The signal responses of impulse and frequency responses of low pass and high pass Digital FIR Filter were generated using park-McClellan Algorithm and other window techniques for Digital FIR Filter designs such as rectangular, Kaiser, hamming, and Han where filter design tool box, signal processing tool box and filter visualization and tool box were employed to achieve the same design for easy implementation of fuzzy decision.

3.1. Real-Time Digital FIR Filters Design

3.1.1. Design objectives

- To design a Low pass discrete FIR Filter that will allow a low frequency component and block a high frequency component signals Using MATLAB Software.
- To design a high pass discrete FIR Filter that will allow a high frequency component and block a low frequency component signals Using MATLAB Software.
- 3.1.2 Design Techniques for low pass and high pass Digital FIR Filter.

For simplicity, we assume the filter to be an FIR filter of length 3 with an impulse response:

 $h(0) = h(2) = \beta_0, h(1) = \beta_1$

Y (n) = $\sum_{k=N_1}^{N_2} h(k)x(n-k)$, $N_1 < N_2$. The digital filtering is performed using the difference equation:

$$y(n)=h(0)x(n)+h(1)x(n-1)+h(2)x(n-2)$$
 (1)

 $y(n) = \beta_0 x(n) + \beta_1 x(n-1) + \beta_0 x(n-2)$ (2)

Where y(n) & x(n) represent the output and input sequences respectively. Also, from our design, we chose the filter coefficients β_0 & β_1 such that the output of the filter becomes the cosine sequence with frequency 0.7π rad/time span. Now, using equation of the frequency response.

$$H[e^{j\omega}] = \sum_{K=N_1}^{N_2} h(k)e^{-j\omega k}$$
(3)

Which is seen to be a polynomial $ine^{-j\omega}$. The frequency response of the above filter is given by:

$$H[e^{j\omega}] = h(0)e^{-j\omega0} + h(1)e^{-j\omega1} + h(2)e^{-j\omega2}$$
(4)
$$H[e^{j\omega}] = 0 + 0 e^{-j\omega} + 0 e^{-j\omega2}$$

$$H[e^{j\omega}] = \beta_0 + \beta_1 e^{-j\omega} + \beta_0 e^{-j\omega}$$

$$H[e^{j\omega}] = \beta_0 + \beta_0 e^{-j\omega^2} + \beta_1 e^{-j\omega} = \beta_0 (1 + e^{-j\omega^2}) + \beta_1 e^{-j\omega}$$
(5)
But, $\cos\omega = \frac{1}{2} (e^{j\omega} + e^{-j\omega})$.

$$2\cos \omega = e^{j\omega} + e^{-j\omega}$$

$$2\cos \omega = e^{j\omega} + e^{-j\omega}$$
(6)
$$H[e^{j\omega}] = \beta_0(e^{j\omega} + e^{-j\omega})e^{-j\omega} + \beta_1 e^{-j\omega} = \beta_0(2\cos\omega)e^{-j\omega} + \beta_0(2\cos\omega)e^{-j\omega} + \beta_0(2\cos\omega)e^{-j\omega} + \beta_0(2\cos\omega)e^{-j\omega} + \beta_0(2\cos\omega)e^{-j\omega} = \beta_0(2\cos\omega)e^{-j\omega} + \beta_0(2\cos\omega)e^{-j\omega} +$$

 $(2\beta_0\cos\omega + \beta_1)e^{-j\omega}$

The magnitude & phase functions of the filter are $|H[e^{j\omega}]| = |2\beta_0 \cos\omega + \beta_1|$

(7)



 $\theta(\omega) = -\omega$

Step I: In order to stop the low frequency component from appearing at the output of the filter, the magnitude function at $\omega_s = 0.7 \pi$ rad/time span should be set to Zero. Similarly, to pass the low frequency component within any attenuation, we need to ensure to ensure that the magnitude function at $\omega_p = 0.3 \pi$ rad/ time span set to 1.

Thus, the two conditions that must be satisfied are:

 $2\beta_0\cos(0.7) + \beta_1 = 0$ (8)

 $2\beta_0 \cos(0.3) + \beta_1 = 1$ (9)Or converting to degrees, we have

 $2\beta_0 \cos 40.1 + \beta_1 = 0$

(10) $2\beta_0 \cos 17.2 + \beta_1 = 1$ (11)

Solving simultaneously using Cramer's rule or determinant method, we have

$$\begin{split} \Delta_{0} &= \begin{vmatrix} 2\cos 40.1 & 1\\ 2\cos 17.2 & 1 \end{vmatrix} = 2\cos 40.1 - 2\cos 17.2 \\ &= -0.3807 \\ \Delta_{\beta_{0}} &= \begin{vmatrix} 0 & 1\\ 1 & 1 \end{vmatrix} = -1 \\ \Delta_{\beta_{1}} &= \begin{vmatrix} 2\cos 40.1 & 0\\ 2\cos 17.2 & 1 \end{vmatrix} = 2\cos 40.1 = 1.5298 \\ \beta_{0} &= \frac{\Delta_{\beta_{0}}}{\Delta_{0}} = \frac{-1}{-0.3807} = 2.6267 ; \\ \beta_{1} &= \frac{\Delta_{\beta_{1}}}{\Delta_{0}} = \frac{1.5298}{-0.3807} = -4.0184 \end{split}$$

Substituting, we have

y (n) = $\beta_0 x(n) + \beta_1 x(n-1) + \beta_0 x(n-2)$, the output y(n) = 2.6267x(n) - 4.0186x(n-1) +becomes 2.6267x(n-2)(12)OR y (n) =2.6267[x (n) + x (n-2)] - 4.0184x (n-1) and the input becomes x (n) = $\{\cos(0.3n) + \cos(0.7n)\}u(n)$ Let, n=0:

v(n) =

$$2.6267x(n) - 4.0184x(n-1) + 2.6267x(n-2)$$
 (13)

y(0) = 2.6267x(0) - 4.0184x(-1) + 2.6267x(-2); Initial conditions are x(-1) and x(-2).Note that x(0) is the first input value and not an initial condition.

n = 1: y(1) = 2.6267x(1) - 4.0184x(0) + 2.6267x(-1);n = 2: y(2) = 2.6267x(2) - 4.0184x(1) + 2.6267x(0);

Step II: From the first principle, for high pass digital filter; In order to stop the high frequency component from appearing at the output of the filter, the magnitude function at $\omega_p = 0.7$ rad/time span should be equal to 1. Similarly, to pass the high frequency component within any attenuation, we need to ensure to ensure that the magnitude function at $\omega_s = 0.3$ rad/s is equal to 0.

$$2\beta_0 \cos(0.7) + \beta_1 = 1 \tag{14}$$

(15)

$$2\beta_0\cos(0.3) + \beta_1 = 0$$

Or converting to degrees, we have

 $2\beta_0 \cos 40.1 + \beta_1 = 1$ (16)

 $2\beta_0 \cos 17.2 + \beta_1 = 0$ (17)Solving simultaneously using Cramer's rule or determinant method, we have

$$\Delta_{0} = \begin{vmatrix} 2\cos 40.1 & 1\\ 2\cos 17.2 & 1 \end{vmatrix} = 2\cos 40.1 - 2\cos 17.2$$
$$= -0.3807$$
$$\Delta_{\beta_{0}} = \begin{vmatrix} 1 & 1\\ 0 & 1 \end{vmatrix} = 1$$

$$\begin{split} \Delta_{\beta_1} &= \begin{vmatrix} 2\cos 40.1 & 1 \\ 2\cos 17.2 & 0 \end{vmatrix} = -2\cos 17.2 \\ &= -1.9106 \\ \beta_0 &= \frac{\Delta_{\beta_0}}{\Delta_0} = \frac{1}{-0.3807} = -2.6267 \\ \beta_1 &= \frac{\Delta_{\beta_1}}{\Delta_0} = \frac{-1.9106}{-0.3807} = 5.0186 \\ \text{Substituting, we have} \\ y(n) &= \beta_0 x(n) + \beta_1 x(n-1) + \beta_0 x(n-2), \text{ the output becomes:} y(n) = -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2) \\ 2.6267 x(n-2) \\ \text{OR y } (n) &= -2.6267 [x (n) + x (n-2)] + 5.0186 x (n-1) \text{ and the input becomes:} x (n) = \{\cos(0.3n) + \cos(0.7n)\}u(n) \\ \text{let, } n &= 0; \\ y(n) &= -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(0) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-1) - 2.6267 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-1) + 5.0186 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-2) + 5.0186 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-2) + 5.0186 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-2) + 5.0186 x(n-2); \\ y(0) &= -2.6267 x(n) + 5.0186 x(n-2) + 5.0186 x(n-2)$$

n = 1: y(1) = -2.6267x(1) + 5.0186x(0) - 2.6267x(-1);n = 2: y(2) = -2.6267x(2) + 5.0186x(1) - 2.6267x(0);Initial conditions are x(-1) and x(-2).

The initial conditions are set to zero to satisfy linear and time-invariant properties.

IV. SYSTEM DESIGN ANALYSIS

The main goal of this design is to pass a desired frequency component and block another frequency component. In the design, the first 200 output samples were passed in both low pass and high pass filtering; the input sequence, x(n)=cos(0.3*n)+cos(0.7*n) was used to obtain the output sequence(n)= $\beta_0 x(n) + \beta_1 x(n-1) + \beta_0 x(n-2)$ where the filtering coefficient $\beta_0 = 2.6267$ and $\beta_1 = -4.0184$ for Low pass Digital FIR filter and $\beta_0 = -2.6267$ and $\beta_1 = 5.0186$ for High pass Digital FIR filter using different design techniques, employing Parks-McClellan Algorithm in the design.

4.1 Frequency Response of the Optimized Digital FIR Filter

The frequency responses for the optimized Low pass and High pass Digital FIR filter are designed using 45-order and 52order respectively of optimized parks-McClellan Algorithm as shown in Appendix A and Appendix B. The cut-off frequencies of 0, 0.3, 0.7 and 1 were used assigning the desired amplitude of 1,1,0,0 and 0,0,1,1 for the Low pass and High Pass Digital FIR Filter respectively; generating the filter coefficient using three design tool box viz: filter design tool box(firgr), signal processing tool box(firpm) and filter visualization and tool box(fvtool) thereby allowing a Low frequency component of 0.3 and blocking a high frequency component of 0.7 for Low pass as shown in Figure 1 and allowing a High frequency component of 0.7 and blocking a low frequency component of 0.3 for High pass as shown in Figure 2

4.2. Impulse Response and convolution of Digital FIR Filters.

Appendix C and Appendix D show how to generate the impulse responses of second order Low pass and high pass Digital FIR FILTER were presented in Figure 3 and Figure 4 respectively, the two filter coefficient $\beta_{0,}$ and $\beta_{1,}$ gotten from first principle as shown in equations above; coefficient $\beta_0 =$ 2.6267 and $\beta_1 = -4.0184$ for Low pass Digital FIR filter and



 $\beta_0 = -2.6267$ and $\beta_1 = 5.0186$ for High pass Digital FIR filter as well as the input sequence, x=cos (0.3*n) +cos (0.7*n) for the first 200 samples which were filtered to produce the output sequence. In Appendix E, combined impulse response was generated and the plot shown in Figure 5. Owing to finite length of Digital FIR Filter, in Appendix F, there was convolution of the both filters in frequency domain as plotted in Figure 6 below to show their filtering actions while their direct convolution is shown in Appendix G, Figure 7.



Fig. 1. Frequency Response of the Optimized Low pass Digital FIR Filter



Fig. 2. Frequency Response of the Optimized High pass Digital FIR Filter



Fig. 3. Impulse Response of Second Order Low pass Digital FIR Filter



Fig. 4. Impulse Response of Second Order High Pass FIR Filter



Fig. 5. Combined Impulse Responses of Low pass and High pass FIR Filters.





Fig. 6. Convolution of Digital FIR Filter

4.3. Real-Time Fuzzy Control Modeling.

In the design of Low pass and high pass Digital FIR filter for the greenhouse control as shown in Figure 8-Figure 12, The fuzzy control system was built based on the inputs: Temperature and Humidity controlled by the Filter frequency response. We let the fuzzy variables Warm, Hot and Very Hot represent the Temperature Sensor and Low, Normal, High and very High represent the Humidity Sensors where the individual responses of the Digital FIR Filter for each combination of the inputs be Very Slow, Slow, Fast, Very Fast, and Extremely Fast. Gaussian membership function was used for both the linguistic input variable and the set of Output Control responses of the Digital FIR Digital filter. A defuzzifier collects the information provided by each of the rules and makes a decision from sentences defined by the inference engine about the output of models for a clear standard signal. The defuzzification priority procedure takes a weighted sum of the specified consequences of the rules according to the firing force of the rules. Fuzzy rules are a collection of linguistic statements that describe how the fuzzy controller should make a decision regarding the classification of an input or the control of an output. Fuzzy rules are always written as follows:

- 1. If (Temperature is Warm) and (Humidity is Normal) then (Frequency Response is Very Slow).
- 2. If (Temperature is Hot) and (Humidity is High) then (Frequency Response is Fast).
- 3. If (Temperature is Hot) and (Humidity is Low) then (Frequency Response is Slow).
- 4. If (Temperature is Hot) and (Humidity is Very High) then (Frequency Response is Very Fast).
- 5. If (Temperature is Very Hot) and (Humidity is High) then (Frequency Response is Extremely Fast).
- 6. If (Temperature is Very Hot) and (Humidity is Very High) then (Frequency Response is Extremely Fast).





Fig. 8. FIS Editor for Real-Time Fuzzy Digital FIR Filter





Fig. 12. Rule Editor for Real-Time Fuzzy Digital FIR Filter

As shown in table 1 below, in the analysis, temperature has universe of discourse 15 to 30° C with three degree of membership; humidity has universe of discourse 50 to 90% with four degree of membership; all the control output responses have 0 to 1 with five degree of membership for the next generation greenhouse control of fruits and vegetables



nursery; done for every 30 seconds interval using temperature and Humidity Sensors. The Digital FIR Filters operate based on the performance of these sensors.

TABLE 1.	Range and	the number	of membership	function of	f fuzzy control
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Linguistic variables	Universe of discourse	Number of Membership function	Fuzzy variable	Type of Membership Function
Temperature	15-30°C	3	Warm Hot Very Hot	Gauss mf
Humidity	50-90%	4	Low Normal High High Very	Gauss mf
Frequency Response	0-1	5	Very Slow Slow Fast Very Fast Extremely Fast	Gauss mf

4.4. Fuzzy Real-Time Implementation of Digital FIR Filter for Greenhouse Control

The block diagram of the proposed Fuzzy Digital FIR Filter for Real-time Control Applications is shown in Figure 13. M-File MATLAB editor for Gaussian membership function of Digital FIR Filter was generated as shown in Figure 14, Appendix H. For the greenhouse control as shown in Table 2, Temperature and relative humidity ranges for optimal growth of various fruit and vegetable plants in a closer typical greenhouse where most fruit and vegetable plants require a temperature of 15-30 ° C and a relative humidity of 50-90 % for up to 180 samples.



The implementation of the greenhouse filtering and control of temperature and humidity is shown in Figure 15 and Figure 16, as illustrated in Appendix I and J. In Figure 17 and Figure

http://ijses.com/ All rights reserved 18, the rule viewer and surface viewer of the Real-time Fuzzy based Digital FIR Filter for Greenhouse control of next generation fruits and vegetables nurseries where average daily temperature of 22.5°C and absolute relative humidity of 70% was achieved with sampled frequency of the greenhouse still maintained at 0.5π rads/s.



Fig. 14. Gaussian Membership Function of Digital FIR Filter

Time Span(n)	Temperature(°C)	Relative Humidity (%)				
30	15.0000	48.0959				
60	18.0000	54.5246				
90	21.0000	60.9533				
120	24.0000	67.3820				
150	27.0000	73.8107				
180	30.0000	80.2394				



Fig.15. Next Generation Greenhouse Temperature and Humidity Control.





Fig. 16. Greenhouse FIR Digital Filtering & Control of Temperature and Humidity



Fig. 17. Rule Viewer of the Real-Time Fuzzy FIR Filter for Greenhouse Control



Fig. 18. Surface Viewer of the Real-Time Fuzzy FIR Filter for Greenhouse Control

V. CONCLUSION AND RECOMMENDATION FOR FUTURE STUDIES

5.1. Conclusion

Digital FIR filter is a non-recursive linear time invariant system with finite impulse response. The filter allowed a certain threshold of frequency to be passed while blocking the other. In the design, a Low pass Digital FIR Filter allowed a low frequency component of 0.3π rad/s and blocked a high frequency of 0.7π Rad/s component signals; a High pass Digital FIR Filter allowed a High frequency component of 0.7π Rad/s and blocked a low frequency of 0.3π rad/s component signals Using MATLAB editor by creating M-file in each case. Interestingly, it has no feedback but helps in fuzzy logic design in the design of a given control plant where an extremely fast frequency response is generated employing linguistic inputs to build a Gaussian membership functions. This makes the filter to perform a Fuzzy filtering. The stability and simplicity of the Digital FIR Filter help in easy optimization of the filter using MATLAB utilizing the filter design tool box (firgr), signal processing tool box (firpm) as well as the filter visualization and tool box (fvtool); employing the Parks-McClellan Algorithm, Window methods and Frequency sampling method. More so, the principle of Convolution technique was deployed to estimate the performance of the both filters in real-time greenhouse control of temperature of Humidity where daily average temperature of 22.5°C and absolute relative humidity of 70% was achieved with cut-off frequency of the greenhouse for both the Lowpasss and High pass Digital FIR Filter still maintained at 0.5π rads/s.

5.2. Recommendation for Further Studies

As a future work, the use of fuzzy logic and Multi-rate systems in decision support system need to be more investigated and applied in various FIR filter based greenhouse control applications. Gaussian MF's are suitable for problems which require continuous differentiable curves and therefore



smooth transitions, whereas the triangular do not possess these abilities. Therefore, the use of Gaussian membership functions instead of triangular membership functions currently used in the developed fuzzy system for students' evaluation need to be investigated and tested. The effect of using Digital FIR filter in improving the energy efficient in controlling greenhouses should be considered for further investigation. Other filters such as unscented bandpass filter, which is much more suitable for non-linear systems, need to be investigated and applied.

Appendix A

Frequency Response of The Optimized Low pass Digital FIR Filter

Create an M-file and save it with LowPassFQ.m; then run the program

%%===Design of The Frequency Response For optimized low pass Digital FIR Filter Using Parks-McClellan Algorithm

F1=[0, 0.3, 0.7, 1]; % Cut-off frequencies of the FIR DIGITAL FILTER

A1= [1, 1, 0, 0]; % Desired Amplitudes of the filter Design h1=firgr (45, F1, A1, [10, 1]); % filter Design @ 45-order optimized Park-McClellan low pass digital FIR FILTER Subplot (2, 1, 1);

set(gca,'FontSize',18,'LineWidth',10,'FontWeight','bold','Font Name', 'Times New Roman')

Freqz (h1, 1, 1024); % Computes and Plots the frequency Response of Optimized LOWPASS DIGITAL FIR FILTER Title ('FREQUENCY RESPONSE OF THE OPTIMIZED LOWPASS DIGITAL FIR FILTER');

%end

OR

Using Signal Processing Tool Box, we have

h1=firpm (45, F1, A1, [10, 1]); % Compute the filter Coefficient @ 45-order optimized Park-McClellan LOWPASS DIGITAL FIR FILTER

Freqz (h1, 1, 1024); % Computes and Plots the frequency Response of Optimized LOWPASS DIGITAL FIR FILTER title ('FREQUENCY RESPONSE OF THE OPTIMIZED LOWPASS DIGITAL FIR FILTER');

%end

OR

Using filter tool and visualization b2=firls(45,[0,0.3,0.7,1],[1,1,0,0]);% compute filter Coefficient @ 45-order of optimization design H2=dfilt.dffir (b); % Filter Analysis Fvtool (H2); % filter design visualization % end

Appendix B

Frequency Response of The Optimized High pass Digital FIR Filter.

Create an M-file and save it with *HighPassFQ.m; then run the program*

%%%==== DESIGN OF THE FREQUENCY RESPONSE FOR OPTIMIZED HIGHPASS DIGITAL FIR FILTER USING Parks-McClellan Algorithm %% HighPassFQ.m F2=[0,0.3,0.7,1];% Cut-off frequencies of the FIR DIGITAL FIR FILTER Wp=0.7;% frequency passband

Ws=0.3;% frequency stopband

A2=[0,0,1,1];% Desired Amplitudes of the filter Design

h2=firgr(52,F2,A2,[20,1]);% filter Design @ 52-order optimized

Parks-McClellan HIGHPASS DIGITAL FIR FILTER subplot(2,1,1)

set(gca,'FontSize',18,'LineWidth',10,'FontWeight','bold','Font Name','Times New Roman')

freqz(h2,1,1024);% Computes and plots the frequency Response of

Optimized HIGHPASS DIGITAL FIR FILTER

title('FREQUENCY RESPONSE OF THE OPTIMIZED HIGHPASS DIGITAL FIR FILTER');

%end

OR

Using Signal Processing Tool Box, we have

h2=firpm(52,F2,A2,[20,1]);]);% Compute the filter coefficient @ 52-order optimized Parks-McClellan HIGHPASS DIGITAL FIR FILTER

freqz(h2,1,1024);% Computes and plots the frequency Response of Optimized HIGHPASS DIGITAL FIR FILTER title('FREQUENCY RESPONSE OF THE OPTIMIZED HIGHPASS DIGITAL FIR FILTER'); % end.

OR

Using Filter tools and Visualization

b=firls(52,[0,0.3,0.7,1],[0,0,1,1],[20,1]);% compute filter coefficient @ 52-order of optimization design Hd=dfilt.dffir(b);% filter Analysis fvtool(Hd);% filter design visualization % end

Appendix C

MATLAB Code Generation for Digital Low pass FIR Filter Create an M-file and save it with *LOWPASSIMPULSE.m; then run the program*

Clear all

% DESIGN OF SECOND_ORDER LOWPASS DIGITAL FIR FILTER

n=0:199; % Time span

b= [2.6267 -4.0184 2.6267]; % The filter coefficients of Low pass Digital FIR FILTER

zi= [0 0]; % show initial conditions

p1=cos (0.3*n); % cosine coefficient of LOWPASS Digital FIR FILTER

p2=cos (0.7*n); % cosine coefficient of HIGHPASS Digital FIR FILTER

P=p1+p2; % Sum of two cosine coefficients

y=filter (b, 1, P, zi); % Generate the filter output sequence of Low pass Digital FIR FILTER

plot(n,y,'k-',n,p1,'r:',n,p2,'b--',n,P,'og-','LineWidth',2);% plot the input and output sequences axis ([0 199 -6 8]);



ylabel ('AMPLITUDE', 'FontSize', 16, 'FontWeight', 'bold', 'FontName', 'Times New Roman'); xlabel ('TIMESPAN','FontSize', 16,'FontWeight','bold','FontName','Times New Roman'); legend('OUTPUT','1ST INPUT','2ND INPUT','SUM OF INPUTS', 'FontSize', 16, 'FontWeight', 'bold', 'FontName', 'Times New Roman'): title('IMPULSE RESPONSE OF SECOND ORDER DIGITAL LOWPASS FIR FILTER', 'FontSize', 18, 'FontWeight', 'bold', 'FontName', 'Times New Roman'); %end. Appendix D MATLAB Code Generation for Impulse Response of Digital High pass FIR Filter Create an M-file and save it with HIGHPASSIMPULSE.m; then run the program clear all % Design of SECOND_ORDER Highpass Digital FIR FILTER %HIGHPASSIMPULSE.m n=0:199;% Time span b=[-2.6267 5.0186 -2.6267];% The filter coefficients of **Highpass Digital FIR FILTER** zi=[0 0]:% show initial conditions close all: c1=cos(0.3*n);% cosine coefficient of Lowpass Digital FIR FILTER c2=cos(0.7*n);% cosine coefficient of Highpass Digital FIR FILTER C=c1+c2;% Sum of two cosine coefficients y=filter(b,1,C,zi);% Generate the filter output of Highpass **Digital FIR FILTER** plot(n,y,'g*-',n,c1,'r-',n,c2,'k:',n,C,'ob-','LineWidth',2);% plot the input and output sequences axis([0 199 -6 8]); ylabel('AMPLITUDE','FontSize',16,'FontName','Times New Roman'); xlabel('TIME SPAN', 'FontSize', 16, 'FontName', 'Times New Roman'); legend('OUTPUT','1ST INPUT','2ND INPUT','SUM OF INPUTS', 'FontSize', 16, 'FontName', 'Times New Roman'); title('IMPULSE RESPONSE of SECOND ORDER DIGITAL HIGHPASS FIR FILTER', 'FontSize', 18, 'FontName', 'Times New Roman'): % end. Appendix E MATLAB CODE Generation for Both Digital Low pass & High pass FIR Filter. Create an M-file and save it with COMBIMPULSE.m; then run the program Roman', 'FontWeight', 'bold') % end clear all; close all; % Design of second_order Lowpass and Highpass Digital FIR Filter; n=0:199;% Time span

b1=[2.6267 -4.0184 2.6267];% filter coefficient of Lowpass digital FIR Filter

b2=[-2.6267 5.0186 -2.6267];% filter coefficient of highpass digital FIR Filter zi=[0 0];% show initial conditions $x=\cos(0.3*n)+\cos(0.7*n)$;% input coefficient of Digital FIR Filter y1=cumsum(x); v2=filter(b1,1,v1,zi);y3=filter(b2,1,y1,zi); stairs(y2,'ob-','LineWidth',2); hold on stairs(y3,'og-','LineWidth',2); legend('LOWPASS Digital FIR Filter','HIGHPASS Digital FIR Filter', 'FontSize', 16, 'FontName', 'Times New Roman'); axis([0 199 -6 8]); xlabel('Timespan(n)','FontSize',16,'FontName','Times New Roman'); ylabel('Amplitude','FontSize',16,'FontName','Times New Roman'); title('IMPULSE RESPONSE OF DIGITAL FIR FILTER', 'FontSize', 18, 'FontName', 'Times New Roman'); %end Appendix F MATLAB Program for Convolution of Digital FIR Filter.

%=====CONVOLUTION OF DIGITAL FIR FILTER

%%CONVOLUTION.m

n=0:199:% time span

 $x=\cos(0.3*n)+\cos(0.7*n)$;% input sequence Digital FIR FILTER

b1=[2.6267 -4.0184 2.6267];% filter coefficient of LOWPASS **Digital FIR FILTER**

b2=[-2.6267 5.0186 -2.6267];% filter coefficient of HIGHPASS Digital FIR FILTER

y1=filter(b1,1,x);% filtered output of LOWPASS Digital FIR FILTER

y2=filter(b2,1,x);% filtered output sequence of HIGHPASS Digital FIR FILTER

z1=conv(y1,y2);% convoluted output of LOWPASS & HIGHPASS Digital FIR FILTER

cylinder(z1); title('CONVOLUTION OF LOWPASS & HIGHPASS DIGITAL FIR FILTER', 'FontSize', 16, 'Fontname', 'Times New Roman', 'FontWeight', 'bold'); grid off

xlabel('LOWPASS FILTERED OUTPUT(y1)','FontSize',16, 'Fontname', 'Times New Roman', 'FontWeight', 'bold') ylabel('HIGHPASS FILTERED

OUTPUT(y2)', 'FontSize', 16, 'Fontname', 'Times New Roman', 'FontWeight', 'bold') zlabel('CUT-OFF FREQUENCY', 'FontSize', 16, 'Fontname', 'Times

Appendix G MATLAB Program for Direct Convolution of Digital FIR Filter.

Create an M-file and save it with DIRECTCONV.m; then run the program

New



%%DIRECTCONV.m n=0:199;% Time span; x=cos(0.3*n)+cos(0.7*n);% input sequence of the Digital FIR Filter a1=2.6267;%first filter coefficient of Lowpass Digital FIR Filter b1=-4.0184;% second filter coefficient of Lowpass Digital FIR Filter a2=-2.6267;% first filter coefficient of Highpass Digital FIRFilter b2=5.0186;% second filter coefficient of Highpass Digital FIR filter r1=(b1-a1).*rand(10,1)+a1;%random value of Lowpass Digital FIR Filter. r2=(b2-a2).*rand(20,1)+a2;% random value of Highpass **Digital FIR Filter** u1=filter(r1,1,x);% filtered Lowpass Digital FIR filter u2=filter(r2,1,x);% filtered Highpass Digital FIR Filter w1=conv(u1,x);% convoluted Lowpass Digital FIR Filter w2=conv(u2,x);% convoluted Highpass Digital FIR Filter stairs(w1,'r','LineWidth',2); hold on stairs(w2,'g','LineWidth',2); legend('Lp Digital FIR Filter', 'Hp Digital FIR Filter', 'FontSize', 16, 'Fontname', 'Times New Roman', 'FontWeight', 'bold') grid on xlabel('INPUT', 'FontSize', 16, 'Fontname', 'Times New Roman', 'FontWeight', 'bold'); ylabel('OUTPUT', 'FontSize', 16, 'Fontname', 'Times New Roman', 'FontWeight', 'bold'); RANDOM **SIGNAL** BY title('FILTERING Α DIRECTCONVOLUTION', 'FontSize', 18, 'Fontname', 'Times New Roman', 'FontWeight', 'bold'); grid off % end Appendix H MATLAB Code for Gaussian Membership Function of Low Pass & High Pass Digital FIR Filter. Create an M-file and save it with GAUSSIAN.m ; then run the program clear all; close all; %%====Design of LOWPASS Digital FIR FILTER USING GAUSSIAN MEMBERSHIP FUNCTION % GAUSSIAN.m n=0:199;% Time span x=cos(0.3*n)+cos(0.7*n);% input of Digital FIR FILTER y1=gaussmf(x,[-4.0184 0 2.6267 2.6267]);% output of LOWPASS **Digital FIR FILTER** subplot(3,1,1); plot(x,y1,'r','LineWidth',5) xlabel('INPUT(x)','FontSize',16,'FontName','Times New Roman', 'FontWeight', 'bold'); ylabel('OUTPUT(y)','FontSize',16,'FontName','Times New Roman', 'FontWeight', 'bold');

legend('LOWPASS DIGITAL FIR FILTER','FontSize',16, 'FontName', 'Times New Roman', 'FontWeight', 'bold'); MEMBERSHIP OF title('GAUSSIAN **FUNCTION** LOWPASS DIGITAL FIR FILTER', 'FontSize', 16, 'FontName', 'Times New Roman', 'FontWeight', 'bold'); %%====Design of HIGHPASS Digital FIR FILTER USING GAUSSIAN MEMBERSHIP FUNCTION y2=gaussmf(x,[-2.6267 -2.6267 0 5.0186]);% output of HIGHPASS Digital FIR FILTER subplot(3,1,2);plot(x,y2,'g','LineWidth',5); xlabel('INPUT(x)','FontSize',16,'FontName','Times New Roman', 'FontWeight', 'bold'); ylabel('OUTPUT(y)','FontSize',16,'FontName','Times New Roman', 'FontWeight', 'bold'); legend('HIGHPASS DIGITAL FIR FILTER','FontSize',16, 'FontName', 'Times New Roman', 'FontWeight', 'bold'); **MEMBERSHIP** title('GAUSSIAN **FUNCTION** OF HIGHPASS DIGITAL FIR FILTER', 'FontSize', 16, 'FontName', 'Times New Roman' ,'FontWeight','bold'); subplot(3,1,3); plot(x,y1,'r',x,y2,'g','LineWidth',5); legend('LOWPASS FIR FILTER', 'HIGHPASS FIR FILTER', 'FontSize' ,16,'FontName','Times New Roman','FontWeight','bold'); xlabel('INPUT(x)','FontSize',16,'FontName','Times New Roman', 'FontWeight', 'bold'); ylabel('OUTPUT(y)','FontSize',16,'FontName','Times New Roman', 'FontWeight', 'bold'); title('GAUSSIAN MEMBERSHIP **FUNCTION** FOR LOWPASS & HIGHPASS DIGITAL FIR FILTER', 'FontSize', 16, 'FontName', 'Times New Roman', 'FontWeight', 'bold'); % end Appendix I MATLAB Program for Greenhouse Temperature and Humidity Control. Create M-file it with an and save

GREENHOUSECONTROL2.m; then run the program clc;

clear all; close all; %% GREENHOUSE TEMPERATURE AND RELATIVE

HUMIDITY CONTROL % GREENHOUSECONTROL2.m n=30:30:199;% time span T=[15 18 21 24 27 30];% Temperature control RH=2.1429*T+15.9524;% Relative Humidity Y=[15 48.0959;18 54.5246;21 60.9533;24 67.3820;27 73.8107;30 80.2494]; bar(n,Y,'hist'); legend('TEMPERATURE(DEGREE CELSIUS','RELATIVE HUMIDITY(%)'); xlabel('TIMESPAN(n)','FontSize',16,'FontWeight','bold','Font Name', 'Times New Roman');



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ylabel('GREENHOUSE

CONTROL

INPUTS', 'FontSize', 16, 'FontWeight', 'bold', 'FontName', 'Times New Roman');

title('GREENHOUSE TEMPERATURE AND HUMIDITYCONTROL','FontSize',18,'FontWeight','bold','Fon tName','Times New Roman'); grid on

%end.

Appendix J

MATLAB Program for Implementation of Greenhouse Filtering, Temperature and Humidity Control Create an M-file and save it with

GREENHOUSEFILTERING.m; then run the program

%% IMPLEMENTATION OF GREENHOUSE FILTERING, TEMPERATURE AND HUMIDITY CONTROL

%GREENHOUSEFILTERING.m

n=0:199; % Time span

x=cos (0.3*n) +cos (0.7*n);

b1= [2.6267 -4.0184 2.6267]; % The filter coefficients of Low pass Digital FIR FILTER

b2= [-2.6267 5.0186 -2.6267]; % The filter coefficients of High pass Digital FIR FILTER

y1=gaussmf(x, [-4.0184 0 2.6267]); % output of LOWPASS Digital FIR FILTER

y2=gaussmf(x, [-2.6267 0 5.0186]); % output of HIGHPASS Digital FIR FILTER

z1=conv (y1, y2); % convoluted output of LOWPASS & HIGHPASS Digital FIR FILTER

T= [15 18 21 24 27 30]; % Temperature control

RH=[48.0959 54.5246 60.9533 67.3820 73.8107 80.2494];% Relative Humidity

TK=T+273; % Absolute Temperature

z2=conv (TK, RH); % convoluted Inputs of Temperature & Humidity control

Cylinder (z1);

hold on

Cylinder (z2);

xlabel('LOWPASSCUT-OFF', 'FontSize',

16, 'FontWeight', 'bold', 'FontName', 'Times New Roman');

ylabel('HIGHPASS CUT-OFF', 'FontSize', 16, 'FontWeight', 'bold', 'FontName', 'Times New Roman'); zlabel('FUZZY CONTROL INPUTS', 'FontSize', 16, 'FontWeight', 'bold', 'FontName', 'Times New Roman'); title('GREENHOUSE FILTERING, TEMPERATURE AND HUMIDITY CONTROL', 'FontSize', 16, 'FontWeight', 'bold', 'FontName', 'Tim es New Roman'); grid on

grid off

%end

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