	EE2016 - Middem - Part-c	Sakthi Harish D
		FE14B124
	Report	9597859141
1. 7	Theory of FIR, Convolutions, DTFT, etc.	
k	FIR (Finite Lupuke Kesponse) Filter	is a Jiller
	whose impulse Response is of finite	duration, because
M	it settles to <u>zero</u> in <u>finite</u> time. They can be discrete-time or continued	٤,
71	divital or analog	ous-time &
	digital or analog.	ov M order N
7	FFOR a Causal Discrete-time FIR gilt	in a weighted
	each value of the output sequence dum of the most recent input va	lues.
	$y[n] = \sum_{i=0}^{N} b_i \cdot x[n-i]$	
K	The above computation is also Kn	own as discrete
	Convolution.	
	$ \frac{f *g[n]}{f *g[n]} = \underset{m=-\infty}{\overset{\infty}{\underset{n=-\infty}{\longrightarrow}}} f[m]g[n-m] \text{ (or)} = $ (for infinite)	$= \underbrace{\overset{M}{\longleftarrow}}_{m=-M} f(n-m)g(m)$ (for finite)
,	*DTFT: Discrete-Time Fourier Trans	agorm: The term
	discrete-time refers to the fact	
	transform operates on discrete	data, often samples
	whose interval has units of time	u.
	$X_{2\pi}(\omega) = \sum_{n=0}^{\infty} x[n] e^{-i\omega n}$	
4	The Fast Fourier Transform (F	FT) is an
,	algorithm for computing one cycl	e of the DFT,
	& its inverse produces one cycle	of the inverse DFT.

2. Fixed Point Prithmetic:

To compute the FIR filter in Atmega 8 microprocessor, we need to convert the floating point decimals to Fixed point. The Coefficients, inputs must be converted to Fixed point format. From the Hardware multiplier reading material, this can be understood. In our case of 8-bits for each coefficient & input, the first/NSB/bit acts as the sign bit 8 the rest 7 bits acts as the Fixed point approximation of the decimal number. Hence only values between (-1,1) can be fed into the system. So our data must necessarily be Normalized.

3. DTFT:

* While the signals in Time Domain may be approximate, the DTFT computed by FFT (Fast Fourier Transform) gives correct output w.r.t the signals in Frequency Domain.

** DTFT of a Dc dignal is ideally an unit impulse function at 0 Hz. We can see that it is the case in our dignal also.

* DTFT of a Sine signal is a jump in the signal at the operating frequency (≈ 1800 Hz). We get 2 such jumps for for both positive to negative value of frequency.

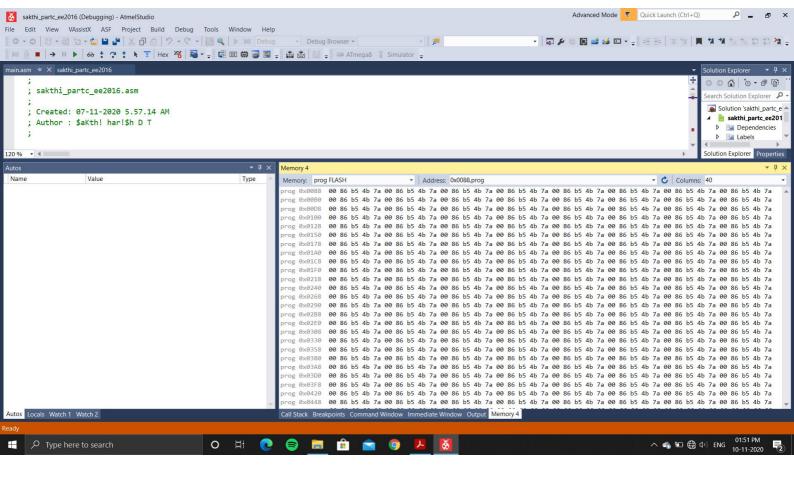
* DTFT of a Whitenoise is ideally Flat, but is not the case in Real life signals. The signal is not the case in Real life signals. The signal is within the limit of Power spectrum.

4. Implementation Details

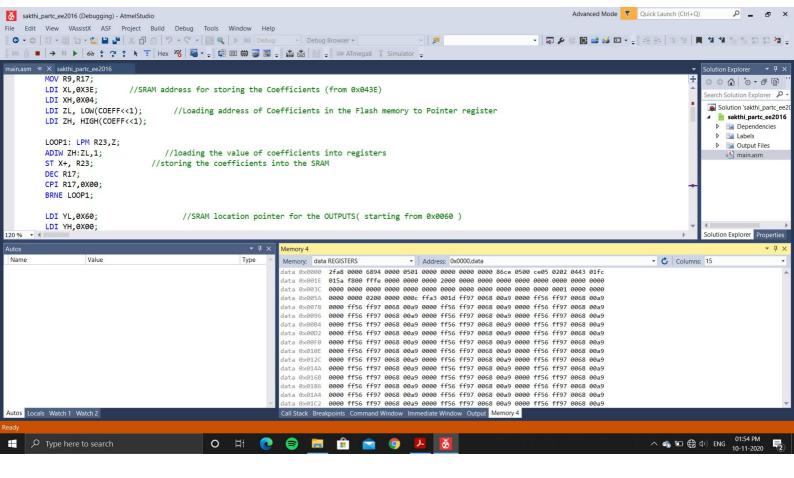
- (a) Input: The input is provided as Fixed Point Integers in ab format at the end of the Cocle. As the dize of input is large, it is not stored in SRAM, but stays in the Rogram Flash Memory.
 - (b) Output: The output is of 2 Bytes each . It is written into the SRAM from the address Ox0060. The output must be copied separately
 - and converting to Floating point numbers by Converted dividing with 27.
 - C) Coefficients: The Coefficients are taken as input in odb format & stored in SRAM address starting from DX 043E
 - Multiplication Details: For multiplication, the FMULS command is used to do Signed Multiplication on Fractions. Rig is the register which has the inputs Fractions. Rig is the register which has the inputs to Ris has coefficients. Result is accumulated into Ris: Rs: R4 as 3 Bytes, but only Ri: Rs is written into SRAM (2 Bytes).
 - NOTE: As the input dignals are Causal, the First N'
 outputs do not have N' product terms, hence
 output for periodic dignals will be non-periodic in the
 beginning.

4) (e) Screenshots of Various Memory locations:

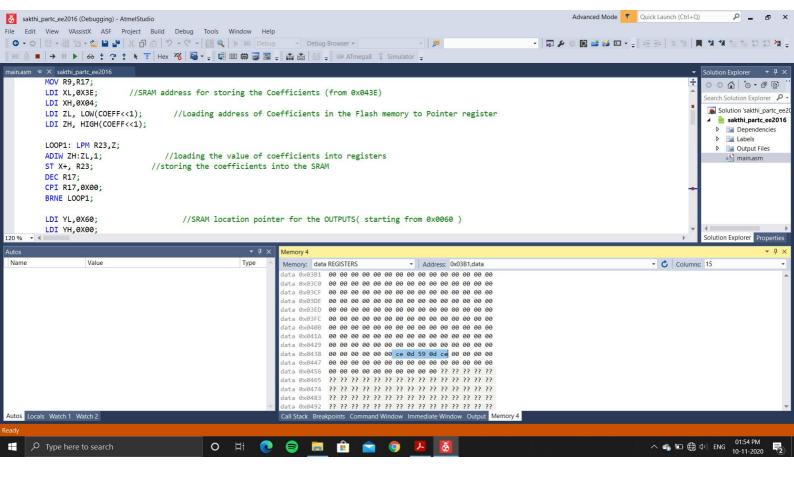
INPUTS Stored in Flash Memory



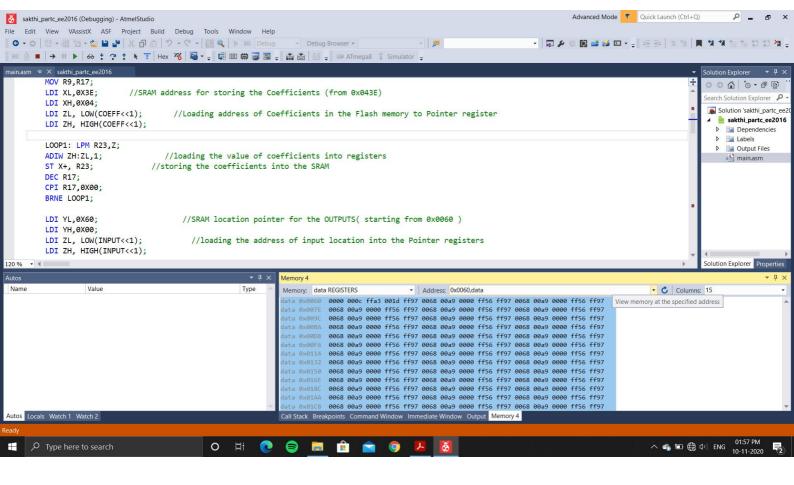
Data Registers



Coefficients Stored in SRAM(5-tap for example)



Outputs stored in SRAM



5) RESULTS:

