

# Methods for Nonlinear Impairments Compensation in Fiber-Optic Communication Systems

by

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## Abstract

Fiber optic links are the backbone of the current high-speed tele- and data communication networks. Two major factors which cause degradation in the performance of this type of communication systems are fiber optic link nonlinear impairments, self-phase and cross-phase modulation, and linear additive noise. To tackle with linear noise, the average energy of constellation points should be increased and this increment also increases the nonlinear impairments. This research explores some possible methods of tackling the issues caused by nonlinear impairments without changing the average energy of constellation points.

In chapter 1, we will review mathematical models introduced in literature for modelling nonlinearities in fiber optic links. In Chapter 2, we will introduce a simplified model for modelling nonlinearities which can help us with compensating procedure by reducing computational complexity and providing feasibility for parallel computations. In Chapter 3, a novel method called Masking Data Samples will be introduced which can reduce the power of self-phase modulation by sampling from that, which is a random variable, and choosing the best sample in sense of SPM noise between all obtained ones. In Chapter 4, a modified version of known tree-structure codes will be introduced which can reduce the power of SPM noise by a new sorting method in the phase of producing look-up table for codes. At the last chapter, a method for exploiting the statistical characteristics and memory of the samples of cross-phase modulation noise will be introduced which can help us with detecting data symbols more accurately.

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## **Dedication**

To my mother, a walking miracle.

# Table of Contents

List of Tables	ix
List of Figures	xi
<b>1 Overview and Literature Review</b>	<b>1</b>
1.1 Overview . . . . .	1
1.2 Modeling Nonlinearities . . . . .	1
1.3 Literature Review . . . . .	7
1.4 Summary . . . . .	8
<b>2 Simplified Model for C Matrix</b>	<b>9</b>
2.1 Model Structure . . . . .	9
2.2 Advantages of Using N-span Model for Computing SPM Noise in a Fiber Optic Link . . . . .	16
2.2.1 Computational Complexity Reduction . . . . .	16
2.2.2 Parallel Computing . . . . .	22
2.3 Optimization Algorithm . . . . .	28
2.3.1 Nelder-Mead Simplex Algorithm . . . . .	28
2.3.2 Optimization Parameters . . . . .	32
2.3.3 Cost Functions . . . . .	33
2.4 Results Obtained from Statistical Fitting . . . . .	34

2.4.1	Model Fitting Results . . . . .	35
2.4.2	Fitting the Model to Compensated Data Samples . . . . .	41
2.5	Results Obtained from Algebraic Fitting . . . . .	52
2.6	Summary . . . . .	54
<b>3</b>	<b>Data Masking For SPM Noise Reduction</b>	<b>55</b>
3.1	Introduction . . . . .	55
3.2	Data Masking . . . . .	55
3.3	Masking Procedure . . . . .	57
3.4	Results . . . . .	60
3.5	Summary . . . . .	61
<b>4</b>	<b>Modified Tree-structure Code for SPM Noise Reduction</b>	<b>62</b>
4.1	Introduction . . . . .	62
4.2	Shaping and Tree-structure codes . . . . .	62
4.3	Modified Version of Tree-structure Codes . . . . .	64
4.3.1	Sorting Method Used in 2, 4, and 8-dimensional Spaces . . . . .	64
4.3.2	Sorting Method Used in Spaces with Dimension Higher than 8 . . . . .	65
4.4	Results . . . . .	66
4.5	Summary . . . . .	66
<b>5</b>	<b>Joint Detection for Exploiting The Memory of XPM noise</b>	<b>67</b>
5.1	Introduction . . . . .	67
5.2	The statistical Characteristics of XPM Noise . . . . .	67
5.3	Joint Detection Scheme . . . . .	73
5.4	Results . . . . .	73
5.5	Conclusion . . . . .	74
	<b>References</b>	<b>75</b>

<b>APPENDICES</b>	<b>78</b>
<b>A Simplex Method</b>	<b>79</b>
<b>B Statistical Fitting Cost Function</b>	<b>82</b>
<b>C Algebraic Fitting Cost Function</b>	<b>84</b>
<b>D Simplex Result Testing</b>	<b>88</b>
<b>E Compensation with C matrix</b>	<b>91</b>
<b>F Data Masking</b>	<b>93</b>
<b>G Tree Code Sorting Method</b>	<b>102</b>
<b>H Corresponding SPM Noise to Energy Vectors</b>	<b>108</b>
<b>I Joint Detection</b>	<b>110</b>
I.1 Codes for Calculating XPMs . . . . .	110
I.2 MATLAB Code for Calculating Joint Probability Density Functions . . . . .	112
I.3 Matlab Code for Finding Covariance Matrices . . . . .	113
I.4 Matlab Code for Finding Conditional Expected Values for XPM (Only one data symbol) . . . . .	114
I.5 Matlab Code for Comparing Joint and Minimum Distance Detection Methods	115



# List of Tables

2.1	Model's Parameters for the first case (Linear block first, NDSF fiber) . . .	36
2.2	Mean squared error and the level of SPM noise for 3 batches of test data samples (Linear block first, NDSF fiber) . . . . .	36
2.3	Model's Parameters for the first case (Linear block first, ELEAF fiber) . .	37
2.4	Mean squared error and the level of SPM noise for 3 batches of test data samples (Linear block first, ELEAF fiber) . . . . .	37
2.5	Model's Parameters for the second case (Linear block first, fitted to $9 \times C$ matrix, NDSF fiber) . . . . .	37
2.6	Mean squared error and the level of SPM noise for 3 batches of test data samples (Linear block first, fitted to $9 \times C$ matrix, NDSF fiber) . . . . .	38
2.7	Model's Parameters for the second case (Linear block first, fitted to $9 \times C$ matrix, ELEAF fiber) . . . . .	38
2.8	Mean squared error and the level of SPM noise for 3 batches of test data samples (Linear block first, fitted to $9 \times C$ matrix, ELEAF fiber) . . . . .	38
2.9	Model's Parameters for the third case (Non-linear block first, NDSF fiber)	39
2.10	Mean squared error and the level of SPM noise for 3 batches of test data samples (Non-linear block first, NDSF fiber) . . . . .	39
2.11	Model's Parameters for the third case (Non-linear block first, ELEAF fiber)	39
2.12	Mean squared error and the level of SPM noise for 3 batches of test data samples (Non-linear block first, ELEAF fiber) . . . . .	40
2.13	The variance of SPM noise after doing compensation (NDSF fiber link) . .	42
2.14	The variance of SPM noise after doing compensation (ELEAF fiber link) .	43

2.15	Model's Parameters for the fourth path (NDSF fiber) . . . . .	46
2.16	The varinace of SPM noise for different cases (NDSF fiber) . . . . .	46
2.17	Model's Parameters for the fourth path (ELEAF fiber) . . . . .	49
2.18	The varinace of SPM noise for different cases (ELEAF fiber) . . . . .	49
2.19	Model's Parameters obtained from algebraic fitting (NDSF fiber) . . . . .	53
2.20	The varinace of SPM noise for different cases (NDSF fiber) . . . . .	53
3.1	Mask with length 400 bits . . . . .	60
3.2	Mask with length 200 bits . . . . .	60
4.1	Different Sorting Methods . . . . .	66

# List of Figures

2.1	Block model . . . . .	10
2.2	Block Levels . . . . .	12
2.3	Cascading the model two times . . . . .	14
2.4	Hierarchical structure . . . . .	14
2.5	Final Model . . . . .	15
2.6	Hierarchical structure for the final model . . . . .	15
2.7	Hierarchical structure for two consecutive data samples . . . . .	17
2.8	Memory Allocation1 . . . . .	18
2.9	Reusing values for computing $SPM'_3$ . . . . .	19
2.10	Memory flow after calculating $SPM'_3$ . . . . .	19
2.11	Computing the output of the linear block . . . . .	20
2.12	Memory flow after computing the output of the span . . . . .	20
2.13	Memory allocation when system is ready for the next data sample . . . . .	21
2.14	Simplified Structure . . . . .	23
2.15	First clock cycle . . . . .	23
2.16	Sixth clock cycle. . . . .	24
2.17	11th clock cycle . . . . .	25
2.18	21th clock cycle . . . . .	26
2.19	Processors continue working in parallel . . . . .	27
2.20	The geometric interpretation of the operations . . . . .	30

2.21	Nelder-Mead Simplex Algorithm flowchart . . . . .	31
2.22	Compensation Scheme . . . . .	34
2.23	16-QAM constellation points after passing through an NDSF fiber link . . . . .	42
2.24	16-QAM constellation points after passing through an NDSF fiber link . . . . .	43
2.25	Compensation Scheme . . . . .	45
2.26	4 examined cases . . . . .	45
2.27	16-Qam constellation points (uncompensated, compensated once by using C matrix, compensated once by using a fitted model. NDSF fiber link) . . . . .	47
2.28	16-Qam constellation points (uncompensated, compensated once by using C matrix, compensated with a model fitted to the values of 3-time compensated SPMs. NDSF fiber link) . . . . .	48
2.29	16-Qam constellation points (uncompensated, compensated once by using C matrix, compensated once by using a fitted model. ELEAF fiber link) . . . . .	50
2.30	16-Qam constellation points (uncompensated, compensated once by using C matrix, compensated with a model fitted to the values of 3-time compensated SPMs. ELEAF fiber link) . . . . .	51
2.31	16-Qam constellation points (Algebraic Fitting, NDSF fiber) . . . . .	54
3.1	Sampling of SPM noise with masks . . . . .	56
3.2	We need previous and next data symbols to compute SPM . . . . .	58
3.3	The procedure for choosing masks . . . . .	58
3.4	Masking Algorithm . . . . .	59
5.1	The effect of XPM noise on constellation points . . . . .	69
5.2	The probability of the energy of next sample conditional on the current sample . . . . .	70
5.3	The probability of the energy of next sample conditional on the current sample . . . . .	71
5.4	The principal component, Yellow line, for the XPM noise corresponding constellation point $0.6708 - 0.6708i$ . . . . .	72
5.5	Comparison of bit error rate plots for two detection methods . . . . .	74

# Chapter 1

## Overview and Literature Review

### 1.1 Overview

Fiber-optic communication systems are the backbone of the recent tele-communication systems. High capacity optical systems are using digital coherent detection and this technology is the main one used for 100-Gb/s transport network [21]. To go beyond this rate, more complex modulation schemes such as 16QAM must be used [6] [20] which is the modulation scheme considered in this thesis. Fiber optic links cause problems which can be modeled with linear, such as chromatic and polarization-mode dispersion, and nonlinear models. The effects of linear impairments can be compensated by signal processing techniques, but there is not straightforward way to tackle the nonlinear types of noise like the Self Phase Modulation (SPM) and Cross Phase Modulation (XPM). Throughout this thesis, some methods will be introduced which can reduce the effects of these types of nonlinear impairments.

In this chapter, first we will try to model these nonlinearities with nonlinear equations. Then a mathematical model, called C matrix, will be introduced. This matrix can help us in modeling the nonlinear effects of a fiber optic link.

### 1.2 Modeling Nonlinearities

In this section, we are going to take a look into some equations which can be used to model nonlinear effects of a fiber optic link. It should be noted that all the material mentioned in this section, is a summary of three documents provided by Ciena Corporation [4] [5] [13].

To model nonlinearities in a fiber optic link, split-step model has been used. This model is the inspiration for the simplified model introduced in the next chapter. Split-step method states that in order to model nonlinearities in a fiber optic link, we can divide the link with length  $L$  into some cascaded smaller segments with length  $\delta$  [13]. In other words, the nonlinearity for the whole length will be a functional composition of nonlinearities produced in small segments, which is much easier to be modeled. To model nonlinearities produced in each segment, we have to consider two effects[13]:

1. Dispersion Operation (shown by  $D(\cdot)$ ).
2. Nonlinear Operation (shown by  $NL(\cdot)$ ).

By considering these operations, and assuming the signal is operating at a high spectral efficiency, we can model the output of a fiber optic link as follows [13]:

$$y(t) = x(t) + \sum_{i=0}^{\frac{L}{\delta}-1} D^i (NL(D^{n-i}(x(t)))) \quad (1.1)$$

where  $L$  is the length of fiber optic link, and  $\delta$  is the step size for the split-step method. Because of the high spectral efficiency, nonlinearities caused by different segments and also Amplified Spontaneous Emission (ASE) are not coupling on each other. By considering this assumption, we use the following approximation:  $NL(D^{n-i}(x(t)) + NL(D^{n-i}(x(t))) + ASE) \sim NL(D^{n-i}(x(t)))$  [13]. These two operators, dispersion and nonlinear, can be modeled in the time domain as follows [13]:

$$NL(x(t)) = x(t) (e^{j\gamma|x(t)|^2} - 1) \approx x(t) (j\gamma|x(t)|^2) \quad (1.2)$$

$$y(t) = D(x(t)) = h(t) * x(t) = IFT\{e^{j\beta(f+\alpha)^2}\} * x(t) \quad (1.3)$$

in the first equation we assumed that  $\lambda$  is small, because the steps ( $\delta$ ) are small. Also, we approximated exponential term with the first component of its Taylor series [13].

The nonlinear part of Equation 1.1 in the frequency domain is as follows [13]:

$$NL(f) = FT \left\{ \sum_{i=0}^{\frac{L}{\delta}-1} D^i (NL(D^{n-i}(x(t)))) \right\} = \sum_{i=0}^{\frac{L}{\delta}-1} FT \{D^i (NL(D^{n-i}(x(t))))\} = \sum_{i=0}^{\frac{L}{\delta}-1} e^{\frac{j\beta(L-i\delta)(f+\alpha)^2}{L}} NL \left( e^{\frac{j\beta i\delta(f+\alpha)^2}{L}} X(f) \right) \quad (1.4)$$

To simplify equation 1.4, we should define two notations:

$$x_i(t) \triangleq D^{n-i}(x(t)) \quad (1.5)$$

$$y_i(t) \triangleq NL(D^{(n-i)}(x(t))) \quad (1.6)$$

by using these notations we get [13]:

$$\begin{aligned} y_i(t) &= j\gamma x_i(t) |x_i(t)|^2 \rightarrow Y_i(f) \triangleq FT \{y_i(t)\} = j\gamma X_i(f) * X_i(f) * X_i^*(-f), X_i(f) \triangleq FT \{x_i(t)\} \\ &\rightarrow Y_i(f) = \int^{X_i}(f_1) [X_i(f) * X_i^*(-f)]_{f-f_1} df_1 \\ &= \int^{X_i}(f_1) \int^{X_i}(f_2) X_i^*(-(f-f_1-f_2)) df_2 df_1 \\ &= \iint^{X_i}(f_1) X_i(f_2) X_i^*(f_1+f_2-f) df_1 df_2 \end{aligned} \quad (1.7)$$

after combining above equations and knowing  $X_i(f) = e^{\frac{j\beta i\delta(f+\alpha)^2}{L}} X(f)$  we have [13]:

$$\begin{aligned} NL(f) &= \\ j\gamma \sum_{i=0}^{\frac{L}{\delta}-1} e^{\frac{j\beta(L-i\delta)(f+\alpha)^2}{L}} &\iint_{\substack{-W \leq f_1, f_2, f_3 \leq W \\ f_3 = f_1 + f_2 - f}} e^{\frac{j\beta i\delta[(f_1+\alpha)^2+(f_2+\alpha)^2-(f_3+\alpha)^2]}{L}} X(f_1) X(f_2) X^*(f_3) df_1 df_2 \\ &= j\gamma e^{j\beta(f+\alpha)^2} \sum_{i=0}^{\frac{L}{\delta}-1} \iint_{\substack{-W \leq f_1, f_2, f_3 \leq W \\ f_3 = f_1 + f_2 - f}} e^{\frac{j\beta i\delta[f_1^2+f_2^2-f^2-f_3^2]}{L}} X(f_1) X(f_2) X^*(f_3) df_1 df_2 \\ &= j\gamma e^{j\beta(f+\alpha)^2} \iint_{\substack{-W \leq f_1, f_2, f_3 \leq W \\ f_3 = f_1 + f_2 - f}} \sum_{i=0}^{\frac{L}{\delta}-1} e^{\frac{j\beta i\delta[f_1^2+f_2^2-f^2-f_3^2]}{L}} X(f_1) X(f_2) X^*(f_3) df_1 df_2 \\ &= j\gamma e^{j\beta(f+\alpha)^2} \iint_{\substack{-W \leq f_1, f_2, f_3 \leq W \\ f_3 = f_1 + f_2 - f}} \frac{e^{j\beta[f_1^2+f_2^2-f^2-f_3^2]} - 1}{\frac{j\beta\delta[f_1^2+f_2^2-f^2-f_3^2]}{L} - 1} X(f_1) X(f_2) X^*(f_3) df_1 df_2 \\ &\approx \frac{L}{\beta\delta} \gamma e^{j\beta(f+\alpha)^2} \iint_{\substack{-W \leq f_1, f_2, f_3 \leq W \\ f_3 = f_1 + f_2 - f}} \frac{e^{j\beta[f_1^2+f_2^2-f^2-f_3^2]} - 1}{f_1^2+f_2^2-f^2-f_3^2} X(f_1) X(f_2) X^*(f_3) df_1 df_2 \end{aligned} \quad (1.8)$$

by assuming a low level of dispersion in each segment, we can use the following approximation:  $e^{\frac{j\beta\delta[f_1^2+f_2^2-f^2-f_3^2]}{L}} \approx 1 + j\beta\delta[f_1^2 + f_2^2 - f^2 - f_3^2]$  and the equation will be simplified

as follows [13]:

$$\begin{aligned}
NL(f) &= \frac{L}{\beta\delta} \gamma e^{j\beta(f+\alpha)^2} \iint_{\substack{-W \leq f_1, f_2, f_3 \leq W \\ f_1 + f_2 = f + f_3}} X(f_1) X(f_2) X^*(f_3) \frac{e^{j\beta[f_1^2+f_2^2-f^2-f_3^2]} - 1}{f_1^2+f_2^2-f^2-f_3^2} df_1 df_2 \\
&= \iint_{\substack{-W \leq f_1, f_2, f_3 \leq W \\ f_1 + f_2 = f + f_3}} X(f_1) X(f_2) X^*(f_3) A(f, f_1, f_2) df_1 df_2
\end{aligned} \tag{1.9}$$

Equation 1.9 is an approximation for the nonlinear effect of a fiber optic link, computed by using split-step method in the frequency domain. As it can be seen in the equation, the nonlinear effect of a fiber optic link consists of an integration and triple multiplications of data symbols multiplied by a factor. We need a discrete representation for this nonlinearity. It should be considered that this equation is a general form. It means that we can model SPM noise if we put data symbols of the channel, for which we want to find nonlinearity, in the equation. If we put data symbols of neighboring channels in the equation, we will obtain the value of XPM noise.

The discrete equation for the nonlinearities in a fiber optic link, has 6 components. If we are launching data symbols on Xpole and Ypole, the nonlinear noise in Xpole at the current time (time zero) can be computed as follows [4]:

$$\Delta A_x = \text{SPM}_1 + \text{SPM}_2 + \sum_w (\text{XPM}_{1w} + \text{XPM}_{2w} + \text{XPM}_{3w} + \text{XPM}_{4w}) \tag{1.10}$$

where summation is over all the neighboring channels. XPM and SPM terms can be computed as follows [4]:

$$\text{SPM}_1 = \sum_{m,n} C_{m,n}^{\text{spm}} A_x(m) A_x(n) A_x^c(m+n) \tag{1.11}$$

$$\text{SPM}_2 = \sum_{m,n} C_{m,n}^{\text{spm}} A_x(m) A_y(n) A_y^c(m+n) \tag{1.12}$$

$$\text{XPM}_{1w} = \sum_{m,n} C_{m,n}^{\text{xpmw}} A_x(m) B_x(n) B_x^c(m+n) \tag{1.13}$$

$$\text{XPM}_{2w} = \sum_{m,n} C_{m,n}^{\text{xpmw}} A_x(m) B_y(n) B_y^c(m+n) \tag{1.14}$$



$$\text{XPM}_{3w} = \sum_{m,n} C_{m,n}^{\text{xpolemw}} B_x(m) A_x(n) B_x^c(m+n) \quad (1.15)$$

$$\text{XPM}_{4w} = \sum_{m,n} C_{m,n}^{\text{ypolemw}} B_x(m) A_y(n) B_y^c(m+n) \quad (1.16)$$

in the above equations,  $w$  is one of the neighboring channels,  $A_x$  and  $A_y$  are data symbols on Xpole and Ypole of the channel for which the nonlinearities are being computed,  $B_x$  and  $B_y$  are data symbols on Xpole and Ypole of the channels which are neighbors to the examined channel, and  $c$  is conjugate operation for complex numbers. It must be noted that  $A$  and  $B$  are complex numbers corresponding to constellation points. The values of  $C$ , shown in the equations, are the counterparts of factor  $A$  in Equation 1.9. These  $C$  values come from matrices known as *C matrices* which can be computed by using machine learning algorithms or analytical equations. Discussing about the methods for computing  $C$  matrices is not in the scope of this thesis. Throughout this thesis, we assume that  $C$  matrices are known, and we do not need to compute them.

After calculating  $\Delta A_x$ , we can find the constellation point which we will be observed after passing data symbols through the fiber optic link, by adding the value of  $\Delta A_x$  to the value of constellation point sent at the time zero. This computed nonlinear noise is not the effective value of that. It adds to each constellation point an average value of noise which will change that position of the constellation points and deform the structure of constellation points in the used modulation. This average value cannot cause problem because it can be calculated statistically and in the detection phase the effect of that can be eliminated by a simple subtraction. In other words, the effective nonlinear noise can be computed as follows [4]:

$$\Delta A_{\text{ex}} = \Delta A_x - E[\Delta A_x | A_x(0)] \quad (1.17)$$

the term  $E[\Delta A_x | A_x(0)]$  shows the value of noise which will be added to the constellation points on average, conditional on the sent constellation points.

Equations 1.2 to 1.16 help us in calculating the value of nonlinearities in time. Another important step for modeling these nonlinearities is finding the statistical characteristics of that. As we discussed before, we must consider the effective nonlinear noise which has zero mean value. For the variance of the effective noise we can consider following equations [4]:

$$\delta_{\text{nlx}}^2 = E[\Delta A_{\text{ex}} \times \Delta A_{\text{ex}}^c] \quad (1.18)$$

$$\delta_{\text{nlx}}^2 = E[\text{SPM}_1 \times \text{SMP}_1^c + \text{SPM}_2 \times \text{SMP}_2^c + 2\text{real}(\text{SPM}_1 \times \text{SMP}_2^c) + \text{XPM}_{w1} \times \text{XMP}_{1w}^c + \dots] \quad (1.19)$$

we will only find a closed form for the first term. The closed form for the other terms can be computed in the same manner. To compute the first term, suppose that we have a multidimensional constellation with  $2N$  real dimensions and constellations are generated based on a permute invariant distribution  $pmf$ . The final result for the first term will be as follows [5]:

$$E[\text{SPM}_1 \times \text{SMP}_1^c] = \alpha_2 m_2^3 + \alpha_4 m_2 m_4 + \alpha_6 m_6 + \alpha_{(4_1)} m_2 m_{(4_1)} + \alpha_{(6_1)} m_{(6_1)} + \alpha_{(6_2)} m_{(6_2)} \quad (1.20)$$

some terms should be specified[5]:

$$m_2 = E \{C_{i2}\} = \sum_{i=1}^K pmf(C_i) C_{i2} \quad (1.21)$$

$$m_4 = E \{C_{i4}\} = \sum_{i=1}^K pmf(C_i) C_{i4} \quad (1.22)$$

$$m_6 = E \{C_{i6}\} = \sum_{i=1}^K pmf(C_i) C_{i6} \quad (1.23)$$

$$m_{4.1} = E \{C_{i4.1}\} = \sum_{i=1}^K pmf(C_i) C_{i4.1} \quad (1.24)$$

$$m_{6.1} = E \{C_{i6.1}\} = \sum_{i=1}^K pmf(C_i) C_{i6.1} \quad (1.25)$$

$$m_{6.2} = E \{C_{i6.2}\} = \sum_{i=1}^K pmf(C_i) C_{i6.2} \quad (1.26)$$

in the above equations  $C_i$ s can be calculated by using the following equations[5]:

$$C_{i2} = \frac{1}{N} \sum_{k=1}^N |c_{ik}|^2 \quad (1.27)$$

$$C_{i4} = \frac{1}{N} \sum_{k=1}^N |c_{ik}|^4 \quad (1.28)$$

$$C_{i6} = \frac{1}{N} \sum_{k=1}^N |c_{ik}|^6 \quad (1.29)$$

$$C_{i4.1} = \frac{1}{\binom{N}{2}} \sum_{k_1}^N \sum_{k_2 > k_1}^N |c_{ik_1}|^2 |c_{ik_2}|^2 \quad (1.30)$$

$$C_{i6.1} = \frac{0.5}{\binom{N}{2}} \sum_{k_1=1}^N \sum_{k_2 > k_1}^N |c_{ik_1}|^2 |c_{ik_2}|^4 + \frac{0.5}{\binom{N}{2}} \sum_{k_1=1}^N \sum_{k_2 > k_1}^N |c_{ik_1}|^4 |c_{ik_2}|^2 \quad (1.31)$$

$$C_{i6.2} = \frac{1}{\binom{N}{3}} \sum_{k_1=1}^N \sum_{k_2 > k_1}^N \sum_{k_3 > k_2}^N |c_{ik_1}|^2 |c_{ik_2}|^2 |c_{ik_3}|^2 \quad (1.32)$$

it should be noted that the values for  $\alpha_s$  in Equation 1.20 must be computed for the examined fiber optic link and it is different from one type of fiber optic link to the other.

In this section, we looked at some important formulas which we can model the nonlinearities by using them. In the next section we will look at some methods used before for compensating these nonlinearities in fiber optic links.

### 1.3 Literature Review

Many methods have been introduced for reducing the mentioned self-phase modulation (SPM) noise. One of the introduced methods is using materials with negative nonlinear refractive-index coefficient [14]. Using this type of material is not applicable because of bandwidth limitation [22]. Another introduced method for self-phase modulation compensation is using data-driven phase modulator [23]. This system works with a phase modulator with magnitude proportional to the detected pulse intensity. The sign is opposite of the nonlinear phase shift caused by SPM. Other proposed method is using Optical Phase Conjugate (OPC). This method can compensate chromatic dispersion as well. In this technique, first optical pulses distorted by Group Velocity Dispersion (GVD), then they will be passed through the OPC [17]. OPC simply generates replicas of incident optical waves which is shown that can be used to correct channel dispersion [24]. SPM also can be reduced by using fractionally spaced analog tap delays [19]. Also, this method can eliminate dispersion with enough number of tap delays [19].

For cross-phase modulation compensation, one of the introduced methods is using intensity-dependent phase-modulation. In this technique, the intensity of phase modulator must be controlled by the received signals from other channels [3]. Other method for mitigating XPM is using-nonzero dispersion fiber to induce walk off [9]. Another introduced method is back propagation which is introduced in [16] and [11]. In this method, inverse Schrödinger equations are used to estimate what is the transmitted signal [9]. Both linear and nonlinear impairments can be compensated by using this technique [9].

In this thesis we used a method introduced by Kim B. Roberts, Leo Strawczynski, and Maurice S. O’Sullivan in their patent: “Electrical domain compensation of non-linear effects in an optical communications system”. In this method, an estimation of self-phase modulation noise is calculated by passing data symbols through an approximated model for a fiber optic link, called C matrix, and after that the data symbols will be pre-distorted by the computed SPM values by subtracting the value of computed SPM noise from constellation points. This subtraction will help in mitigating the SPM noise, which will be generated by the fiber optic link [15]. For reducing the effect of cross-phase modulation, non-zero dispersion fiber links will be used which can generate the mentioned walk off effect.

## 1.4 Summary

In this chapter, first we looked at the procedure for finding a mathematical model of nonlinear impairments. Then we introduced some known methods for reducing nonlinear effects of a fiber optic link. At the end, it should be mentioned that the method proposed by Kim B. Roberts, Leo Strawczynski, and Maurice S. O’Sullivan, and used in this thesis, has some advantages:

- It can be done completely in digital domain. In other words, there is no necessity for changing the physical structure and material of fiber optic links or detectors.
- This method does not affect chromatic dispersion by which the cross-phase modulation noise can be compensated.

# Chapter 2

## Simplified Model for C Matrix

As we discussed before, one promising way for SPM noise reduction is compensating the effect of a fiber optic link before launching the data through that. To do this compensation procedure, we need to compute the amount of SPM noise which will affect the current data and subtract that value from the constellation point. Finding the value of SPM by the formula discussed before costs memory, and is computationally complex. Our goal in this chapter is introducing a model by which we can approximate the formula for computing SPM to the result of some simple and iterative formulas.

In this chapter, first we will introduce the structure of the model. We will discuss how the model can reduce computational complexity, and can provide a feasibility for doing computations in a parallel form. We then will introduce the optimization algorithm by which we can find parameters of the model, and discuss the advantages of that model. Next, we will look at the results obtained from the model for two different types of fiber optic links and two different fitting methods called statistical and algebraic fitting. Finally we will compare the results with the ones obtained from the complex model for C matrix.

### 2.1 Model Structure

The idea we used for finding the simple model was considering a fiber link with length  $L$  as  $N$  cascaded blocks called spans. In other words, instead of modeling the whole fiber link with a complex model, we can consider the fiber as  $N$  cascaded blocks with low computational complexity. The combination of these simple models can be used to model the fiber optic link. By using these blocks, we can reduce computational complexity, and the amount of memory at the expense of the performance degradation of the compensating procedure.

To justify the simplified model we used, we should take a deeper look into the formula for calculating the SPM noise affecting the current data symbol. As it have been seen, the simplified fromula is as follows:

$$SPM[n] = \sum_{m=-M}^{m=M} \sum_{j=-M}^{j=M} C_{m,j} d[n+m] \times d[n+j] \times \text{conj}(d[n+m+j]) \quad (2.1)$$

where  $M$  depends on the type and the length of the fiber optic link, which we intend to model. Also, the accessible amount of memory limits the value of  $M$ . By looking closely to the formula, we can find out that this complex model for SPM noise consists of two main parts:

1. Memory
2. Quadruple Multiplications

these two characteristics should be considered in the new model. We used the following model as simple blocks which can be cascaded to build up the model of whole fiber optic link:

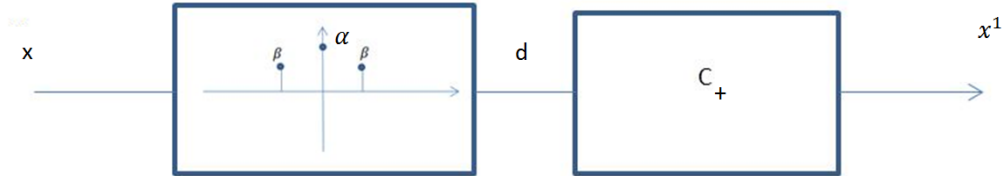


Figure 2.1: Block model

as it can be seen, this model consists of two blocks:

1. Linear filter, which models Chromatic Dispersion: this block models the memory, which can be seen in Equation 2.1. We can see the transfer function of this block in Figure 2.1.
2. Non-linear block ( $C_+$ ): this block is called truncated C matrix and it is supposed to model the Quadruple Multiplications.

mathematical expressions for these two blocks are as follows:

- Linear block:

$$d^i[n] = \alpha x^i[n] + \beta (x^i[n-1] + x^i[n+1]) \quad (2.2)$$

where  $d^i$  is the output of the linear block for i-th span, and  $x^i$  is the input of the i-th span.

- Non-linear block ( $C_+$ ):

$$SPM^i[n] = \sum_{m=-1}^{m=1} \sum_{j=-1}^{j=1} (C_{m,j})_+ d^i[n+m] \times d^i[n+j] \times \text{conj}(d^i[n+m+j]) \quad (2.3)$$

where  $d^i$  is the output of the linear block for i-th span.  $C_+$  is a  $3 \times 3$  matrix with the following form:

$$\begin{bmatrix} 0 & (C_{-1,0})_+ & 0 \\ (C_{0,-1})_+ & (C_{0,0})_+ & (C_{0,1})_+ \\ 0 & (C_{1,0})_+ & 0 \end{bmatrix} \quad (2.4)$$

- Two blocks together:

$$x^{i+1}[n] = d^i[n] + SPM^i[n] \quad (2.5)$$

where  $d^i$  is the output of the linear block for i-th span,  $SPM^i$  is the SPM noise computed by truncated C matrix, and  $x^{i+1}$  is the output of the i-th span.

In order to find the optimum simplified model for the C matrix, we should find optimum values for the following parameters:

- $C_+$  matrix (5 complex numbers).
- $\alpha$  (1 complex number).
- $\beta$  (1 complex number).
- The number of spans (N).

In the next section, we will look at the optimization algorithm in more details. For now, first we want to look at the mathematical expression for the SPM noise computed by this model, then we will discuss why using this model helps us in reducing computational complexity. For finding the SPM noise affecting data samples, first we should look at the

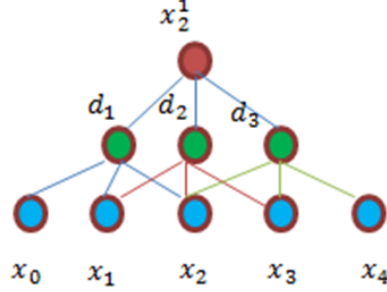


Figure 2.2: Block Levels

structure of the model shown in Figure 2.2. In this Figure, the output of the linear block, Chromatic Dispersion, is shown by green circles and the output of the nonlinear block,  $C_+$ , is shown by a red circle.

To find closed form for the SPM noise, we put formulas 2.2, 2.3, 2.4, and 2.5 together. After simplification, we will obtain the following expressions for the SPM noise affecting middle sample (showed by  $x_2^1$  in Figure 2.2):

$$x_2^1 = d_2 + SPM_2 \quad (2.6)$$

$$\begin{cases} d_1 = \alpha x_1 + \beta (x_0 + x_2) \\ d_2 = \alpha x_2 + \beta (x_1 + x_3) \\ d_3 = \alpha x_3 + \beta (x_2 + x_4) \end{cases} \quad (2.7)$$

$$SPM_2 = (C_{00})_+ d_2 |d_2|^2 + \{(C_{01})_+ + (C_{10})_+\} d_2 |d_3|^2 + \{(C_{-1,0})_+ + (C_{0,-1})_+\} d_2 |d_1|^2 \quad (2.8)$$

$$First Term = (C_{00})_+ \times (\alpha x_2 + \beta (x_1 + x_3)) \times (|\alpha x_2|^2 + 2Real \{ \alpha x_2 \beta^* (x_1^* + x_3^*) \} + |\beta (x_1 + x_3)|^2) \quad (2.9)$$

the whole expression for the first term:

$$\begin{aligned} First Term = & (C_{00})_+ \times \\ & (\alpha x_2 |\alpha x_2|^2 + 2\alpha x_2 Real \{ \alpha x_2 \beta^* (x_1^* + x_3^*) \} + \alpha x_2 |\beta (x_1 + x_3)|^2 + \\ & \beta x_1 |\alpha x_2|^2 + 2\beta x_1 Real \{ \alpha x_2 \beta^* (x_1^* + x_3^*) \} + \beta x_1 |\beta (x_1 + x_3)|^2 + \\ & \beta x_3 |\alpha x_2|^2 + 2\beta x_3 Real \{ \alpha x_2 \beta^* (x_1^* + x_3^*) \} + \beta x_3 |\beta (x_1 + x_3)|^2) \end{aligned} \quad (2.10)$$



other terms:

$$\begin{aligned}
\text{SecondTerm} &= \{(C_{01})_+ + (C_{10})_+\} \times \\
&(\alpha x_2 |\alpha x_3|^2 + 2\alpha x_2 \text{Real} \{\alpha x_3 \beta^* (x_2^* + x_4^*)\} + \alpha x_2 |\beta (x_2 + x_4)|^2 + \\
&\beta x_1 |\alpha x_3|^2 + 2\beta x_1 \text{Real} \{\alpha x_3 \beta^* (x_2^* + x_4^*)\} + \beta x_1 |\beta (x_2 + x_4)|^2 + \\
&\beta x_3 |\alpha x_3|^2 + 2\beta x_3 \text{Real} \{\alpha x_3 \beta^* (x_2^* + x_4^*)\} + \beta x_3 |\beta (x_2 + x_4)|^2)
\end{aligned} \tag{2.11}$$

$$\begin{aligned}
\text{ThirdTerm} &= \{(C_{-1,0})_+ + (C_{0,-1})_+\} \times \\
&(\alpha x_2 |\alpha x_3|^2 + 2\alpha x_2 \text{Real} \{\alpha x_1 \beta^* (x_0^* + x_2^*)\} + \alpha x_2 |\beta (x_0 + x_2)|^2 + \\
&\beta x_1 |\alpha x_1|^2 + 2\beta x_1 \text{Real} \{\alpha x_1 \beta^* (x_0^* + x_2^*)\} + \beta x_1 |\beta (x_0 + x_2)|^2 + \\
&\beta x_3 |\alpha x_1|^2 + 2\beta x_3 \text{Real} \{\alpha x_1 \beta^* (x_0^* + x_2^*)\} + \beta x_3 |\beta (x_0 + x_2)|^2)
\end{aligned} \tag{2.12}$$

after putting all terms together, the mathematical expression for the middle sample will be as follows:

$$\begin{aligned}
x_2^1 &= \alpha x_2 + \beta (x_1 + x_3) + \\
&\left\{ \begin{aligned} &(C_{00})_+ \times \\ &(\alpha x_2 |\alpha x_2|^2 + 2\alpha x_2 \text{Real} \{\alpha x_2 \beta^* (x_1^* + x_3^*)\} + \alpha x_2 |\beta (x_1 + x_3)|^2 + \\ &\beta x_1 |\alpha x_2|^2 + 2\beta x_1 \text{Real} \{\alpha x_2 \beta^* (x_1^* + x_3^*)\} + \beta x_1 |\beta (x_1 + x_3)|^2 + \\ &\beta x_3 |\alpha x_2|^2 + 2\beta x_3 \text{Real} \{\alpha x_2 \beta^* (x_1^* + x_3^*)\} + \beta x_3 |\beta (x_1 + x_3)|^2) \end{aligned} \right\} + \\
&\left\{ \begin{aligned} &\{(C_{01})_+ + (C_{10})_+\} \times \\ &(\alpha x_2 |\alpha x_3|^2 + 2\alpha x_2 \text{Real} \{\alpha x_3 \beta^* (x_2^* + x_4^*)\} + \alpha x_2 |\beta (x_2 + x_4)|^2 + \\ &\beta x_1 |\alpha x_3|^2 + 2\beta x_1 \text{Real} \{\alpha x_3 \beta^* (x_2^* + x_4^*)\} + \beta x_1 |\beta (x_2 + x_4)|^2 + \\ &\beta x_3 |\alpha x_3|^2 + 2\beta x_3 \text{Real} \{\alpha x_3 \beta^* (x_2^* + x_4^*)\} + \beta x_3 |\beta (x_2 + x_4)|^2) \end{aligned} \right\} + \\
&\left\{ \begin{aligned} &\{(C_{0,-1})_+ + (C_{-1,0})_+\} \times \\ &(\alpha x_2 |\alpha x_3|^2 + 2\alpha x_2 \text{Real} \{\alpha x_1 \beta^* (x_0^* + x_2^*)\} + \alpha x_2 |\beta (x_0 + x_2)|^2 + \\ &\beta x_1 |\alpha x_1|^2 + 2\beta x_1 \text{Real} \{\alpha x_1 \beta^* (x_0^* + x_2^*)\} + \beta x_1 |\beta (x_0 + x_2)|^2 + \\ &\beta x_3 |\alpha x_1|^2 + 2\beta x_3 \text{Real} \{\alpha x_1 \beta^* (x_0^* + x_2^*)\} + \beta x_3 |\beta (x_0 + x_2)|^2) \end{aligned} \right\}
\end{aligned} \tag{2.13}$$

As it can be seen, the mathematical expressions calculated for this model are complex. We will introduce a dual form for this model which leads to simpler equations, but first we want to discuss why using this model reduces computational complexity. Suppose that we modeled a fiber optic link by cascading blocks of simplified model two times, Figure 2.3. Also, assume that we have a stream of data, and we want to find the SPM noise affecting each data symbol after passing through the fiber optic link which we try to model. If we want to compute SPM noise samples by using the simplified model, we can build up a hierarchical structure as can be seen in Figure 2.4. It is shown in Figure 2.4 that when a new data sample arrives, shown by circle filled by white color at the bottom, we do not need to do all computations again in order to compute the SPM noise affecting the new

data symbol. In other words, we only need to compute the new values in the states shown by white filled circles, and reuse other values which have been calculated for the SPM noise affecting the previous data symbol.

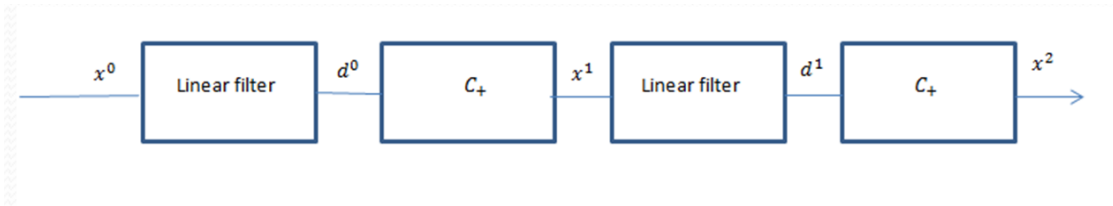


Figure 2.3: Cascading the model two times

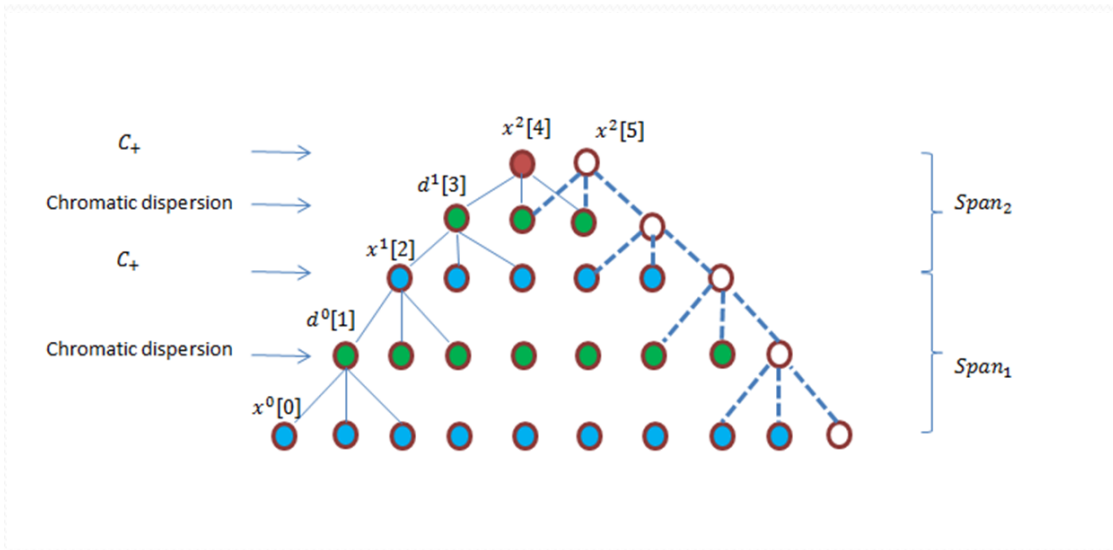


Figure 2.4: Hierarchical structure

To consider other advantages of using N-span model for modeling a fiber optic, first we will introduce the model which leads to simpler equations, and then we will consider advantages of the model. The dual form of the previous model can be obtained simply by changing the order of the blocks. Because the system is not linear, changing order of the blocks will lead to a system with complete different characteristics. The new model can be seen in Figure 2.5. In this new model, first data symbols pass through the non-linear block and then the linear block. It must be considered that we only changed the order of blocks and not the functionality of them.

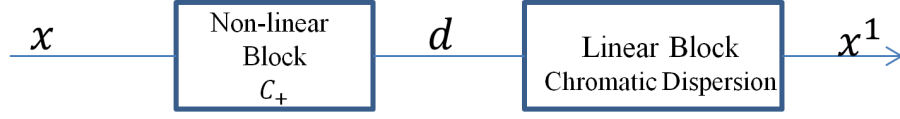


Figure 2.5: Final Model

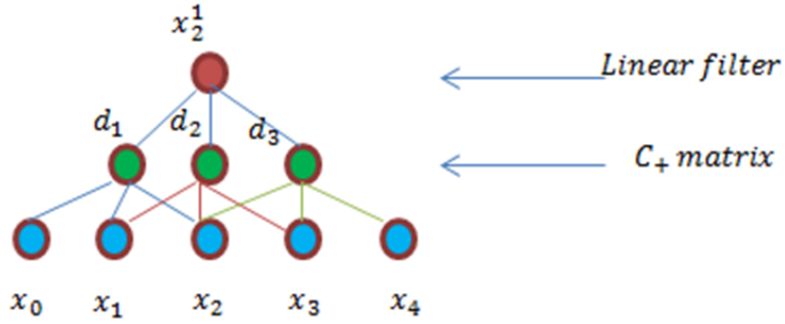


Figure 2.6: Hierarchical structure for the final model

Now we are going to find mathematical equations for the new model. The hierarchical structure for this model can be seen in Figure 2.6. The equations for the middle data symbol after passing through this new structure are as follows:

$$x_2^1 = \alpha d_2 + \beta (d_1 + d_3) \quad (2.14)$$

$$x_2^1 = \alpha x_2 + \alpha SPM_2 + \beta (x_1 + SPM_1 + x_3 + SPM_3) \quad (2.15)$$

$$\begin{cases} SPM_1 = (C_{00})_+ x_1 |x_1|^2 + \{(C_{01})_+ + (C_{10})_+\} \times x_1 |x_2|^2 + \{(C_{0,-1})_+ + (C_{-1,0})_+\} \times x_1 |x_0|^2 \\ SPM_2 = (C_{00})_+ x_2 |x_2|^2 + \{(C_{01})_+ + (C_{10})_+\} \times x_2 |x_3|^2 + \{(C_{0,-1})_+ + (C_{-1,0})_+\} \times x_2 |x_1|^2 \\ SPM_3 = (C_{00})_+ x_3 |x_3|^2 + \{(C_{01})_+ + (C_{10})_+\} \times x_3 |x_4|^2 + \{(C_{0,-1})_+ + (C_{-1,0})_+\} \times x_3 |x_2|^2 \end{cases} \quad (2.16)$$

$$\begin{aligned} x_2^1 = & \alpha x_2 + \beta x_1 + \beta x_3 + \\ & \alpha \times \{ C_{00} x_1 |x_1|^2 + \{ C_{01} + C_{10} \} \times x_1 |x_2|^2 + \{ C_{0,-1} + C_{-1,0} \} \times x_1 |x_0|^2 \} + \\ & \beta \times \left\{ \begin{aligned} & C_{00} \times (x_2 |x_2|^2 + x_3 |x_3|^2) + \{ C_{01} + C_{10} \} \times (x_2 |x_3|^2 + x_3 |x_4|^2) + \\ & \{ C_{0,-1} + C_{-1,0} \} \times (x_2 |x_1|^2 + x_3 |x_2|^2) \end{aligned} \right\} \end{aligned} \quad (2.17)$$

if we compare equation 2.17 with `refmodelleq`, we will find out that the new reordered model leads to a simpler equation which can be computed easily and iteratively for each

data symbol. Although this simplicity will cause less accurate value compared with the first model, in the last section of this chapter we will see that this lack of accuracy is negligible.

Advantages of the reordered model are as follows:

- For finding SPM values, we need to calculate simple equations.
- Equations have iterative structure. It means that we can reuse values from the computed values for the previous data symbol to compute SPM noise for the current one.
- Computations can be done in a parallel form.
- The memory used in this model is much low compared with Formula 2.1.

in the next section we will take a deeper look into these aforementioned advantages.

## 2.2 Advantages of Using N-span Model for Computing SPM Noise in a Fiber Optic Link

In this section first we will see how this N-span model can reduce computational complexity. Then we will discuss how these calculations can be done in a parallel form.

### 2.2.1 Computational Complexity Reduction

This N-span model can reduce computational complexity in two ways:

- To compute SPM noise for a data symbol, we can reuse SPMs computed for the previous ones. In other words, we only need to calculate output of the non-linear block once for the new data symbol.
- In the mathematical equation for computing SPM noise, we can see some terms have been computed. If we store these values, we can reuse them in computing SPM for the new data symbols.

If we want to exploit the pattern exists in the equations of the model and be able to reuse computed values, we should store some specific values after computation. To illustrate these points, let's take a look at the computations done in one span. Figure 2.7 shows the case in which there are 6 data symbols. We want to find out that while we are calculating SPM for data sample  $x_2$ , which values should be stored in order to achieve complexity reduction in computing SPM for the next data symbol named  $x_3$  in the figure.

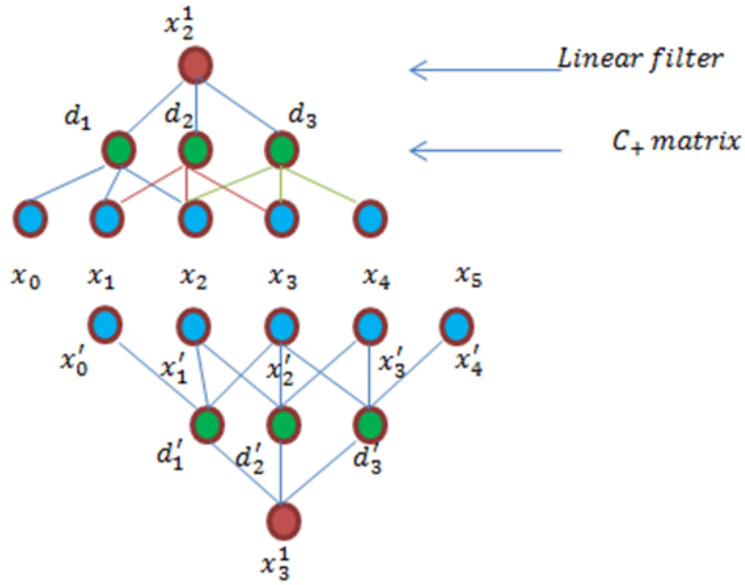


Figure 2.7: Hierarchical structure for two consecutive data samples

The mathematical equations for computing  $x_3^1$  are as follows (in this part, for simplicity, we did not write the symbol  $+$  for the values of  $C_+$ ):

$$x_3^1 = \alpha d'_2 + \beta (d'_1 + d'_3) \quad (2.18)$$

$$x_3^1 = \alpha x'_2 + \alpha SPM'_2 + \beta (x'_1 + SPM'_1 + x'_3 + SPM'_3) \quad (2.19)$$

$$\begin{cases} SPM'_1 = C_{00}x'_1|x'_1|^2 + \{C_{01} + C_{10}\} \times x'_1|x'_2|^2 + \{C_{0,-1} + C_{-1,0}\} \times x'_1|x'_0|^2 \\ SPM'_2 = C_{00}x'_2|x'_2|^2 + \{C_{01} + C_{10}\} \times x'_2|x'_3|^2 + \{C_{0,-1} + C_{-1,0}\} \times x'_2|x'_1|^2 \\ SPM'_3 = C_{00}x'_3|x'_3|^2 + \{C_{01} + C_{10}\} \times x'_3|x'_4|^2 + \{C_{0,-1} + C_{-1,0}\} \times x'_3|x'_2|^2 \end{cases} \quad (2.20)$$

as it can be seen in Figure 2.8, only  $SPM'_3$  should be computed because  $SPM'_1$  and  $SPM'_2$  can be rewritten as  $SPM_2$  and  $SPM_3$  respectively. The values of these unknowns have been calculated in the computations performed for the previous data symbol. Also, the memory allocation which helps us in reusing previous is shown in the figure. As we can see, there is a static memory block which contains fixed values of the model. Also, there are two dynamic memory blocks, one for raw data samples and one for computed values.

$$\left\{ \begin{array}{l} SPM'_1 = C_{00}x_2|x_2|^2 + \{C_{01} + C_{10}\} \times x_2|x_3|^2 + \{C_{0,-1} + C_{-1,0}\} \times x_2|x_1|^2 = SPM_2 \\ SPM'_2 = C_{00}x_3|x_3|^2 + \{C_{01} + C_{10}\} \times x_3|x_4|^2 + \{C_{0,-1} + C_{-1,0}\} \times x_3|x_2|^2 = SPM_3 \\ \boxed{SPM'_3 = C_{00}x_4|x_4|^2 + \{C_{01} + C_{10}\} \times x_4|x_5|^2 + \{C_{0,-1} + C_{-1,0}\} \times x_4|x_3|^2} \end{array} \right. ?$$

Static memory	Dynamic memory 2	Dynamic memory 1
$\alpha$	$x_4$	$SPM_2 + x_2$
$\beta$		$SPM_3 + x_3$
$C_{00}$		$ x_3 ^2$
$C_{0,1} + C_{1,0}$		$ x_4 ^2$
$C_{-1,0} + C_{0,-1}$		

Figure 2.8: Memory Allocation1

Figure 2.9 shows that how we can reuse values stored in memory to calculate  $SPM'_3$ . It shows that for computing the value of  $SPM'_3$ , we only need to do 5 multiplications. Figure 2.10 shows that after computing  $SPM'_3$ , which values should be stored in the memory blocks, and which values should be removed from the memory. After the step shown in Figure 2.10, we have access to the output of the non-linear block of the span.

In Figure 2.11 we can see that how we can calculate the output of the linear block, by reusing the values stored in the memory blocks. Figure 2.12 indicates the last step of rearranging memory. At the end, the memory allocation will be as shown in Figure 2.13. It should be considered that in all steps we only need 5 spots in the dynamic memory blocks and 5 spots in the static one. If we model the whole fiber optic link with  $N$  span, we need  $5N$  dynamic spots in memory and 5 static ones. This amount of memory is affordable for the small values of  $N$ .

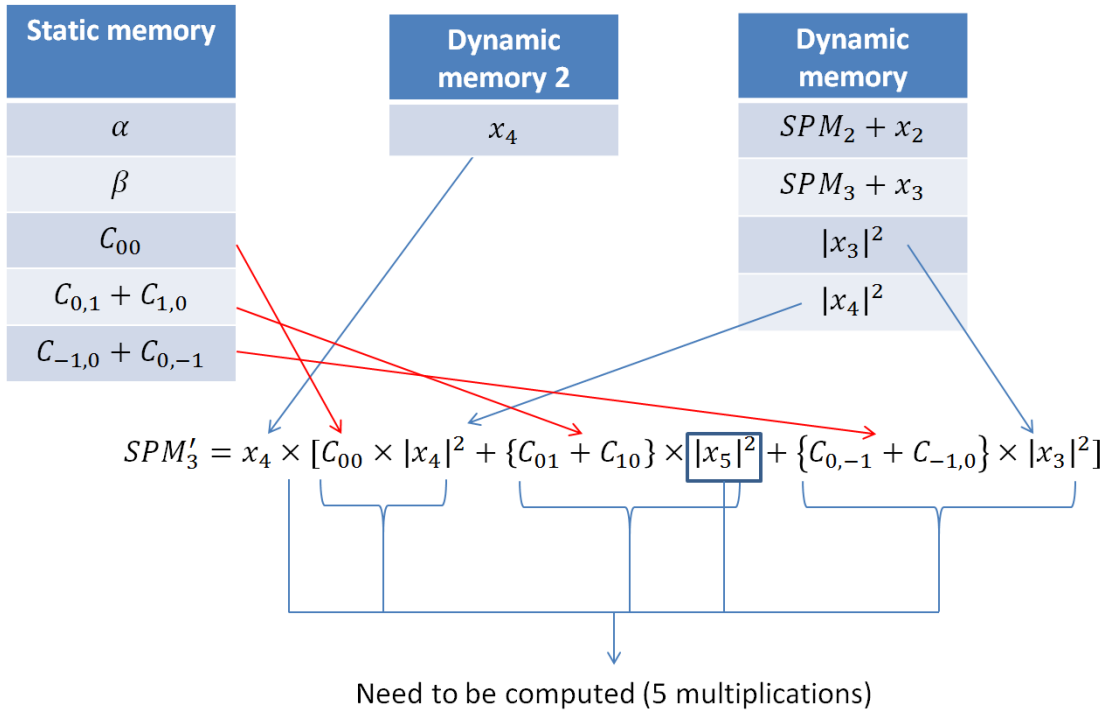


Figure 2.9: Reusing values for computing  $SPM'_3$

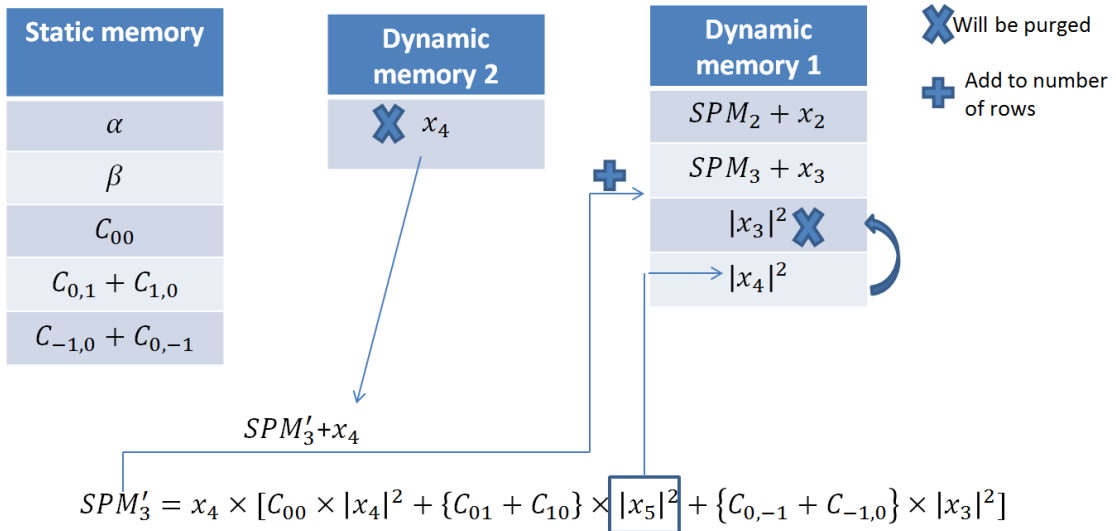


Figure 2.10: Memory flow after calculating  $SPM'_3$

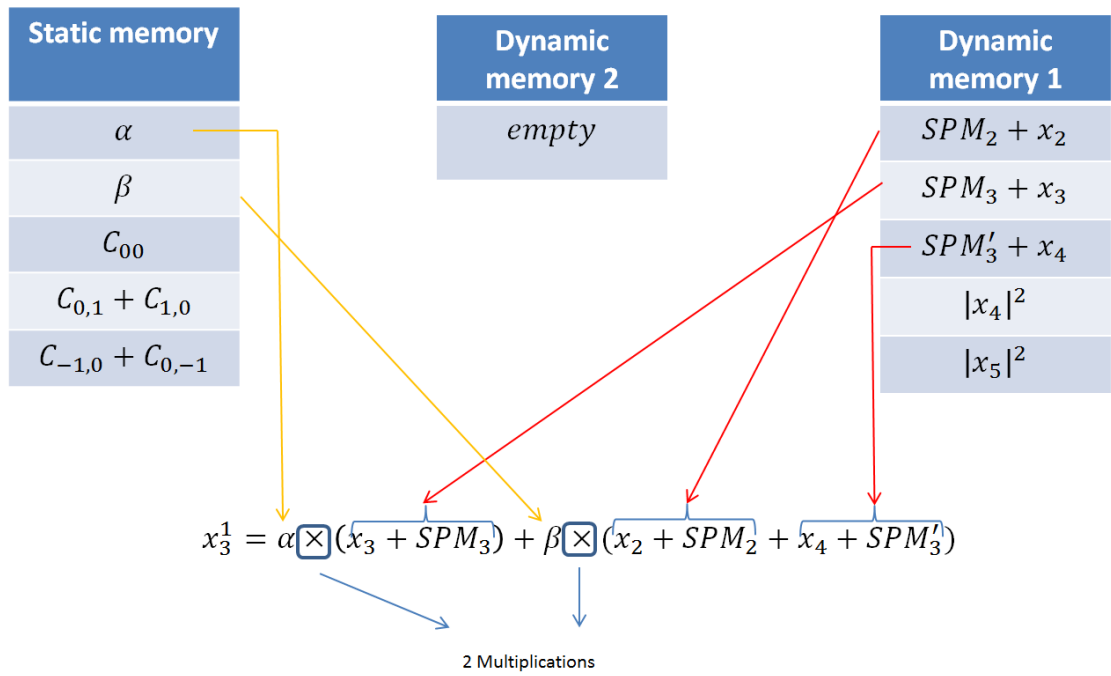


Figure 2.11: Computing the output of the linear block

~~X~~ Will be purged

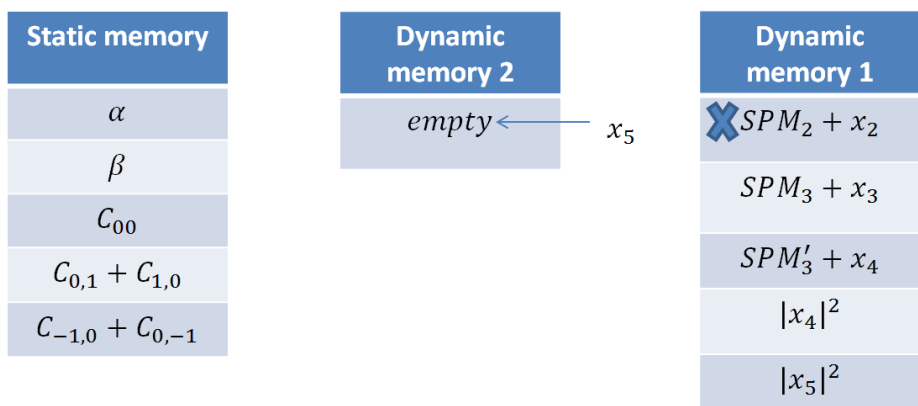


Figure 2.12: Memory flow after computing the output of the span



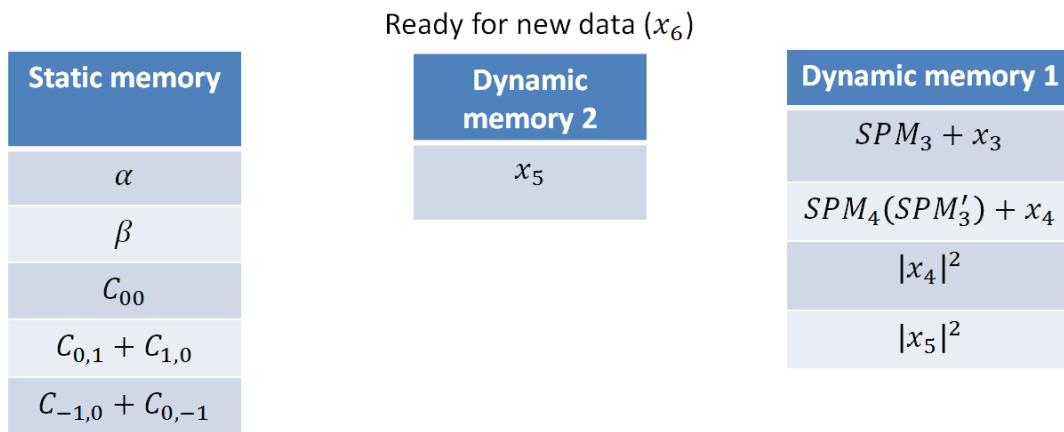


Figure 2.13: Memory allocation when system is ready for the next data sample

## 2.2.2 Parallel Computing

Another good feature of the N-span model for a fiber optic cable is that in this scheme we can do computations in parallel form. This parallel computation is possible because of the fact that for computing the corresponding values to a node, we only need to know the values of the adjacent nodes. This fact can be seen in the hierarchical structure shown in Figure 2.6. To exploit this feature, we can allocate computations of each span to a processor. To illustrate that, suppose we modeled a fiber optic cable with 5-span model. As it is shown in Figure 2.14, the linear and non-linear blocks are considered as a hidden layer for the sake of simplicity. The blue filled circles are the nodes which their values have been stored in memory, therefore we can use them in calculations. Also, processors have an access to those nodes' values. Grey filled circles are the nodes which their values have been purged from memory blocks. Figure 2.15 shows the first clock cycle of this scheme. From the first cycle to the fifth one, only one processor works because all the values needed for computations in the second span have not been ready yet. After fifth iteration, the second processor will start to work as it is shown in Figure 2.16. It should be noticed that the first processor does not stop working; it will continue computing in parallel form and compute values needed for the second processor. After 10th iteration the third processor will start working and so on, shown in Figure 2.17. In 5-span model, after 20th clock cycle all processor will be working and computing values needed for the next span in parallel, shown in Figure 2.18, and 2.19. The output of the last span, computed by processor 5, is the output of model which should be approximately same as the result of SPM noises equation, equation 2.1.

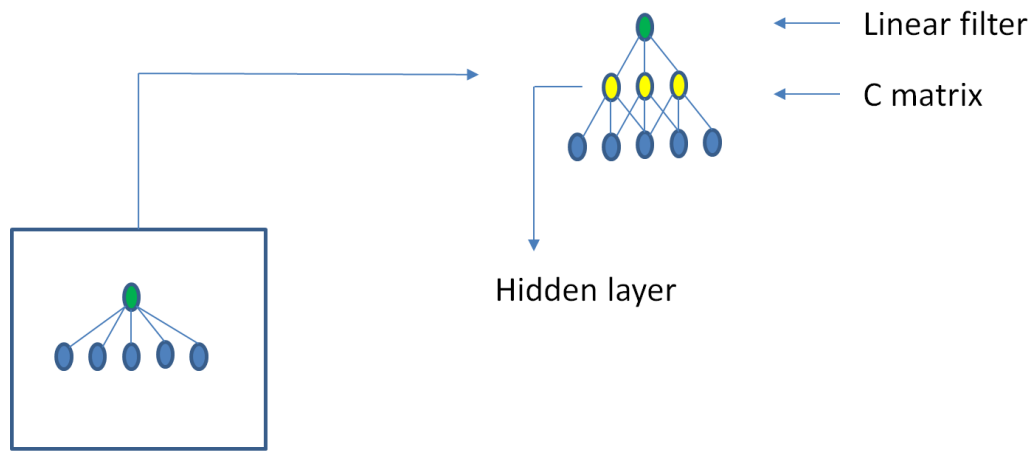


Figure 2.14: Simplified Structure

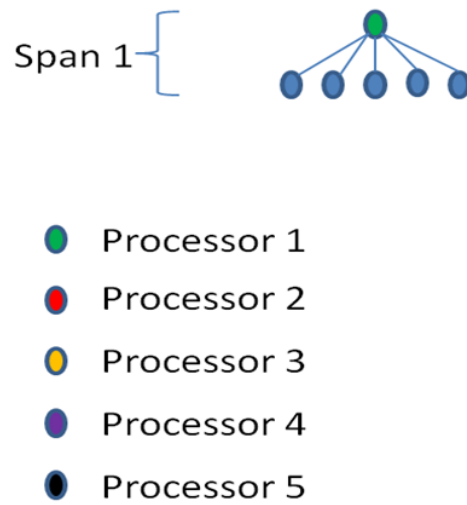


Figure 2.15: First clock cycle

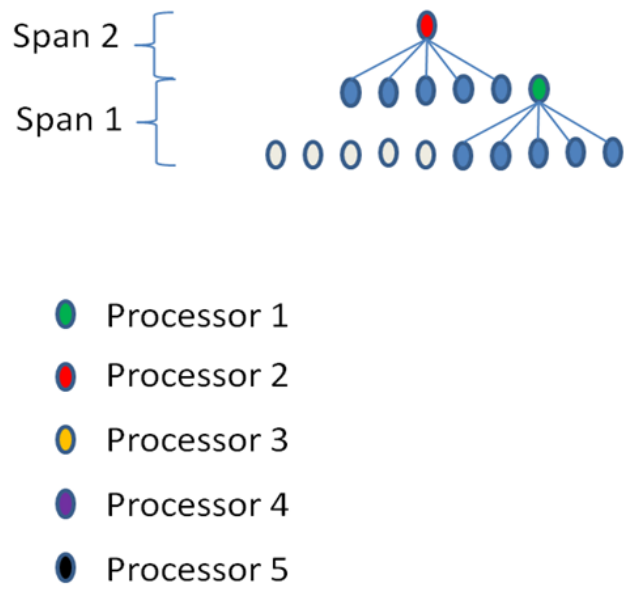
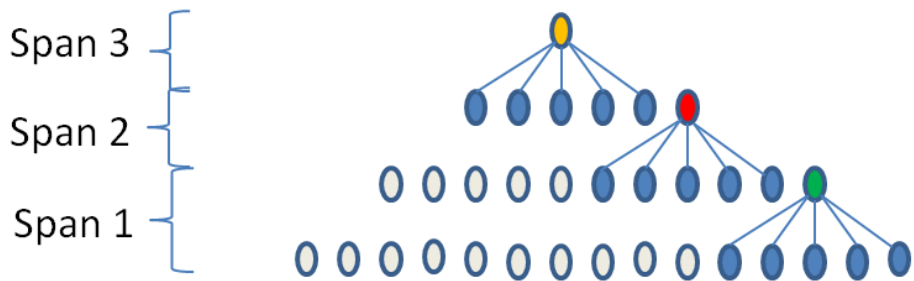


Figure 2.16: Sixth clock cycle.



- Processor 1
- Processor 2
- Processor 3
- Processor 4
- Processor 5

Figure 2.17: 11th clock cycle

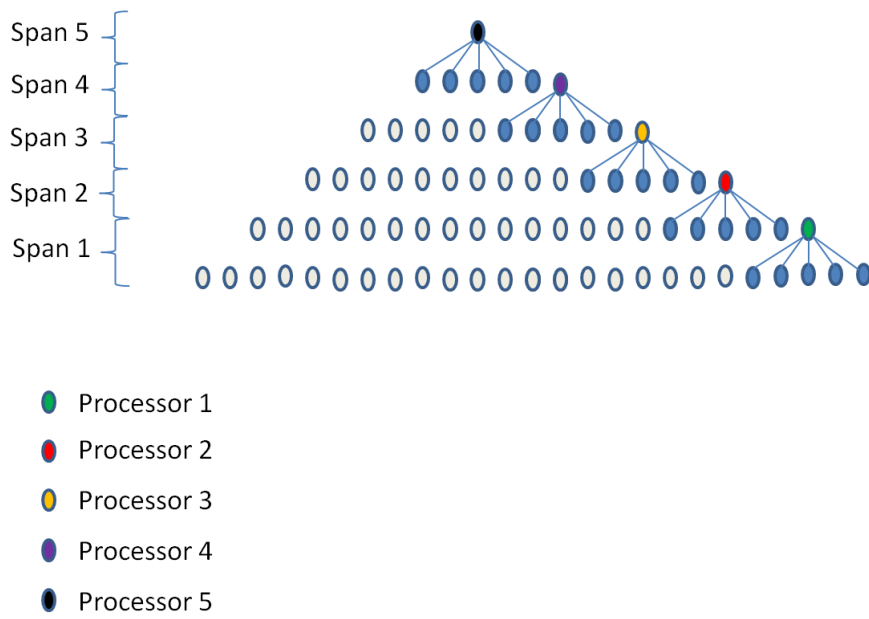


Figure 2.18: 21th clock cycle

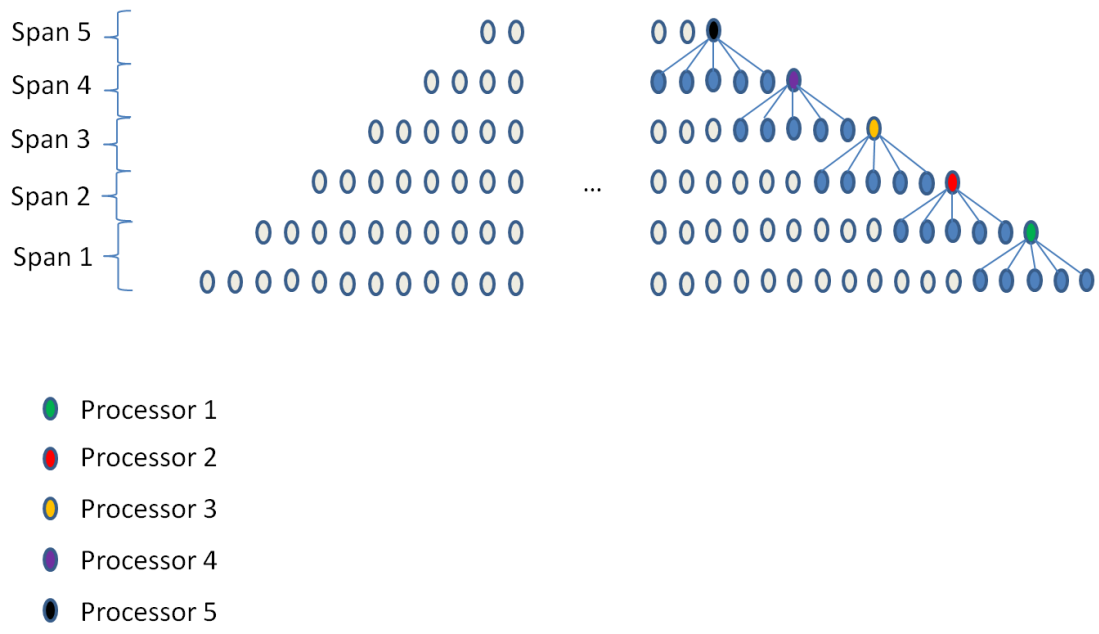


Figure 2.19: Processors continue working in parallel

## 2.3 Optimization Algorithm

In this section first we will look at the algorithm used for finding the parameters of the model, mentioned on page 32. Then we will discuss about advantages of using this method. At the end, we will look at two methods used for fitting the N-span model to the equation 2.1, fitting the model based on statistical features or algebraic formula.

### 2.3.1 Nelder-Mead Simplex Algorithm

For the optimization problem, we used Nelder-Mead simplex algorithm. This method proposed in Nelder and Mead's article "A simplex method for function minimization" [12]. In this article authors introduce a method for minimizing a cost function. To shed light on the way that this algorithm works, suppose that we have a cost function with N free parameters to be set in order to minimizing the cost function. For cost function with N parameters, this algorithm works in N dimensional space and N+1 test points. At first, we build up a random polytope in N dimensional space with N+1 vertices, and we try to shrink that polytope until it tends to one point. That point will be the optimum set of parameters. The steps of this algorithm are as follows:

- The built polytope has N+1 vertices which are test points, each vertex corresponds to N parameters which can be put in the cost function. First we compute cost function at those vertices, and sort them in ascending order. We name the vertex with the highest value as  $V_h$ , and the one with lowest value as  $V_l$ . We name corresponding values obtained from putting  $V_h$  and  $V_l$  in cost function as  $f_h$  and  $f_l$ .
- We find the centroid of the all vertices except  $V_h$ . This centroid can be calculated by following equation:

$$\bar{V} = \frac{\sum_{i=1}^{N+1} V_i}{N}, \text{ and } i \neq h \quad (2.21)$$

- Now we should replace  $V_h$  by a new vertex. This new vertex will be calculated by three operations, and choosing the proper operation between all three will be done by considering conditions. The mentioned operations are as follows:

1. Reflection: in this operation we put the reflection of  $V_h$  instead of that. The reflection of  $V_h$  is called  $V^*$  and the coordinates of that can be calculated by the equation

$$V^* = (1 + \alpha)\bar{V} - \alpha V_h \quad (2.22)$$



$\alpha$  is called reflection coefficient and is a positive constant. If  $f^*$  lies between  $f_h$  and  $f_l$ , then we replace  $V_h$  by  $V^*$  and start again with the new simplex.

2. Expansion: if  $f^* < f_l$ , then we replace  $V^*$  by  $V^{**}$  which can be calculated by equation

$$V^{**} = (1 - \gamma)\bar{V} + \gamma V^* \quad (2.23)$$

$\gamma$  is called the expansion coefficient which is greater than 1. Now the result of cost function after putting  $V^{**}$  inside that is called  $f^{**}$ . If  $f^{**} < f_l$  we replace  $V_h$  by  $P^{**}$  and continue with the new simplex, but if  $f^{**} > f_l$  it means that expansion was not successful. Then, we replace  $V_h$  by  $V^*$  and continue with the new simplex.

3. Contraction: after finding if  $f^* > f_i$  for  $i \neq h$  we should choose between  $f_h$  and  $f^*$  the one with lowest value and name that  $V_h$ . now we should form  $V^{**}$  as equation

$$V^{**} = (1 - \beta)\bar{V} + \beta V_h \quad (2.24)$$

$\beta$  is called contraction coefficient and  $0 < \beta < 1$ . We replace  $V_h$  by  $V^{**}$  if  $f^{**} \leq \min(f_h, f^*)$ . In the case where  $f^{**} \leq \min(f_h, f^*)$ , it means that all three operations were unsuccessful. In that case we replace all  $V_i$ 's by  $\frac{V_i + V_i}{2}$  and continue with the new simplex. This operation is called Shrink.

- The algorithm could be terminated by setting a fixed number for the maximum number of iterations or a bound for the variance of the cost function at the vertices of the simplex. The MATLAB code used for implementing this algorithm, is shown in Appendix A.

you can see the flowchart of this algorithm in Figure 2.21 [12]. Also, you can see the geometric interpretation of three operations in Figure 2.20. The only point which should be considered is the importance of choosing the first simplex properly. First guess has a high importance in this algorithm. At first, we need  $N+1$  vectors with  $N$  elements as the vertices of the simplex. There are many methods for choosing that initial simplex guess properly. We used Pfeffer's method (credit L.Pfeffer at Stanford). The first step of this method is generating a random vertex, a point in  $N$  dimensional space. Next step is repeating that vertex, vector with length  $N$ ,  $N+1$  times and build a  $(N + 1) \times N$  matrix. We cannot use  $N+1$  same vertices for the algorithm. We modify elements on the main diagonal of the matrix. For non-zero elements, we multiply them by a fixed number which is  $1 + \delta$ . For zero elements, we put a fixed number  $\delta_0$  instead of them. After doing this procedure, we will get  $N+1$  vectors each with  $N$  elements. This optimization method has some main features as follows:

- We do not need to compute derivatives for finding the minimum point in  $N$  dimensional space. In other methods, like gradient descent, we need to compute derivative and take a step in the direction in which the slope of the cost function is negative and with the highest absolute value. We always need to approximate the derivative of the cost function and evaluate the cost function many times, which is not an easy task.
- Computing complexity of this method will be scaled less by scaling dimension compared with other methods. For example in gradient descent method, we need to compute cost function  $N$  times in each iteration to be able to compute derivative; however, in this method we only need to evaluate cost function once in most of the iterations.
- Same as many other optimization methods, in this method initial guess has a considerable effect on the final result. If we do not choose initial guess properly, the algorithm may stuck into a local minimum. For solving this problem, we used the known  $C$  matrix to guess the good initial values for the model's parameters. Also, we perturbed the result and performed the algorithm again after convergence.

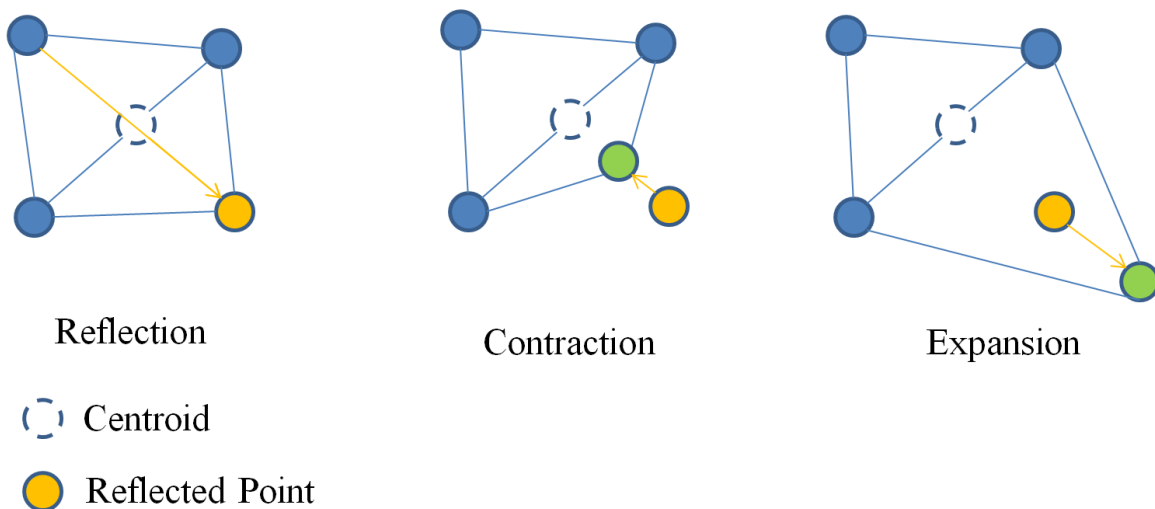


Figure 2.20: The geometric interpretation of the operations

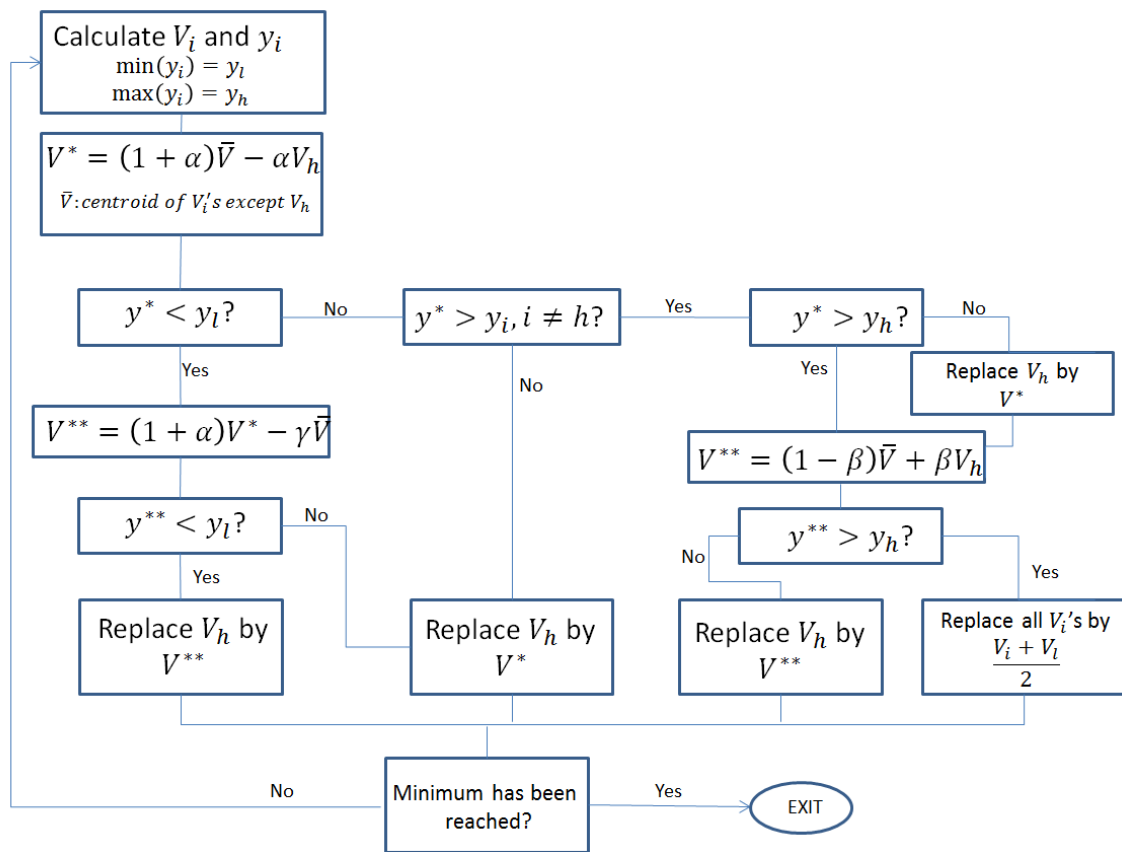


Figure 2.21: Nelder-Mead Simplex Algorithm flowchart

### 2.3.2 Optimization Parameters

In this subsection we will take a look into the optimization parameters. As we discussed on page 32 we have 15 free parameters (7 complex numbers and one natural number):

- $C_+$  matrix (5 complex numbers).
- $\alpha$  (1 complex number).
- $\beta$  (1 complex number).
- The number of spans (N).

To reduce the number of parameters, we can make the following assumptions:

- After running the optimization method several times, we found out that the best value for number of spans is 10. Changing the number of spans do not change the accuracy too much.
- By looking at the given C matrix, we found out that the imaginary parts of  $C_{0,1}, C_{1,0}, C_{-1,0}, C_{0,-1}$  are the same. Because of this symmetry, we can consider only one parameter for the imaginary parts of  $(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ . This assumption will reduce optimization parameters.

The final optimization parameters are as follows(We have shown  $\alpha, \beta$  by  $Ch_0$  and  $Ch_1$  respectively):

1. The real part of  $Ch_0$
2. The imaginary part of  $Ch_0$
3. The real part of  $Ch_1$
4. The imaginary part of  $Ch_1$
5. The real part of  $(C_{0,0})_+$
6. The imaginary part of  $(C_{0,0})_+$
7. The imaginary part of  $(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$

8. The real part of  $(C_{0,-1})_+$
9. The real part of  $(C_{-1,0})_+$
10. The real part of  $(C_{0,1})_+$
11. The real part of  $(C_{1,0})_+$

### 2.3.3 Cost Functions

The last thing needed to be discussed in this subsection is the cost function which will be optimized. For statistical and algebraic fitting, we used different cost functions.

- **Statistical Fitting:** First M data symbols will be generated, and passed through the formula obtained from the Equation 2.1. A value for SPM noise will be computed for each data symbol. Now the cost function will be defined as mean squared error between the SPMs obtained from the model and the ones obtained from Equation 2.1. The MATLAB code used for calculating this cost function, can be seen in Appendix B.

$$f = E\{|SPM_{model} - SPM_{formula}|^2\} \quad (2.25)$$

This method is called statistical fitting since we used many data symbols generated statistically in order to fit the model. In other words, model's parameters will be found statistically. The beauty of this method is that instead of fitting model to the SPMs obtained from Equation 2.1, we can fit model in a way that the output of the model be the SPMs obtained after one time compensating the constellation points. It means that we can build up or model in a way that we will be able to do n times compensation with the output of that. In other words, if we fit the model to the output of the fourth time compensated SPMs, we can do four-time compensation which needs a huge amount of computations by passing data symbols through the N-span model. We can see the compensation scheme in Figure 2.22. This figure shows that in order to compensate SPM noise for constellation points four times, we need to compute SPM four times.

- **Algebraic Fitting:** In this method, first we try to find the algebraic formula for the output of the N-span model in terms of 11 optimization parameters. After that, we can define a cost function which is the mean squared error between the terms of the formula obtained from the model and the terms which can be seen in Equation 2.1. To find the terms in algebraic formula obtained from the model, we defined a

Compensation Scheme:

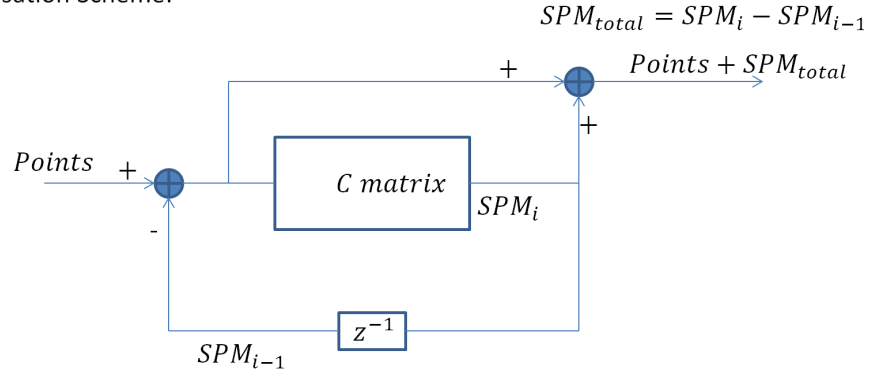


Figure 2.22: Compensation Scheme

matrix called H which contains exponents of the data symbols or the conjugate of them. Each row shows a term in the algebraic formula. Also, we defined a column matrix containing the factor which multiplies in that term, called Y. For each span in the model, we modified those two matrices. Because in the equation 2.1 the sum of exponents is equal to 3, in each iteration we kept the rows in matrix H for which the sum of components is less than 3. In the last iteration we kept all the rows. The cost function is as follows:

$$f = E\{|Y - (\text{Corresponding terms in the Formula})|^2\} \quad (2.26)$$

the MATLAB code used for calculating this cost function, is shown in Appendix C.

The advantage of the statistical fitting method is that we can fit the model to the response of the n-time compensated data samples, but in the algebraic method because of the complexity of the formula we cannot find the closed form formula for the n-time compensated case. The disadvantage of the statistical method is that we need too many data symbols to obtain an accurate model; however, in algebraic method we do not need to generate any data symbols. Also, we can compute the terms showing in the N-span model's formula offline.

## 2.4 Results Obtained from Statistical Fitting

In this section, the results obtained from using the discussed statistical fitting for two different types of fiber optic links will be discussed. These two types are Non Dispersion-

shifted Fiber (NDSF) and Enhanced-Large Effective Area Fiber (ELEAF). For each type of fiber optic cable, we will look at the results for 3 different cases:

1. N-span model with the structure shown in Figure 2.1. In this case data samples first pass through the linear filter and after that the non-linear one.
2. same structure as previous case, but higher noise rate. In other words, in this case all elements of given C matrix was multiplied by 9 to show that the method works for a higher level of SPM noise as well.
3. N-span model with the structure shown in Figure 2.5. In this case data samples first pass through the non-linear filter and after that the linear one.

at the end of this section, we will show how the N-span model can help us in the compensating procedure.

### 2.4.1 Model Fitting Results

The results obtained for different cases mentioned on page 45 can be see in the following tables. Results obtained for the first case can be seen in Tables 2.1 to 2.4. Tables 2.1 and 2.3 show the parameters of the model to which the algorithm has been converged, and Tables 2.2 and 2.4 indicate the value of mean squared error for 3 different batch of test data samples for the two mentioned types of fiber optic links. Results for the second case can be seen in Tables 2.5 to 2.8, and for the third case is shown in Tables 2.9 to 2.12. As it can be seen, the mean squared error is much less than the variance of SPM noise for all the cases and both fiber optic link types. Also, these tables show that the error presents in our model is negligible. The effectiveness of the model will become more obvious in the next subsection. In the next subsection, the results obtained from doing compensation by using the model will be shown. The MATLAB code used for testing obtained models, can be seen in Appendix D.

Parameter	Value
$Ch_0$ real part	1.002759695603447e+00
$Ch_0$ imaginary part	1.905774255578641e-02
$Ch_1$ real part	-4.103173108615886e-06
$Ch_1$ imaginary part	2.655393085086261e-05
$(C_{0,0})_+$ real part	6.834800468002517e-22
$(C_{0,0})_+$ imaginary part	5.700352795373526e-03
$(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ imaginary part	6.960992528122716e-04
$(C_{0,-1})_+$ real part	-1.904621702703570e-21
$(C_{-1,0})_+$ real part	-1.012850925209589e-22
$(C_{0,1})_+$ real part	7.895281727145560e-22
$(C_{1,0})_+$ real part	2.885560014804130e-22

Table 2.1: Model's Parameters for the first case (Linear block first, NDSF fiber)

	$E \left\{ abs(SPM_{Cmatrix})^2 \right\}$	$E \left\{ abs(SPM_{model} - SPM_{Cmatrix})^2 \right\}$
Test1	3.16e-2	6.22e-4
Test2	3.19e-2	6.1e-4
Test3	3.12e-2	5.92e-4

Table 2.2: Mean squared error and the level of SPM noise for 3 batches of test data samples (Linear block first, NDSF fiber)



Parameter	Value
$Ch_0$ real part	1.002457916312066e+00
$Ch_0$ imaginary part	6.510715052869914e-03
$Ch_1$ real part	1.321422122108467e-05
$Ch_1$ imaginary part	-1.176524585524196e-06
$(C_{0,0})_+$ real part	1.196134340278425e-21
$(C_{0,0})_+$ imaginary part	1.090428879479461e-02
$(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ imaginary part	4.158828944716305e-03
$(C_{0,-1})_+$ real part	6.225128262003851e-22
$(C_{-1,0})_+$ real part	4.809190882481514e-21
$(C_{0,1})_+$ real part	8.792501234831203e-23
$(C_{1,0})_+$ real part	-1.068619721005762e-22

Table 2.3: Model's Parameters for the first case (Linear block first, ELEAF fiber)

	$E \left\{ \text{abs}(SPM_{Cmatrix})^2 \right\}$	$E \left\{ \text{abs}(SPM_{model} - SPM_{Cmatrix})^2 \right\}$
Test1	2.75e-2	7.26e-4
Test2	2.84e-2	7.53e-4
Test3	2.72e-2	7.18e-4

Table 2.4: Mean squared error and the level of SPM noise for 3 batches of test data samples (Linear block first, ELEAF fiber)

Parameter	Value
$Ch_0$ real part	1.08250274280802e+000
$Ch_0$ imaginary part	82.9772648294390e-003
$Ch_1$ real part	9.61861389061846e-006
$Ch_1$ imaginary part	47.4860202739653e-006
$(C_{0,0})_+$ real part	-2.22676188093036e-021
$(C_{0,0})_+$ imaginary part	8.65460177902973e-003
$(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ imaginary part	3.25861930794958e-003
$(C_{0,-1})_+$ real part	-4.52245126720480e-021
$(C_{-1,0})_+$ real part	10.8881692748081e-021
$(C_{0,1})_+$ real part	653.440961468324e-024
$(C_{1,0})_+$ real part	-795.907061222345e-024

Table 2.5: Model's Parameters for the second case (Linear block first, fitted to  $9 \times C$  matrix, NDSF fiber)

	$E \left\{ \text{abs}(SPM_{Cmatrix})^2 \right\}$	$E \left\{ \text{abs}(SPM_{model} - SPM_{Cmatrix})^2 \right\}$
Test1	2.54	8.4e-2
Test2	2.53	7.9e-2
Test3	2.54	8.3e-2

Table 2.6: Mean squared error and the level of SPM noise for 3 batches of test data samples (Linear block first, fitted to  $9 \times Cmatrix$ , NDSF fiber)

Parameter	Value
$Ch_0$ real part	1.08442502280802e+000
$Ch_0$ imaginary part	82.67726484390e-003
$Ch_1$ real part	9.324189061846e-006
$Ch_1$ imaginary part	-68.9660202739653e-006
$(C_{0,0})_+$ real part	-20.82676188093036e-018
$(C_{0,0})_+$ imaginary part	8.9946017902973e-003
$(C_{0,1})_+$ , $(C_{1,0})_+$ , $(C_{-1,0})_+$ , $(C_{0,-1})_+$ imaginary part	3.225861930794958e-003
$(C_{0,-1})_+$ real part	-4.66562126720480e-018
$(C_{-1,0})_+$ real part	2.9681692748081e-018
$(C_{0,1})_+$ real part	3.914406468324e-018
$(C_{1,0})_+$ real part	11.340706122345e-018

Table 2.7: Model's Parameters for the second case (Linear block first, fitted to  $9 \times Cmatrix$ , ELEAF fiber)

	$E \left\{ \text{abs}(SPM_{Cmatrix})^2 \right\}$	$E \left\{ \text{abs}(SPM_{model} - SPM_{Cmatrix})^2 \right\}$
Test1	2.22	1.24e-1
Test2	2.219	1.17e-1
Test3	2.23	1.16e-2

Table 2.8: Mean squared error and the level of SPM noise for 3 batches of test data samples (Linear block first, fitted to  $9 \times Cmatrix$ , ELEAF fiber)

Parameter	Value
$Ch_0$ real part	1.002810211706542e+00
$Ch_0$ imaginary part	2.184768317500213e-02
$Ch_1$ real part	-9.498984090443774e-06
$Ch_1$ imaginary part	4.820760230412762e-06
$(C_{0,0})_+$ real part	1.083998219970828e-21
$(C_{0,0})_+$ imaginary part	1.602542695243064e-03
$(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ imaginary part	8.226441599186519e-04
$(C_{0,-1})_+$ real part	-7.953149956168660e-22
$(C_{-1,0})_+$ real part	2.837351665943020e-21
$(C_{0,1})_+$ real part	-7.019077299722014e-21
$(C_{1,0})_+$ real part	6.950867439930186e-22

Table 2.9: Model's Parameters for the third case (Non-linear block first, NDSF fiber)

	$E \left\{ abs(SPM_{Cmatrix})^2 \right\}$	$E \left\{ abs(SPM_{model} - SPM_{Cmatrix})^2 \right\}$
Test1	3.15e-2	5.73e-4
Test2	3.09e-2	5.55e-4
Test3	3.05e-2	5.73e-4

Table 2.10: Mean squared error and the level of SPM noise for 3 batches of test data samples (Non-linear block first, NDSF fiber)

Parameter	Value
$Ch_0$ real part	1.001306321880084e+00
$Ch_0$ imaginary part	2.594624555857188e-03
$Ch_1$ real part	-7.045662081160751e-06
$Ch_1$ imaginary part	-7.531258349854561e-06
$(C_{0,0})_+$ real part	1.414400856748126e-21
$(C_{0,0})_+$ imaginary part	5.929180799934708e-03
$(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ imaginary part	2.208500192019164e-03
$(C_{0,-1})_+$ real part	-1.981347169777232e-23
$(C_{-1,0})_+$ real part	3.774529601951591e-21
$(C_{0,1})_+$ real part	3.017761938028433e-22
$(C_{1,0})_+$ real part	1.288737396392958e-21

Table 2.11: Model's Parameters for the third case (Non-linear block first, ELEAF fiber)

	$E \left\{ \text{abs}(SPM_{Cmatrix})^2 \right\}$	$E \left\{ \text{abs}(SPM_{model} - SPM_{Cmatrix})^2 \right\}$
Test1	2.77e-2	7.47e-4
Test2	2.82e-2	7.39e-4
Test3	2.81e-2	7.44e-4

Table 2.12: Mean squared error and the level of SPM noise for 3 batches of test data samples (Non-linear block first, ELEAF fiber)

## 2.4.2 Fitting the Model to Compensated Data Samples

As we discussed before, one of the promising ways for reducing SPM noise is doing compensation before launching data samples through the fiber optic link channel. Doing the procedure of compensation repeatedly, can reduce SPM noise. The change in the value of the variance of SPM noise can be seen in Tables 2.13, and 2.14 for the two different types of fiber optic links. Also, the received constellation points after fiber optic link are shown in Figures 2.23, and 2.24. It can be seen that by repeating the compensation procedure, the radius of constellation points clouds will be reduced and the center of them will be nearer to the exact values of 16-QAM constellation points. It means less noise variance and average value.

Although repeating the compensating procedure helps us in reducing SPM noise drastically, it has a huge computational cost if it is done by using large  $C$  matrix and Equation 2.1. Our goal in this subsection is to show how we can do compensation procedure by using a model fitted to the results of compensation. The MATLAB code used for implementing compensation procedure, can be seen in Appendix E.

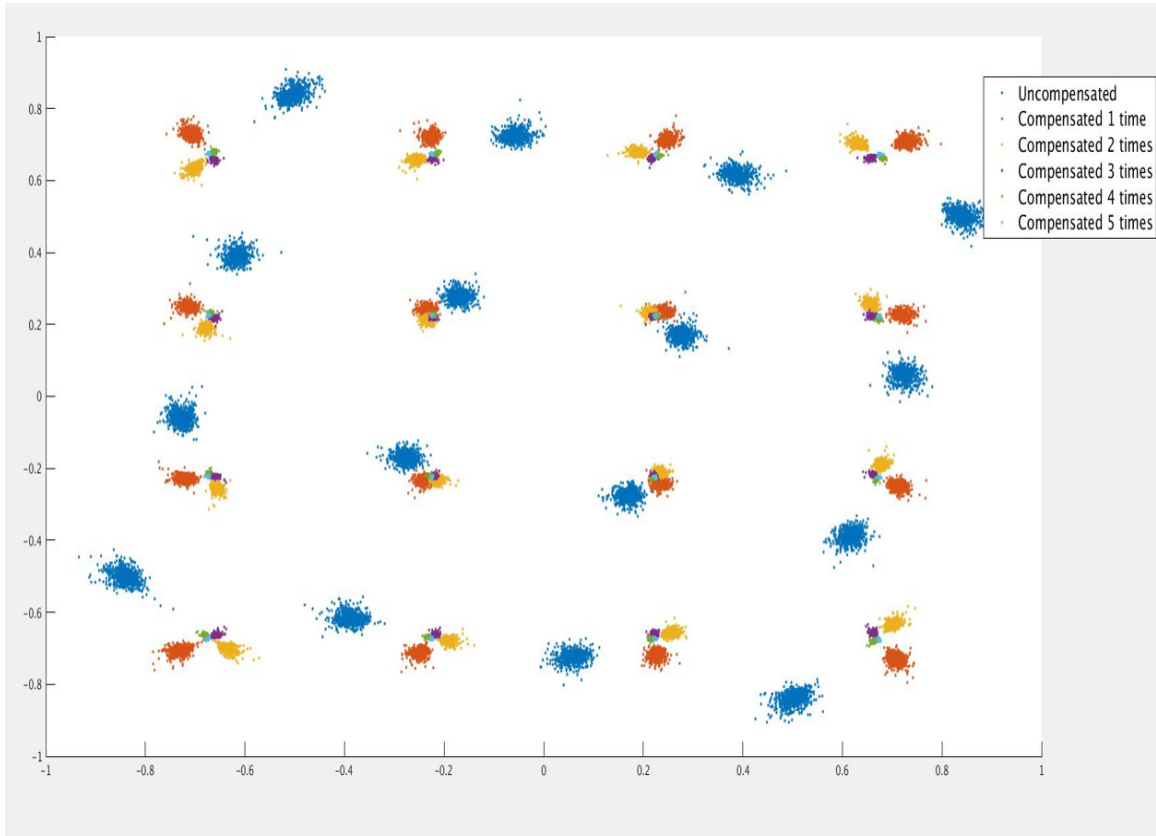


Figure 2.23: 16-QAM constellation points after passing through an NDSF fiber link

	The variance of SPM noise
Uncompensated	3.27e-2
Compensated 1 time	2.94e-3
Compensated 2 times	1.73e-3
Compensated 3 times	1.77e-4
Compensated 4 times	7.04e-5
Compensated 5 times	8.87e-6

Table 2.13: The varinace of SPM noise after doing compensation (NDSF fiber link)

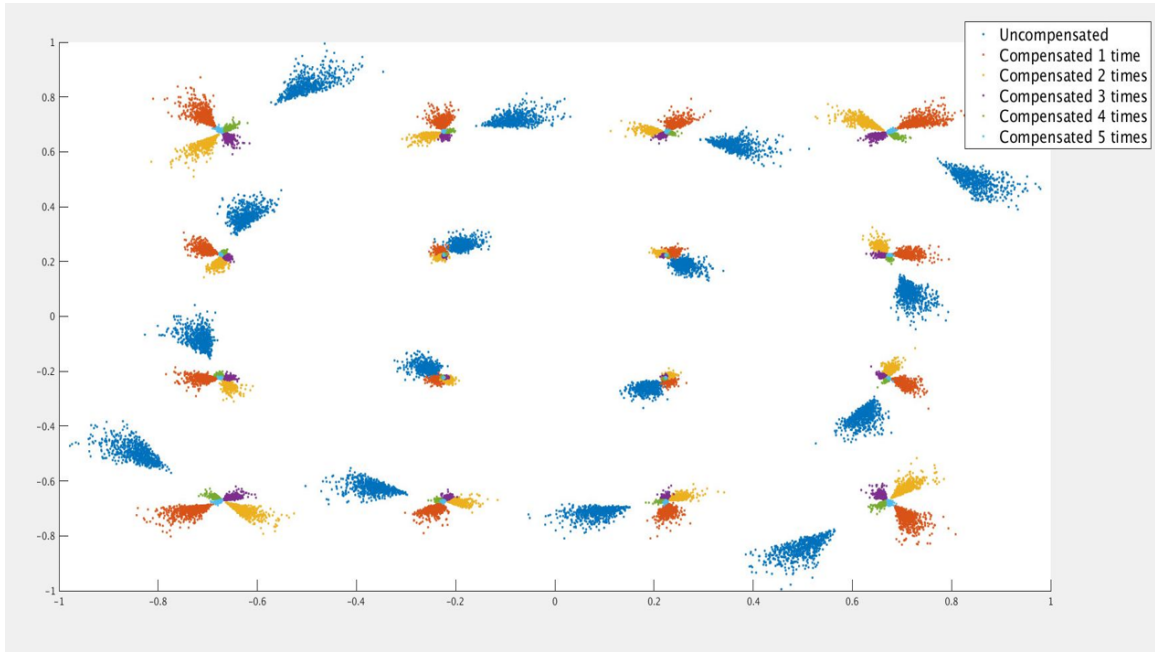


Figure 2.24: 16-QAM constellation points after passing through an NDSF fiber link

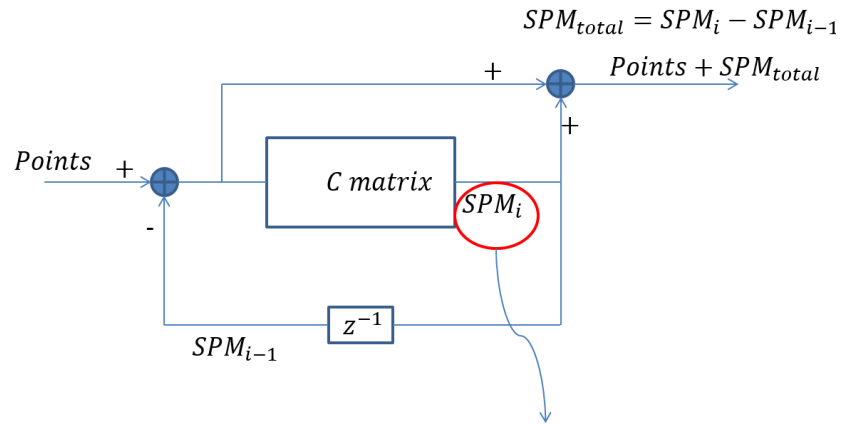
	The variance of SPM noise
Uncompensated	2.87e-2
Compensated 1 time	2.54e-3
Compensated 2 times	1.33e-3
Compensated 3 times	1.35e-4
Compensated 4 times	6.75e-5
Compensated 5 times	8.12e-6

Table 2.14: The variance of SPM noise after doing compensation (ELEAF fiber link)

The idea in this part is finding the model's parameters such that the cost function, mean squared error between SPM values obtained after repeating compensation procedure  $i$  times and passing data samples through the model once, will be minimized. In Figure 2.25 the compensation procedure can be seen. Figure 2.26, shows 4 different cases which are as follows:

- First path: Passing data samples through  $C$  matrix, doing subtraction (Compensation), and passing compensated data samples through  $C$  matrix, which is the model of fiber.
- Second path: Passing data samples through the model fitted to the output of the  $C$  matrix, doing subtraction (Compensation), and passing compensated data samples through  $C$  matrix, which is the model of fiber. This path has been discussed before. Here the only goal is reducing computational complexity by using the fitted model instead of the  $C$  matrix equation itself.
- Third path: Passing data samples through the model fitted to the result of equation 2.1 after doing compensation one time, doing subtraction (Compensation), and passing compensated data samples through  $C$  matrix, which is the model of fiber. In this path the goal is finding a model which can do compensation once with computational complexity of the simple model. In other words, through this path we are doing compensation twice with the complexity of calculating the result of N-span model once.
- Fourth path: Passing data samples through the model fitted to the result of equation 2.1 after doing compensation two times, doing subtraction (Compensation), and passing compensated data samples through  $C$  matrix, which is the model of fiber. In this path the goal is finding a model which can do compensation two times with computational complexity of the simple model. In other words, through this path we are doing compensation three times with the complexity of calculating the result of N-span model once.





We fit a model to this SPM for different "i"s

Figure 2.25: Compensation Scheme

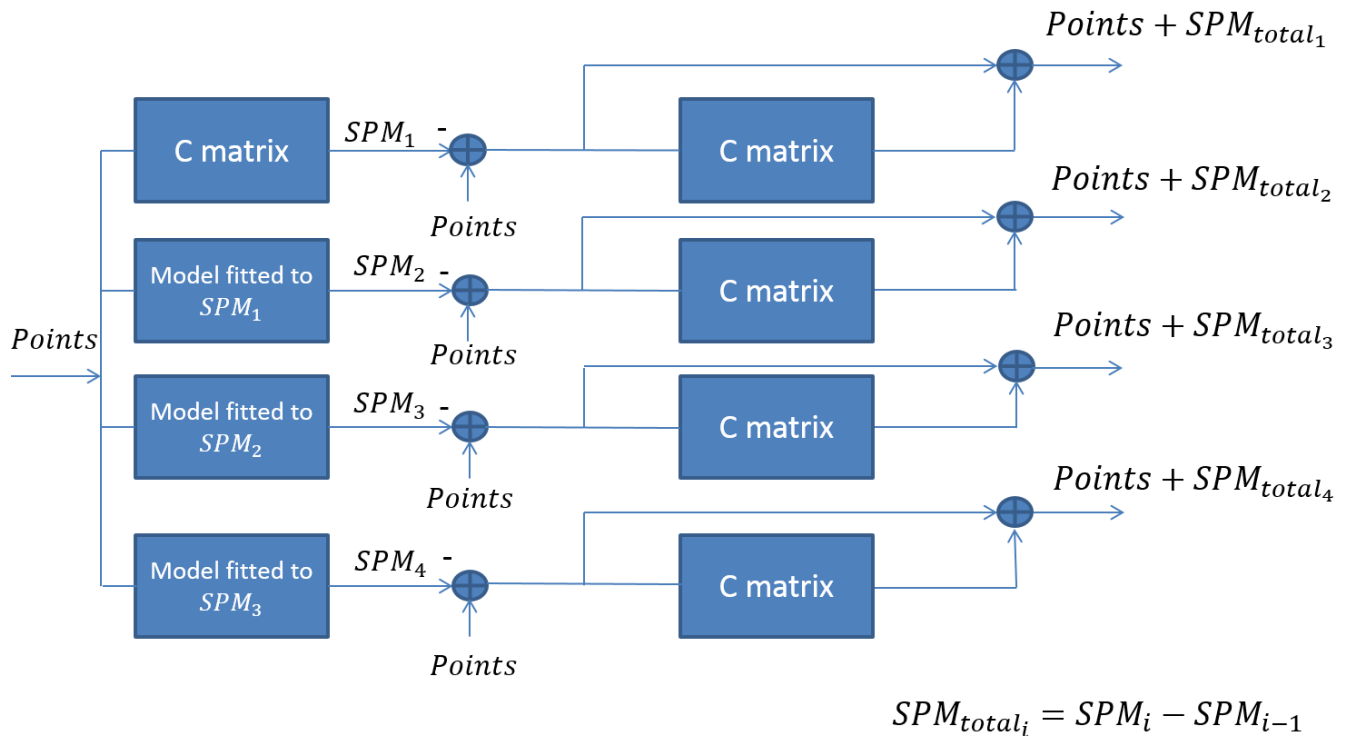


Figure 2.26: 4 examined cases

For NDSF fiber link the model's parameters for the fourth path can be seen in Table 2.15, and the power of SPM noise for different cases can be seen in Table 2.16. Figure 2.27 shows the structure of constellation points without doing compensation and with doing that (paths 1 and 2). By looking at Figure 2.28 we can compare constellation points obtained in paths 1 and 4 with constellation points obtained without doing compensation. Results obtained for ELEAF fiber can be seen in Tables 2.17 and 2.17. Constellation points plots for this type of fiber link are shown in Figures 2.29 and 2.30. As it can be seen, the fitted model can perform i-time compensation with low computational complexity, but we can not achieve compensation more than 3-time with the model which is because of the simplicity of the model.

Parameter	Value
$Ch_0$ real part	9.970984304806163e-01
$Ch_0$ imaginary part	2.027854168292119e-02
$Ch_1$ real part	-1.531630621808465e-05
$Ch_1$ imaginary part	2.348041896811138e-05
$(C_{0,0})_+$ real part	3.830138285315559e-23
$(C_{0,0})_+$ imaginary part	9.953298582455103e-04
$(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ imaginary part	6.098774512500139e-04
$(C_{0,-1})_+$ real part	1.915373992334389e-23
$(C_{-1,0})_+$ real part	-1.941597423336135e-23
$(C_{0,1})_+$ real part	3.253107621663822e-23
$(C_{1,0})_+$ real part	-8.244731903003282e-24

Table 2.15: Model's Parameters for the fourth path (NDSF fiber)

	The variance of SPM noise
Uncompensated	3.1667e-2
Compensated 1-time by C matrix	2.61e-3
Compensated by model fitted to 1-time compensated SPMs	4.73e-3
Compensated by model fitted to 2-time compensated SPMs	1.51e-3
Compensated by model fitted to 3-time compensated SPMs	5.4173e-4

Table 2.16: The variance of SPM noise for different cases (NDSF fiber)

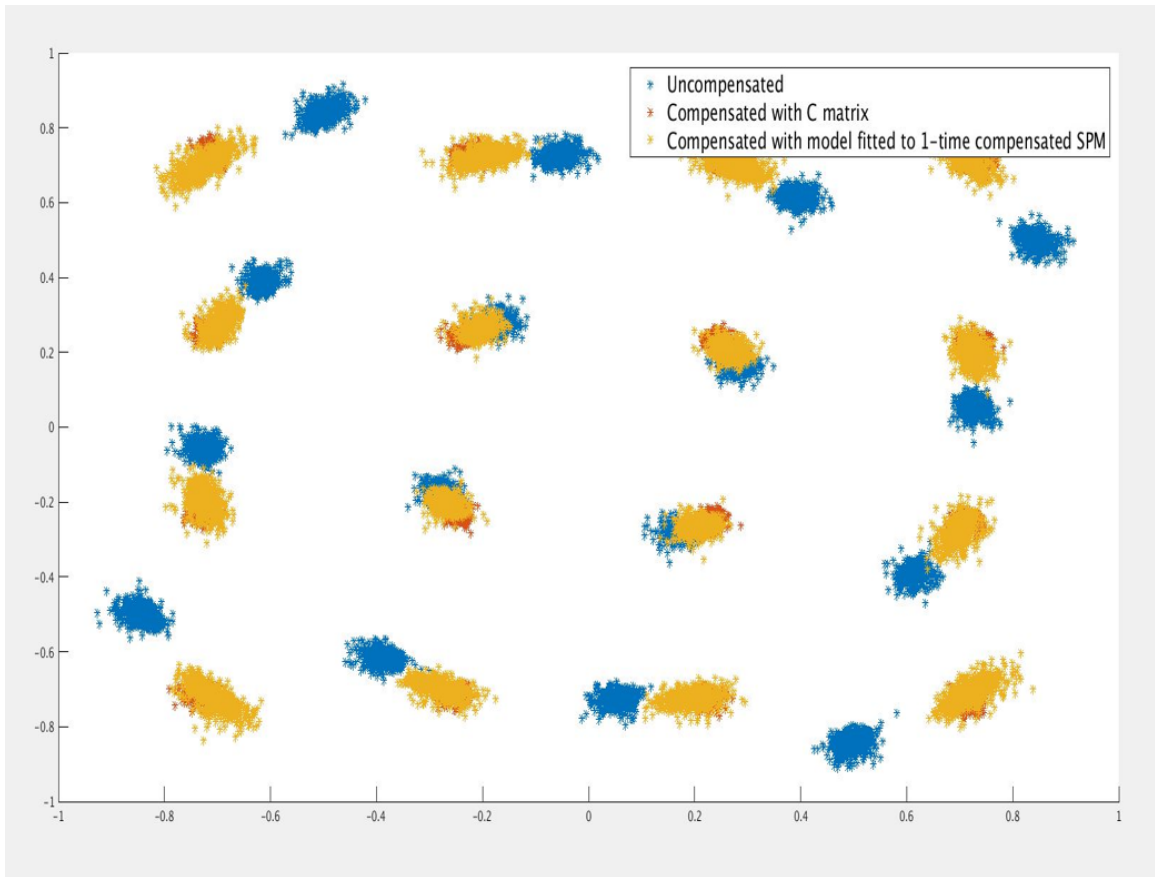


Figure 2.27: 16-Qam constellation points (uncompensated, compensated once by using C matrix, compensated once by using a fitted model. NDSF fiber link)

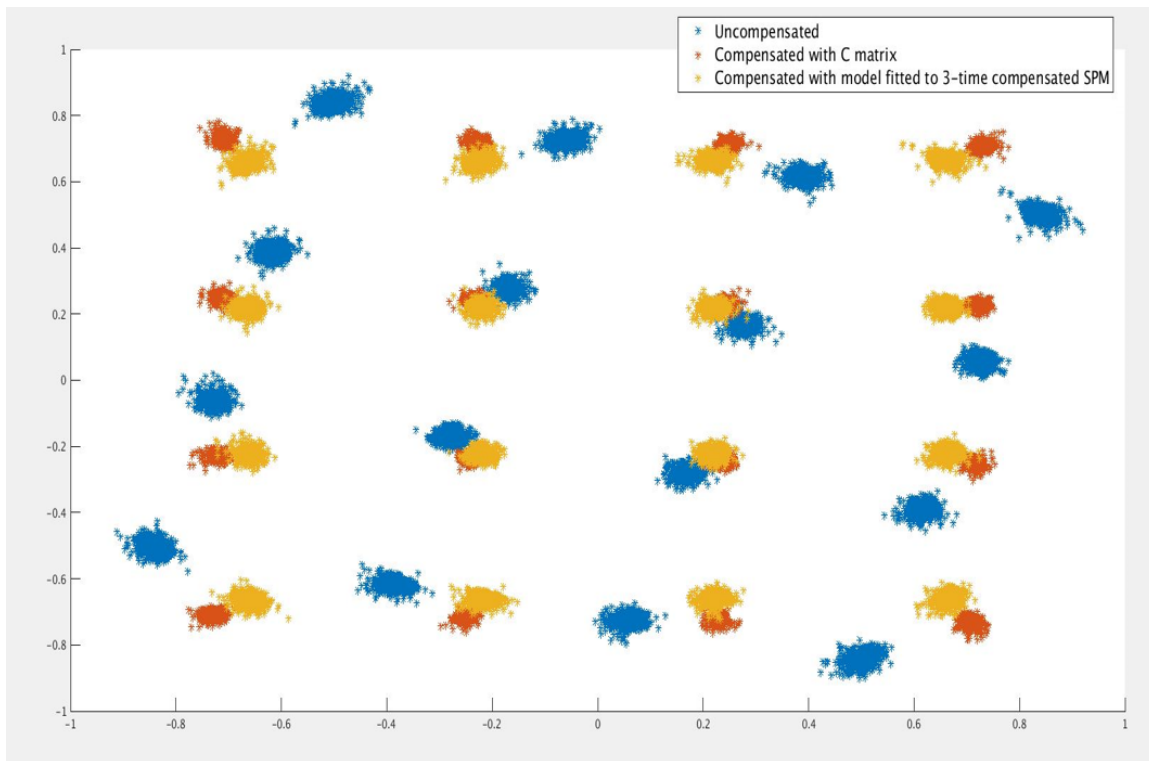


Figure 2.28: 16-Qam constellation points (uncompensated, compensated once by using C matrix, compensated with a model fitted to the values of 3-time compensated SPMs. NDSF fiber link)

Parameter	Value
$Ch_0$ real part	9.973768853101268e-01
$Ch_0$ imaginary part	7.077128349504908e-03
$Ch_1$ real part	1.545819758124695e-05
$Ch_1$ imaginary part	-2.690761489258046e-05
$(C_{0,0})_+$ real part	6.003514059600845e-23
$(C_{0,0})_+$ imaginary part	1.040727297197294e-02
$(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ imaginary part	3.911292579693129e-03
$(C_{0,-1})_+$ real part	-6.926465774724298e-24
$(C_{-1,0})_+$ real part	7.351525596689306e-23
$(C_{0,1})_+$ real part	3.218777070760761e-23
$(C_{1,0})_+$ real part	-1.196015896260585e-23

Table 2.17: Model's Parameters for the fourth path (ELEAF fiber)

	The variance of SPM noise
Uncompensated	2.76e-2
Compensated 1-time by C matrix	2.36e-3
Compensated by model fitted to 1-time compensated SPMs	2.7e-3
Compensated by model fitted to 2-time compensated SPMs	1.56e-3
Compensated by model fitted to 3-time compensated SPMs	6.48e-4

Table 2.18: The variance of SPM noise for different cases (ELEAF fiber)

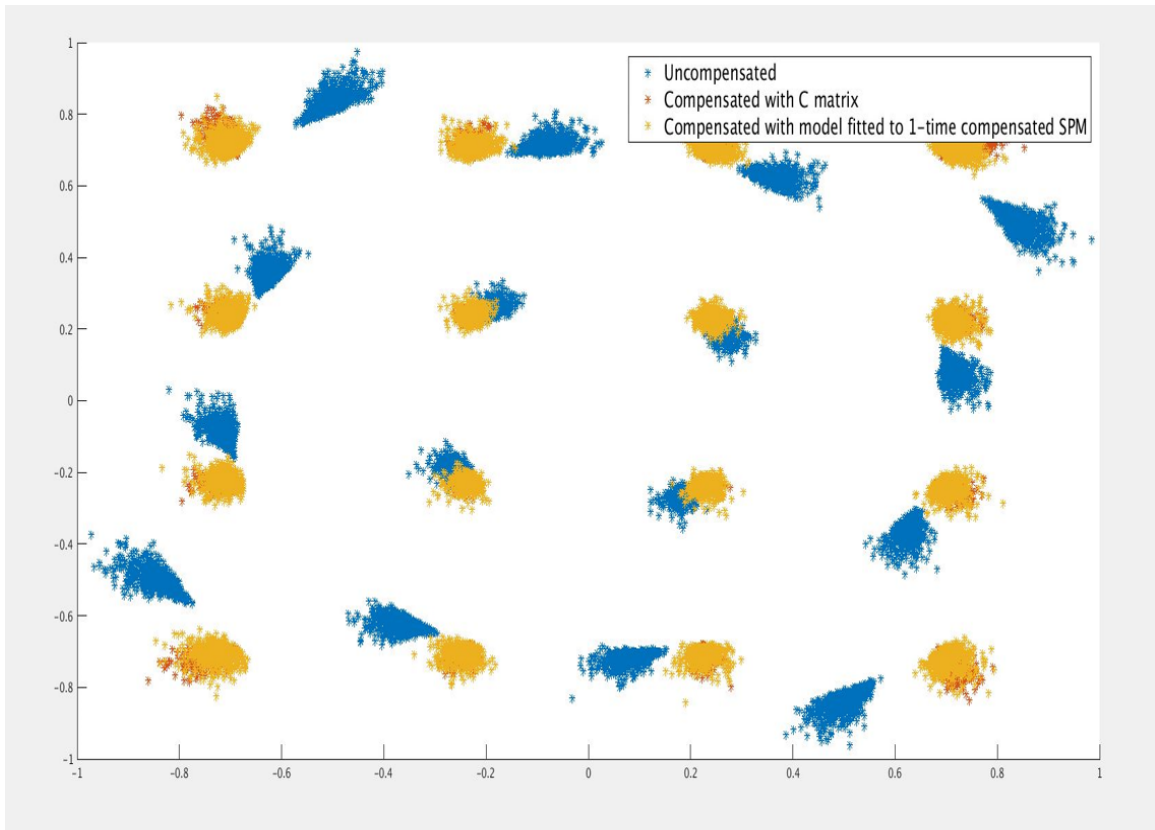


Figure 2.29: 16-Qam constellation points (uncompensated, compensated once by using C matrix, compensated once by using a fitted model. ELEAF fiber link)

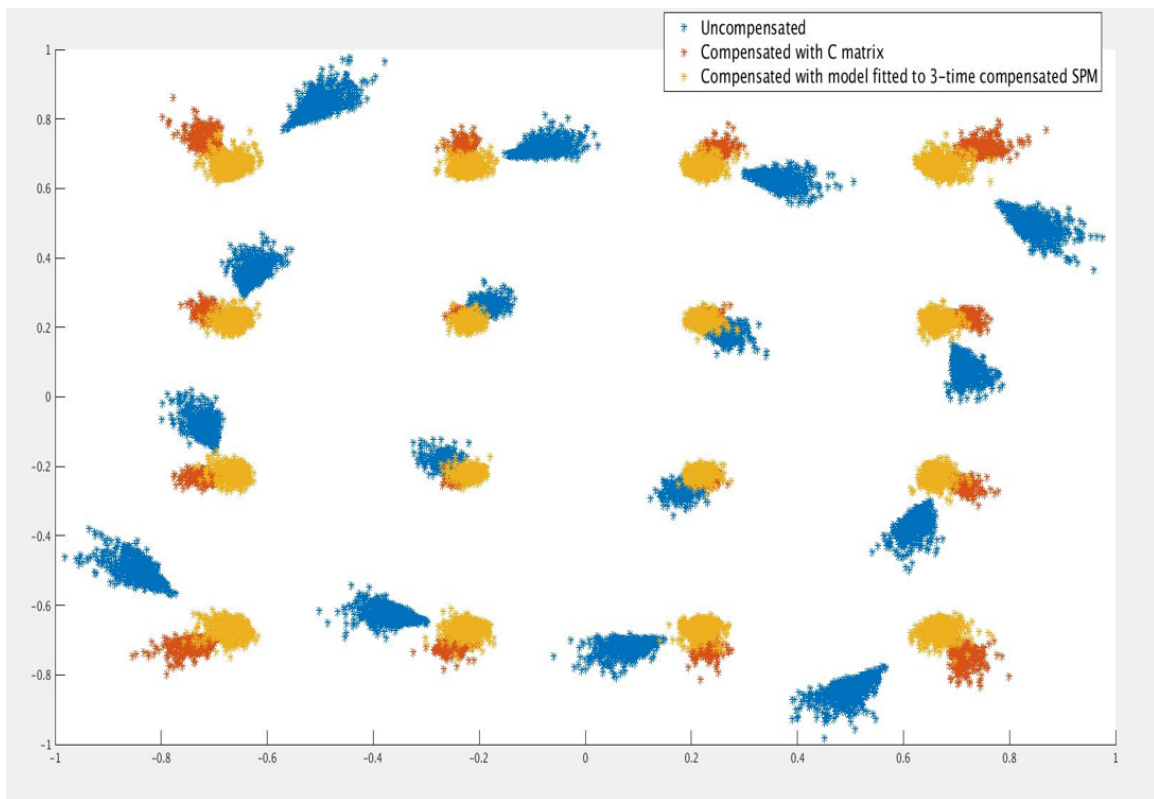


Figure 2.30: 16-Qam constellation points (uncompensated, compensated once by using C matrix, compensated with a model fitted to the values of 3-time compensated SPMs. ELEAF fiber link)

## 2.5 Results Obtained from Algebraic Fitting

In the previous section we discussed about one of the methods for finding the model's parameters called "Statistical Fitting". The disadvantage of that method is that we need to generate a huge number of data samples to achieve a valid model which fits to the statistical features of data samples. In the algebraic fitting method, we do not need to generate data samples. We only need to find the algebraic formula for the output of N-span model in terms of model's parameters and try to minimize the mean squared error between those terms and corresponding terms in Equation 2.1. In order to find the terms in algebraic formula for the output of the N-span model, we tracked the factors multiplying to data samples from one span to other. At the end, we only look at the 5907 valid terms, terms which have triple multiplication of data samples and minimize the mean squared error between those terms and the terms in Equation 2.1 which are known from C matrix. In this part, we use a more stable optimization method since we do not know a proper initial guess for model's parameters. The used optimization method is gradient descent with variable size steps. The step size reduces as the cost function reduces. The gradient method formula is based on starting from an initial guess for current parameters and updating them by subtracting updating factor ( $\alpha$ ) multiplied by gradient of cost function ( $\vec{G}$ ) from current parameters ( $\vec{P}$ ). The updating factor consists of two parts:

1. decaying factor called  $\alpha_{decay}$  which decays as the cost function reduces
2. fixed factor called  $\alpha_{fixed}$ . This factor introduced in BARZILAI, JONATHAN and BORWEIN, JONATHAN M's article "Two-Point Step Size Gradient Methods" [1]. It can be computed as follows:

$$\alpha_{fixed} = \frac{\langle \vec{E}, \vec{G} \rangle}{\|\vec{G}\|^2} \quad (2.27)$$

where  $\vec{E}$  is a vector obtained from perturbing parameteres and evaluating cost function for each perturbation.

The model's parameters obtained for NDSF fiber and the variance of the SPM noise can be seen in Tables 2.19 and 2.20 respectively. The constellation points can be seen in Figure 2.31 which shows that the fitted model can perform compensation same as the actual C matrix.



Parameter	Value
$Ch_0$ real part	9.92837305243569e-1
$Ch_0$ imaginary part	0.0209086758652834
$Ch_1$ real part	-9.34172266997008e-06
$Ch_1$ imaginary part	4.13364849049190e-06
$(C_{0,0})_+$ real part	9.18456578844837e-22
$(C_{0,0})_+$ imaginary part	0.00143577726059748
$(C_{0,1})_+, (C_{1,0})_+, (C_{-1,0})_+, (C_{0,-1})_+$ imaginary part	0.000694231386339778
$(C_{0,-1})_+$ real part	-7.46193308675579e-22
$(C_{-1,0})_+$ real part	2.71613777220239e-21
$(C_{0,1})_+$ real part	-6.35572587111291e-21
$(C_{1,0})_+$ real part	6.91505603010762e-22

Table 2.19: Model's Parameters obtained from algebraic fitting (NDSF fiber)

	The variance of SPM noise
Uncompensated	3.087e-2
Compensated 1-time by C matrix	2.48e-3
Compensated by using the fitted model	1.647e-3

Table 2.20: The varinace of SPM noise for different cases (NDSF fiber)

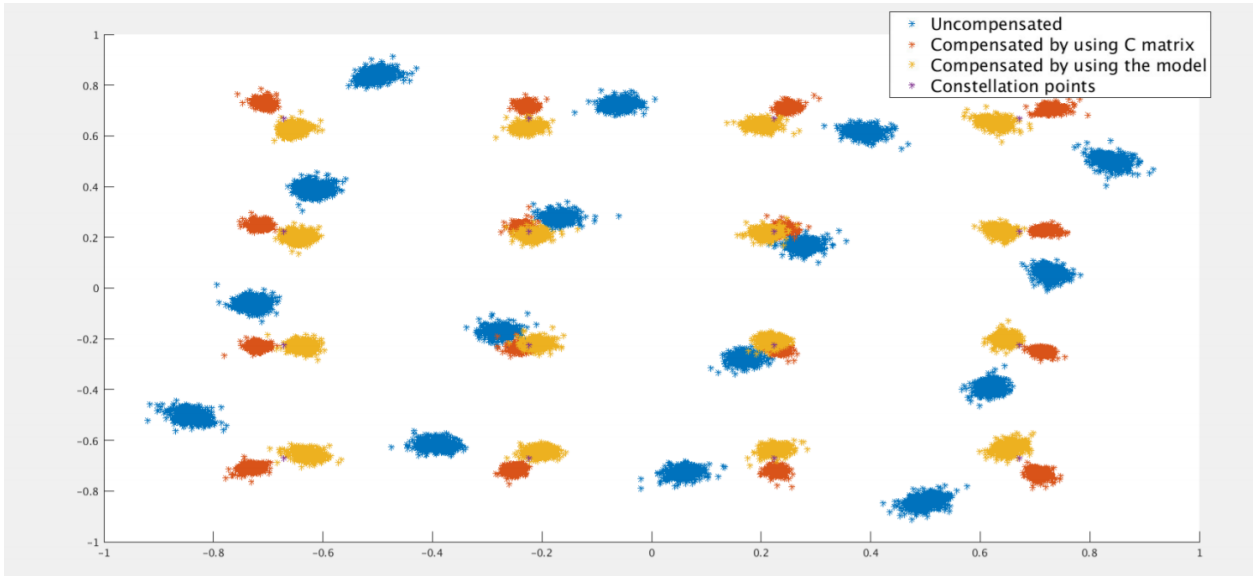


Figure 2.31: 16-Qam constellation points (Algebraic Fitting, NDSF fiber)

## 2.6 Summary

In this chapter, we introduced a simple N-span model which can help us in the compensation procedure. This model leads to similar results compared with C matrix model with an acceptable error and much less computational complexity. Computations needed for evaluating this model can be done in a parallel form. Two methods for finding the parameters of the model have been discussed. Statistical fitting has the ability of fitting to the result of  $i$ -time compensated data samples, but for performing this fitting method we need to generate a huge number of data samples. In contrast with statistical fitting, we do not need to generate data symbols in algebraic fitting. In this method we could not find the closed form for the compensated data, so finding a way for fitting the model algebraically to the compensated data samples can be considered as future work.

# Chapter 3

## Data Masking For SPM Noise Reduction

### 3.1 Introduction

In this chapter we will look at one of the ideas which can be used for reducing SPM noise through a fiber link. This method will be done in offline mode and before sending data through a fiber. First we divide data samples to some consecutive chunks. Then we will find a mask which leads to minimizing SPM noise for that chunk of data. At the end we put the index of used mask at the end of the chunk and send data through the fiber link. In the receiver, we will look at the part of data samples which shows the index of mask and will recover data.

In this chapter, first we will introduce the method and the structure of used masks. After that, we will look at the results obtained from this scheme. At the end, we will discuss about the trade-off in this scheme.

### 3.2 Data Masking

The idea is using  $N$  predefined pseudo-random masks with length  $L$  generated with independent seeds. After finding these masks, we can divide binary data to consecutive chunks with length  $L$ . Now we should mask each chunk with all masks and find the corresponding value of SPM noise for that partition of data. After doing that we will get  $N$  different

values of SPM noise for each partition. We choose the mask leading to the minimum value as the best mask for that part of data and put the index of the mask at the end of the partition. This idea helps since we are looking at  $N$  samples from a random variable, which is the SPM noise, and choosing the minimum of them. Doing that will reduce the power of that random variable which is the SPM noise in this case.

To illustrate, look at figure 3.1. In this figure, in sake of simplicity, it is considered that the SPM is only a function of three consecutive data samples. We only need three complex data samples, 12 bits of data, to compute SPM noise. By using four different masks, four different vectors in the three-dimensional space can be obtained. One of these vectors leads to a minimum SPM noise between all of four vectors. That mask will be considered as the mask for that chunk of data and the index of that mask will be specified with two bits at the end of that partition of data.

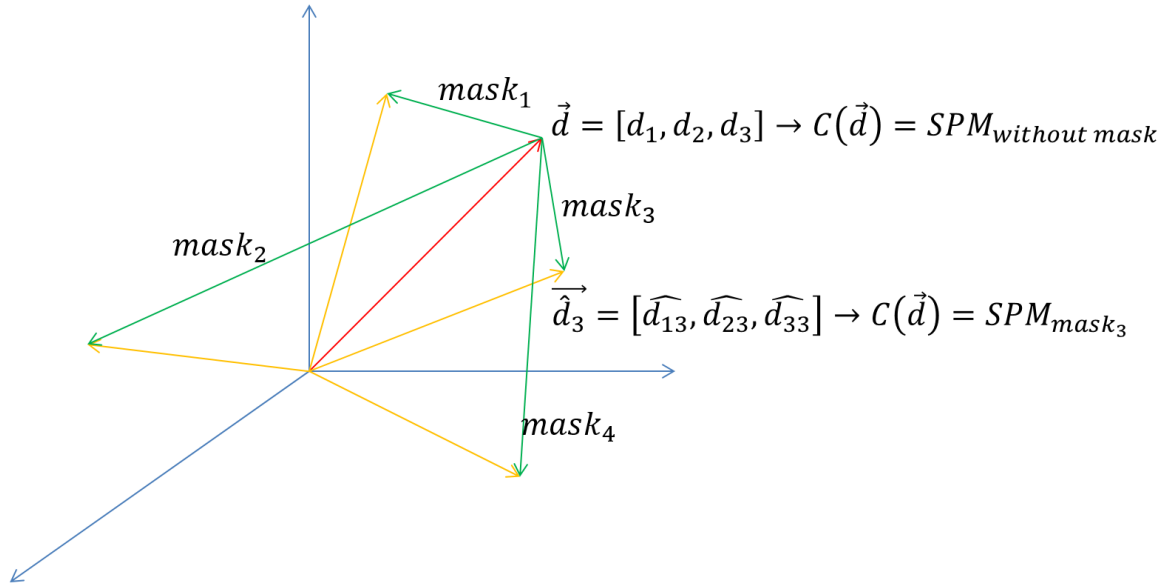


Figure 3.1: Sampling of SPM noise with masks

### 3.3 Masking Procedure

Suppose that the length of masks are  $L$  bits. We need first  $L$  bits of data to be ready before starting the procedure. After that, those  $L$  will be picked and XOR operation will be done between those bits and the bits of four known masks. Those  $L$  bits can be transformed to  $\frac{L}{4}$  data symbols in 16-QAM case. To Compute SPM noise for all of those  $\frac{L}{4}$  data symbols, depends on the dimension of C matrix we need to know some data symbols before the first symbol in the partition and after the last symbol. Suppose that the C matrix is a  $K \times K$  matrix. We need to know  $\frac{K-1}{2}$  data symbols generated before the first data symbol in the partition, and same number of data symbols which will be generated after the last symbol in the chunk of data. For the first iteration we do not have any data symbol before the first chunk, and we always do not know the data symbols which will be generated after that chunk. In that case we should consider the average of data symbols which is zero as data symbols. In other words, we should pad  $\frac{K-1}{2}$  zeros before that chunk of data and also same number of zeros after that for the first partition. For other partitions we have an access to previous data symbols, so we can use them. For next data symbols we always have to put zero instead of them. The mentioned zero padding and previous data using can be seen in Figure 3.2 . After padding zero and using data symbols stored in memory, know we can find SPM noise for each data symbol after masking data bits with different binary mask. Figure 3.3 shows the procedure of choosing proper mask. In that figure,  $\overline{SPM}_i$  is the average power of SPM noises corresponding to the i-th mask. That can be computed by using the following equation:

$$\overline{SPM}_i = \frac{\sum_{j=1}^n SPM_{ji} \times SPM_{ji}^*}{L} \quad (3.1)$$

where  $SPM_{ji}$  means the SPM noise corresponding the j-th data symbol after masking data bits with the i-th mask.

The used algorithm can be seen in Figure 3.4. It should be noted that, we mask the processed chunk with the best mask and after that we will go to process next mask. In other words, we store masked data bits in memory and use them for processing the next partitions of data not the unmasked data symbols.

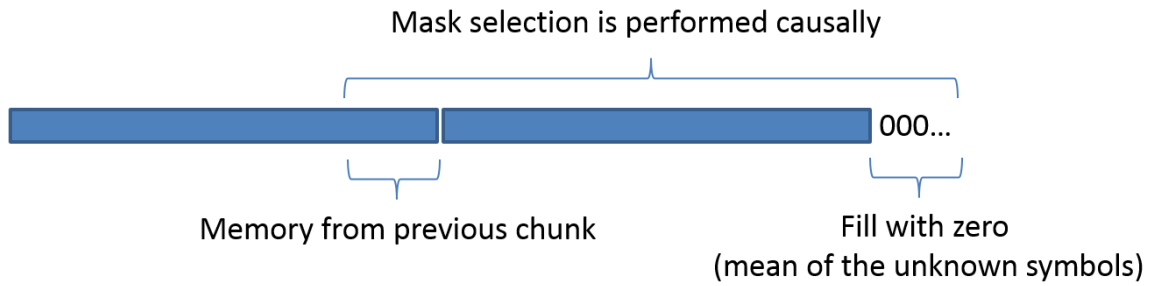


Figure 3.2: We need previous and next data symbols to compute SPM

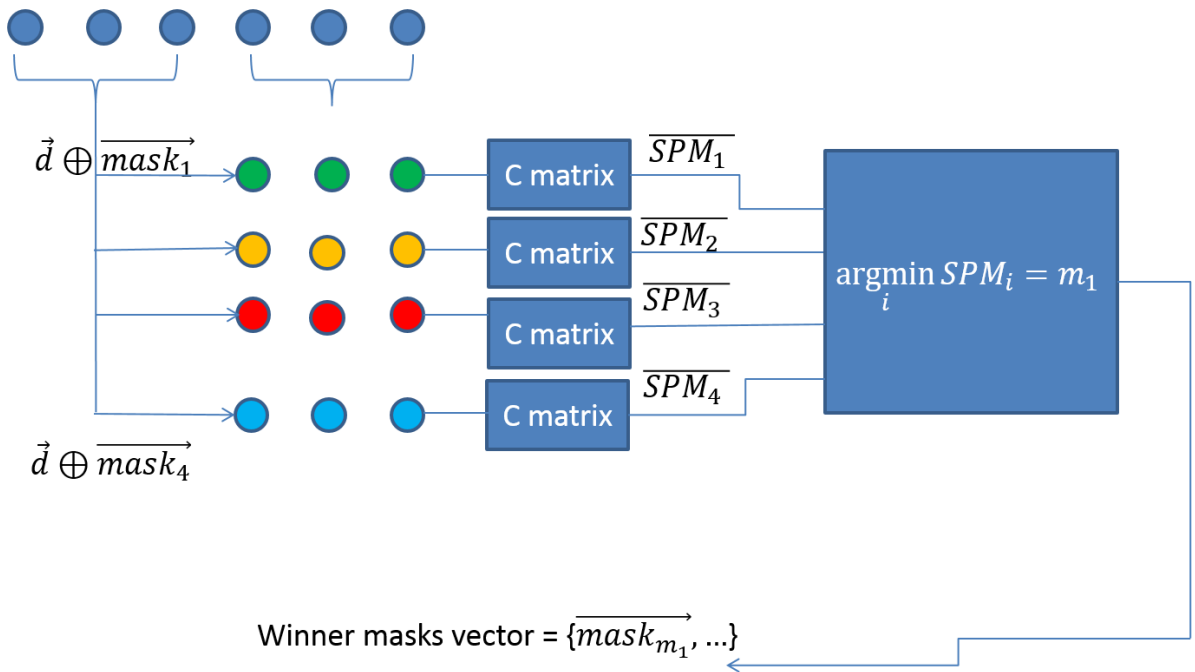


Figure 3.3: The procedure for choosing masks

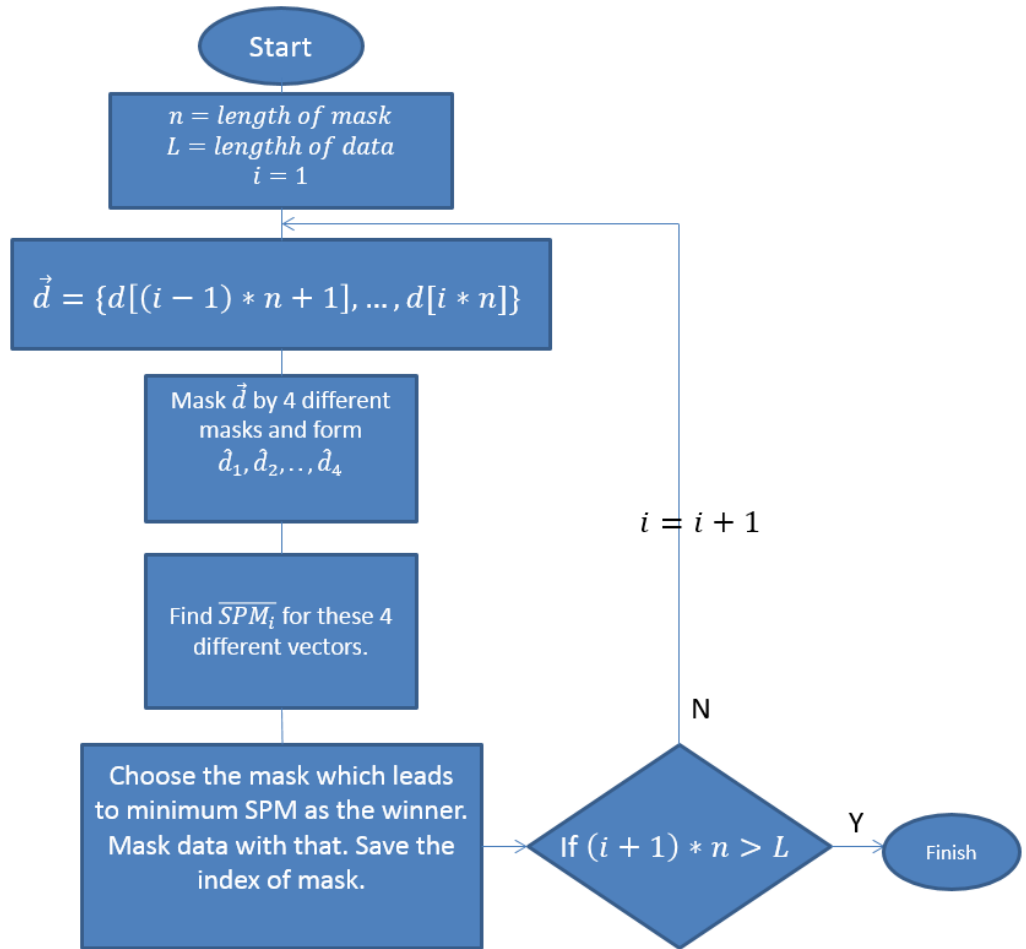


Figure 3.4: Masking Algorithm

### 3.4 Results

The masking procedure have been done for 25-span NDSF fiber link, 12800 data symbols, and  $12800 \times 4$  bits. Masks with two different lengths have been used. The results obtained for the case in which the lengths of masks are  $100 \times 4$  bits can be seen in Table 3.1. For the case in which the lengths of masks are 50 bits, the result can be seen in Table 3.2.

	$E\{SPM \times SPM^*\}$
Without Masking	16.23
With Masking	13.043

Table 3.1: Mask with length 400 bits

	$E\{SPM \times SPM^*\}$
Without Masking	16.13
With Masking	11.283

Table 3.2: Mask with length 200 bits



As it can be seen in first case, with longer mask, the reduction is around 18.75 percent, and for the second case the reduction is 31.25 percent. It shows that as much as the length of the masks decreases, we will get more reduction in the variance of the SPM noise. Although reducing the length of masks will help us with SPM noise reduction, we need to put more indexes for masks within data sample bits. It means that the rate will be less than the case we use longer masks and also the offline computations will be more. So there is a trade-off between the length of masks and the SPM noise reduction. The MATLAB code used for masking data symbols and computing results, can be seen in Appendix F.

### **3.5 Summary**

In this chapter, we have looked at a method for SPM noise reduction which is based on the sampling of SPM noise and choosing the mask leading to the minimum SPM noise. Doing this procedure leads to less SPM noise in expense of data transmission rate.

# Chapter 4

## Modified Tree-structure Code for SPM Noise Reduction

### 4.1 Introduction

In this chapter, we will look at the modified version of tree-structure codes. In this chapter, first we will look briefly at tree-structure codes and the concept of shaping. Then we will introduce a modified version for addressing constellation points in a tree-structure code scheme. At the end, we will look at the results obtained from this modified version in different cases.

### 4.2 Shaping and Tree-structure codes

It has been shown in the first chapter, we can use formulas [1.20](#) to [1.32](#) for computing the variance of SPM noise. Although this formula has been obtained by assuming some simplifying conditions, it is a good metric for computing a range for the values of SPM noise. As it can be seen in the mentioned formula, there is a term which corresponds to the average energy of constellation points. It means that if we want to reduce the SPM noise value, we can reduce the average energy of constellation points in a way in which other moments will not increase at all or will not increase too much to cancel the effect of second moment reduction.

Shaping constellation points is one of the methods for reducing the average energy of constellation points. In simple words, shaping is choosing  $M$  points as constellation points

between all possible points in an N-dimensional space which lead to minimum average energy. The idea is truncating a cube in N-dimensional space. In Wei's article "Trellis-coded modulation with multidimensional constellations" [18], it is mentioned that shaping is the result of the method by which we can transmit a non-integral number of bits per two dimensions. This shaping method is generalized in Forney and Wei's article "Multidimensional constellations. I. Introduction, figures of merit, and generalized cross constellations" [7], and more discussed in Forney's article "Multidimensional constellations. II. Voronoi constellations" [8]. Other significant work in this area has been done in the article of Calderbank and Ozarow called "Non-equiprobable signaling on the Gaussian channel" [2]. This article a shaping method has been introduced which is done directly in 2-D subconstellation by partitioning that subspace into equal sized subregions of increasing average energy. One of the challenges in shaping is addressing problem. It means that after choosing M points in a way which leads to minimum average energy, we should find a mapping method to map data bits to those M chosen data symbols.

The addressing method used in this chapter is introduced in the work of Khandani, AK and Kabal, P [10]. To do the discussed addressing procedure we start from 2 - dimensional space with a base constellation structure, in this case 16-QAM. After that we consider 16 constellation points as 4 different energy shells. We consider those shells as portions and consider different concatenations of these shells which leads to 16 new shells, called partitions. Now we sort these 16 partitions and find all concatenations of them. We need to know what data transmitting rate we need at the end to find out when we should stop the concatenation procedure. Also, we need to know in each step of concatenation how many partitions with same size we need to save in look up table. This addressing method has four phases:

1. Concatenation: concatenation procedure in N-dimensional space means finding all partitions by concatenating partitions remaining from  $\frac{N}{2}$ -dimensional space.
2. Sorting: after finding all partitions in N-dimensional space, we should sort those partitions by a metric. Different metrics and sorting methods can be used in this stage. We will look at this phase in more details later in this chapter when we introduce a sorting method for SPM noise reduction.
3. Clipping: in this stage we choose K partitions from all partitions in this N-dimensional space. Because clipping stage will be done after sorting, only partitions with lower values of the metric will be remained.
4. Merging: For doing concatenation procedure for the higher dimensional space we need to have partitions in current space. We used uniform addressing which means that

the number of links, partitions from lower dimensional space, in each new partition is same. This number is specified by desired transmission rate or in other words optimum probability distribution for data symbols.

## 4.3 Modified Version of Tree-structure Codes

As we discussed in the previous section, the addressing procedure in tree-structure codes has 4 phases. The modification has been done in second phase, sorting. In basic tree-structure codes, the sorting stage will be done by sorting partitions based on the average energy of them. This sorting method leads to data symbols for which the average energy is the smallest possible value for a known transmission rate. This method will not lead to the minimum SPM value. We tried to find sorting method leading to lower SPM noise variance with approximately same average energy.

To achieve lower average SPM noise value, the sorting stage divided into 2 phases:

1. Phase1: for 2, 4, and 8-dimensional spaces, sorting will be done based on the value obtained from passing energy vectors, will be defined later on, through the C matrix.
2. Phase2: for spaces with dimension higher than 8, a metric will be defined for each partition which is a combination of SPM values corresponding to partitions from previous space which build that partition.

These two phases will be discussed in more details in the following subsections.

### 4.3.1 Sorting Method Used in 2, 4, and 8-dimensional Spaces

In this phase for each link we have a vector which shows energy indexes. For example in 4-dimensional space we have an energy vector as follows: 1, 2, 3, 2. It means that this link consists of one complex number with energy index 1, and 3 and 2 complex numbers with energy index 2. As we discussed before, we divided 16-QAM constellation points into 4 groups which contains complex numbers which have same real parts and same imaginary parts with different signs. For example these constellation points will have a same energy index:  $\alpha + \beta i, -\alpha + \beta i, \alpha - \beta i, -\alpha - \beta i$ . By considering this method for indexing energy indexes, we will have  $4^4 = 16$  possible vectors of complex numbers for mentioned energy vector 1, 2, 3, 2. To find a corresponding SPM value to this vector, we pass these 16 vectors through C matrix. Since C matrix needs a memory higher than 4 data symbols, we put

zero at the beginning and the end of these 16 vectors. After finding SPM values for all of those 16 vectors, we find the average of absolute value to the power of two of those SPMs and consider that as SPM for that energy vector. Now we can use these values in the sorting phase. For the 8-dimensional space this procedure will be done slightly different. In that space, the number of possible vectors for each energy vector is  $4^8 = 65536$ . To reduce the computational cost, we only chose one tenth of those vectors randomly and find the SPM value for each energy vector. The MATLAB code used for finding the average SPM noise for energy vectors can be seen in Appendix H.

After finding SPM value for each energy vector we will sort them based on those values, keep a specified number of them, and merge them into partitions with same number of links inside. Now we will average SPMs in each partition and consider the result as SPM value for that partition for next space. If some energy vectors have same SPM value, we will sort them based on energy. It is called **emphlexicographical ordering**. It means that we order items based on one metric at first, called  $Metric_1$ . Then, we will do ordering based on another metric,  $Metric_2$ , between all the items with a same  $Metric_1$ .

### 4.3.2 Sorting Method Used in Spaces with Dimension Higher than 8

For each link which has been made by concatenation of two partitions from lower dimensional space, we find an SPM value by doing some combination on the SPM values of building partitions. We considered three different cases:

1. Continue sorting based on energy instead of using SPM values.
2. Define new metric which can be computes as follows:

$$SPM_{ij} = SPM_i + SPM_j \quad (4.1)$$

where  $SPM_{ij}$  is the SPM value for the link built by concatenation of partitions  $i$  and  $j$  from the previous lower dimensional space.  $SPM_i$  and  $SPM_j$  are the SPM values corresponding to partitions  $i$  and  $j$  in the previous lower dimensional space.

3. Define new metric which can be computes as follows:

$$SPM_{ij} = SPM_i^2 + SPM_j^2 + \alpha \times SPM_i \times SPM_j \quad (4.2)$$

this metric absorbs the interaction between SPM values of partitions after concatenation in the multiplication term. We used  $\alpha = 8 - \log_2(\text{dimension})$ . It means that the interaction between partitions will decay as long as we continue concatenation.

## 4.4 Results

Results obtained for mentioned cases can be seen in Table 4.1. This results obtained for 25-span NDSF fiber. Concatenations procedure stopped in 128-dimensional space. The MATLAB code used for implementing this modified version of tree-structure codes, can be seen in Appendix G.

Method	$E\{SPM \times SPM^*\}$	Average Energy
Sorting based on energy	3.01e-3	0.283
After 8-dimensional space sorting based on energy	2.87e-3	0.283
After 8-dimensional space sorting based on: $SPM_{ij} = SPM_i + SPM_j$	2.75e-3	0.288
After 8-dimensional space sorting based on: $SPM_{ij} = SPM_i^2 + SPM_j^2 +$ $(8 - \log_2(dimension)) \times SPM_i \times SPM_j$	2.68e-3	0.288

Table 4.1: Different Sorting Methods

## 4.5 Summary

As it has been discussed through this chapter, we can improve the performance of tree-structure codes by changing the sorting method used in building up the look up table for addressing. Changing the sorting method will increase the final average energy, but will help in SPM noise reduction. It means that the result is not optimum in sense of the goal of shaping. We used three different methods for the second phase of sorting. The best method was the one in which we considered the interaction between two partitions after concatenation, and modeled that by multiplication. For that case we achieved 0.49 dB reduction in SPM noise.

# Chapter 5

## Joint Detection for Exploiting The Memory of XPM noise

### 5.1 Introduction

As we discussed before, there is an intrinsic memory inside XPM and SPM noises. In the previous chapters we introduced methods for reducing SPM noise. For reducing the effect of XPM noise, we can not use mentioned methods; In all of those methods, we need to have an access to data symbols from other channels to reduce XPM noise made by them which is not applicable in most of the cases. In this chapter, we are going to introduce a method for reducing the effect of XPM noise on bit error rate and symbol error rate without manipulating data symbols of other channels. In this method, we need to know the statistical characteristics of data symbols on other channels, which can be obtained easily.

In this chapter, first we will take a look into statistical characteristics of XPM noise. Then we will introduce a detection method which can exploit those characteristics of XPM noise in order to perform better detection compared with basic minimum distance detection.

### 5.2 The statistical Characteristics of XPM Noise

In the current systems, minimum distance detection method is used. It will be shown that this detection method can be modified to a more sophisticate one by considering the

statistical features of XPM noise. The effect of XPM noise on the sent constellation points can be seen in Figure 5.1, 16-QAM and the average energy of constellation points is 0.5. It is obvious from this plot that the shape of clouds generated because of XPM noise around the constellation points is not circle. In other words, detection based on the minimum distance is not the best method for detection in this case. To find a better way for detection, first we should find a proper approximation for the probability density functions corresponding to the different clouds of XPM noise.

The XPM noise affects constellation points conditional on the sent data. In other words, the XPM noise adds a mean to the constellation point and generates an elliptical cloud around the new constellation point which is the original one plus the mean. The added mean and the shape of cloud are conditional on the sent constellation points. This fact can help us with the detection phase. Also, there is a correlation in time domain between the samples of XPM noise which comes from the intrinsic memory of nonlinear noise which can be seen the equations used for computing XPM noise values. In Figures 5.2 and 5.3 this correlation can be seen. The range of XPM noise is divided into 10 partitions, based on the absolute value of XPM noise to the power of two. This figure shows that if the current XPM noise sample is in the partitions with low energy, the next sample will be in the partitions with low energy with a high probability. Also, if the current sample has a high energy, it is high probable for the next sample to have a high energy. In other words there is intrinsic information in the time domain which can be used in the detection phase.

Another important characteristic of XPM noise samples is the correlation between the real and imaginary components of each sample conditional on the constellation points. As it can be seen in Figure 5.1, the cloud of noise for each constellation point has a unique shape. For one of the constellation points with the highest energy, we have found the principal component from eigenvalue decomposition. The principal directions for the cloud of noise corresponding to that constellation point can be seen in Figure 5.4. The eigenvector corresponding the maximum eigenvalue is  $-7.12e - 01 - 7.02e - 01i$  which means that the angle of the principal basis is around 45 degrees. In other words, there is intrinsic information inside the joint probability density function of real and imaginary parts of XPM samples which can help us in detection.



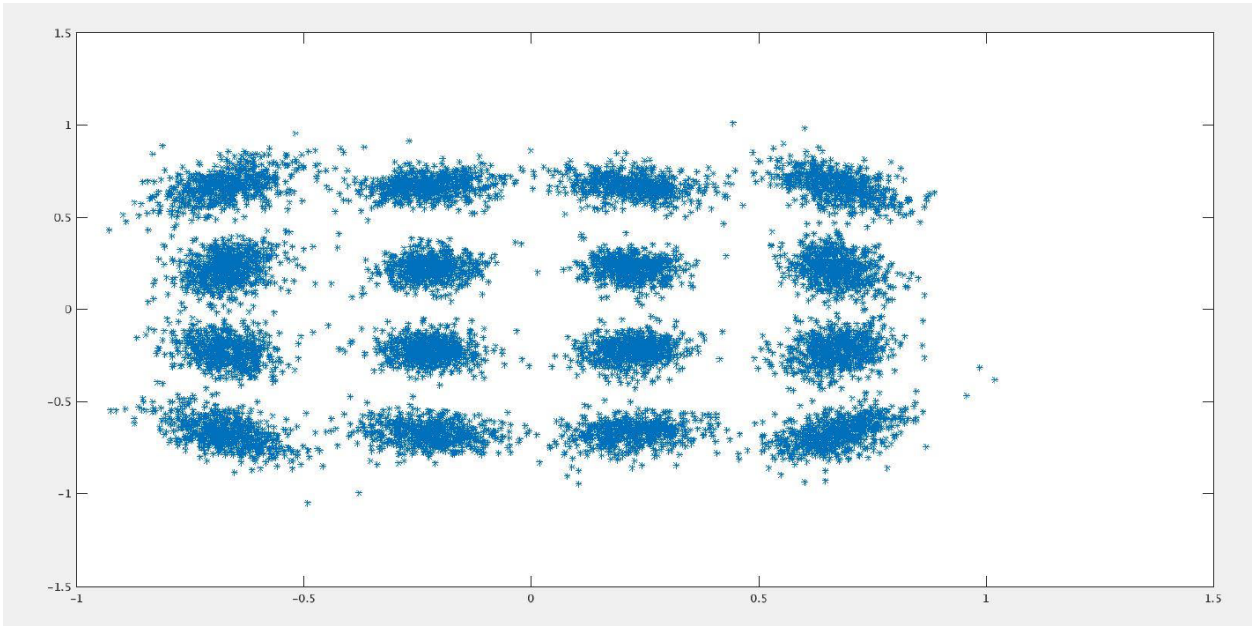


Figure 5.1: The effect of XPM noise on constellation points

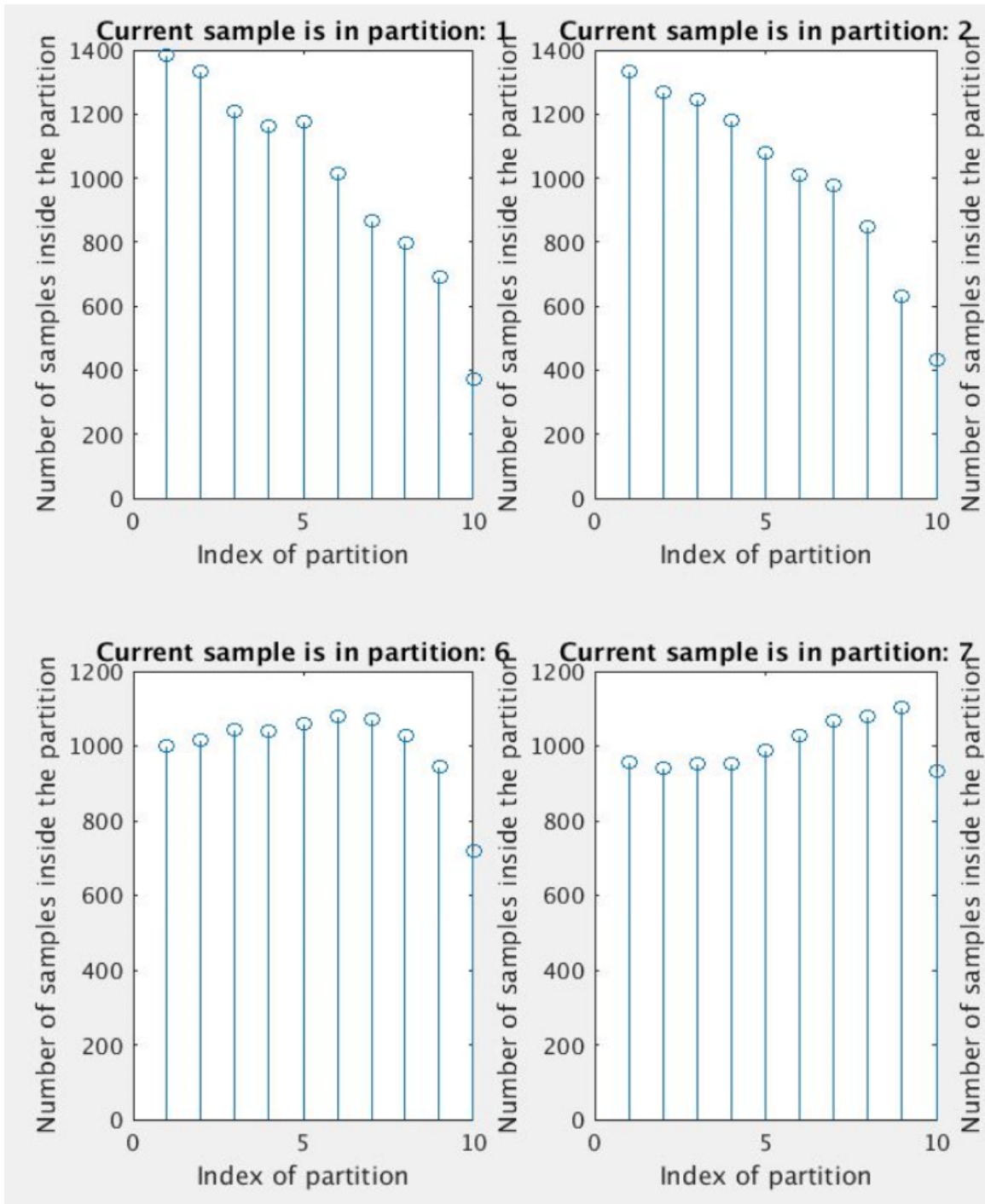


Figure 5.2: The probability of the energy of next sample conditional on the current sample

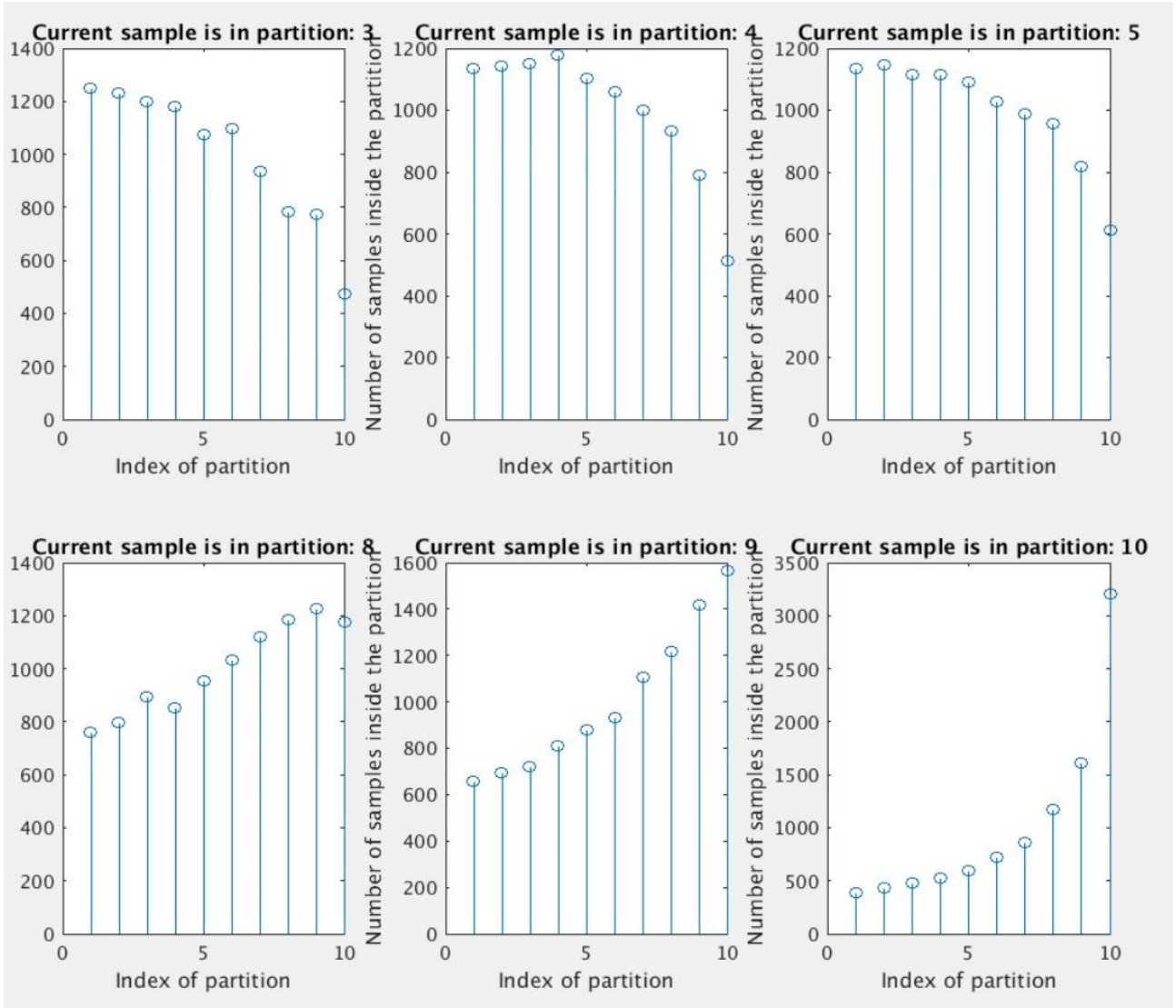


Figure 5.3: The probability of the energy of next sample conditional on the current sample

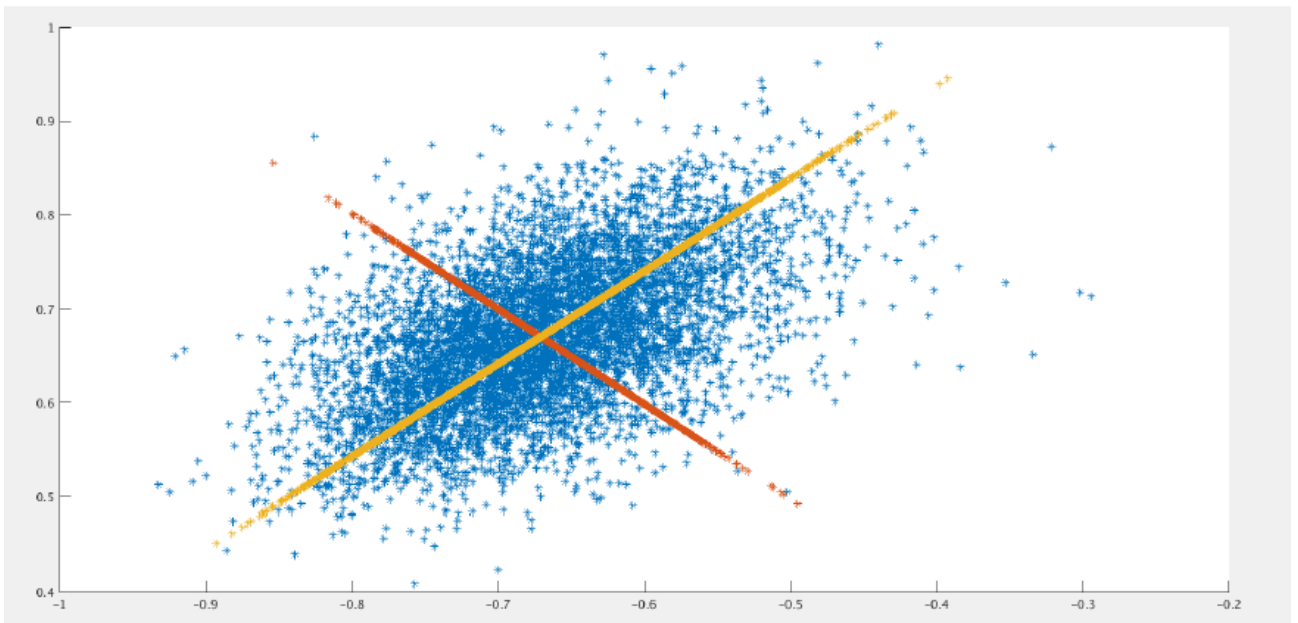
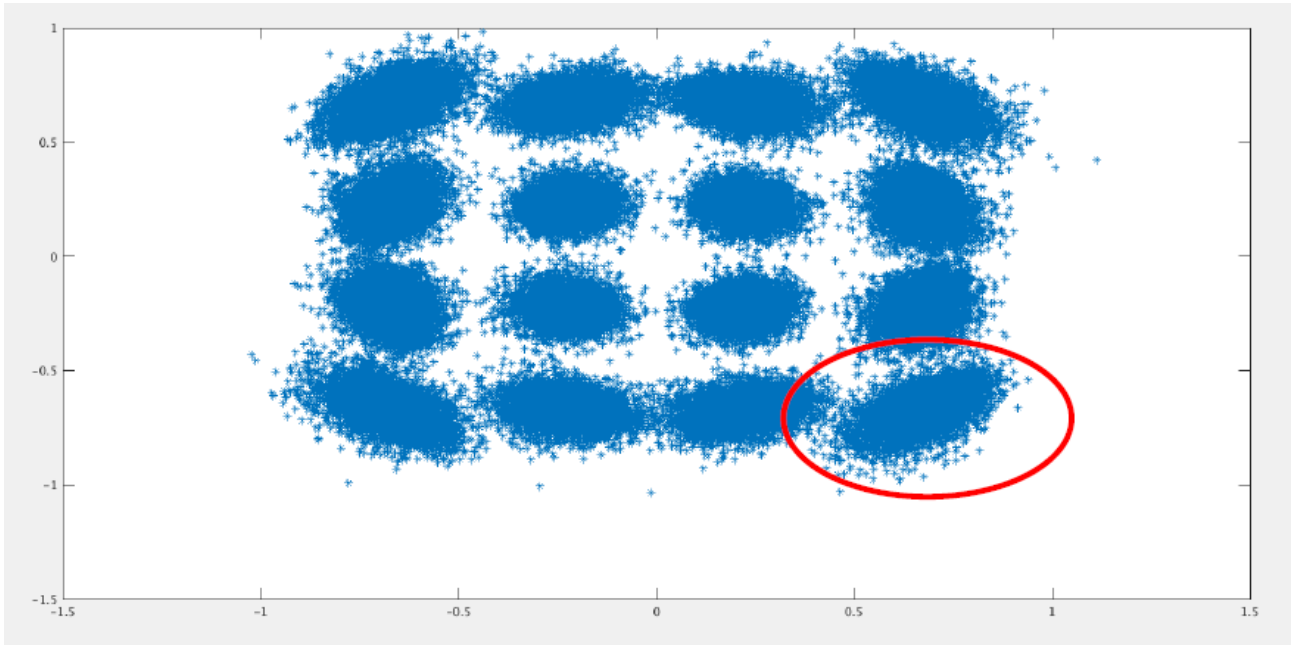


Figure 5.4: The principal component, Yellow line, for the XPM noise corresponding constellation point  $0.6708 - 0.6708i$

## 5.3 Joint Detection Scheme

As we discussed in the previous section, there are two main characteristics for XPM noise samples:

1. Correlation between consecutive XPM noise samples in time.
2. Correlation between the real and imaginary parts of an XPM noise sample.

To exploit these features, in the detection phase we will consider three consecutive received data symbols instead of looking only at the current symbol. If we do detection based on three consecutive symbols, we have to consider  $16^3 = 4096$  possible outcomes. If we consider these three complex numbers as 6 real numbers which are jointly Gaussian, we have to find joint probability density function for these six random variables conditional on all 4096 possible vectors. After finding covariance matrices and average XPM noise vector for all of these possible vectors, we simply put received vector into equation 5.1 and will choose the candidate with the highest conditional probability as the sent vector.

$$f(\vec{X}|\vec{S}) = \frac{\exp(-\frac{1}{2}(\vec{X} - \vec{\mu})^T \Sigma^{-1}(\vec{X} - \vec{\mu}))}{\sqrt{(2\pi)^6 |\Sigma|}} \quad (5.1)$$

where  $\vec{X}$  is the received vector,  $\vec{S}$  is one of the 4096 candidates,  $\vec{\mu}$  is the expected XPM noise vector for the candidate, and  $\Sigma$  is the covariance matrix which must be calculated for each candidate vector.

## 5.4 Results

Two steps should be done for joint detection. First step is finding covariance matrices and expected vectors for each candidate from 4096 possible ones. To do this step, first we generate one million data symbols and after that XPM noise for them. Then we searched for candidate vectors among these symbols and found expected value of XPM noise affecting them and the covariance matrix for them. The last step is doing joint detection by using the obtained probability density functions. To do this simulation XPM noise and also white Gaussian noise has been added to the 100000 generated data symbols. The considered fiber link is 25-span NDSF. As it can be seen in Figure 5.5, for low SNRs this method

does not help much because the effect of white noise which is memoryless is dominant and doing joint detection will not cause any improvement. In high SNRs, using joint detection, in other words exploiting the memory of XPM noise, will reduce the bit error rate. The MATLAB codes written for implementing different steps of joint detection scheme, can be seen in Appendix I.

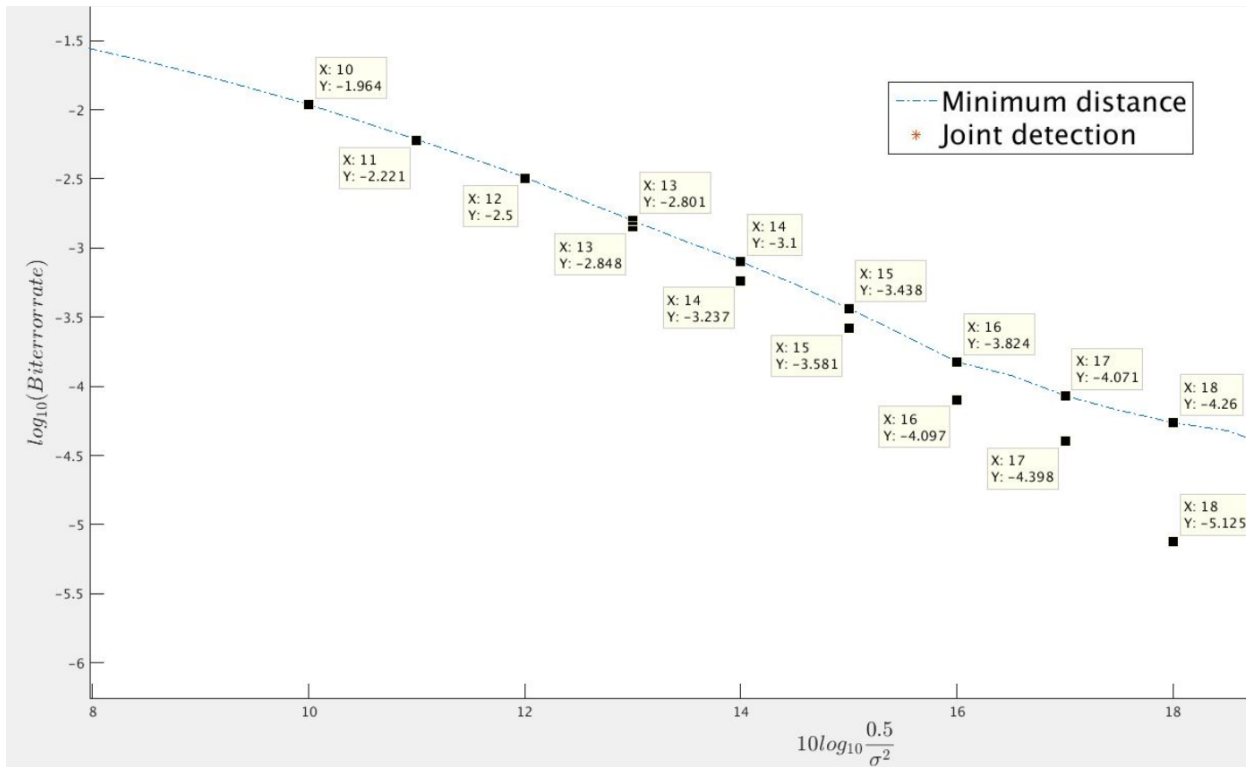


Figure 5.5: Comparison of bit error rate plots for two detection methods

## 5.5 Conclusion

The memory caused by XPM noise can be exploited in the detection phase. Using this detection method may increase computational complexity in the detection phase. For low SNR case the joint detection does not help since the Gaussian noise is memoryless, and joint detection is a way for exploiting existing memory inside XPM which cannot be exploited because Gaussian noise is dominant for low SNRs.

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# APPENDICES

# Appendix A

## Matlab Code for Simplex Method

```
clear all
clc
close all
global H
global C
global indin
string = sprintf('UWCmatrix%dSpanNDSF',9+1);
load(string,'C')
load('Hicsizinja')

poi1 =0.2236 ;
poi2 = 0.6708;
Cnst=[ poi1+poi1*i  -poi1+poi1*i  poi1-poi1*i  -poi1-poi1*i...
      poi1+poi2*i  -poi1+poi2*i,...
      poi1-poi2*i  -poi1-poi2*i  poi2+poi1*i  -poi2+poi1*i...
      poi2-poi1*i  -poi2-poi1*i,...
      poi2+poi2*i  -poi2+poi2*i  poi2-poi2*i  -poi2-poi2*i];
x=[1
2e-02
1e-06
1e-06
1e-22
1e-03
10e-04
```

```

1e-22
1e-22
1e-22
1e-22];
C(find(isnan(abs(C))))=0;
C =C;
evals = errorSPM(x);
%%
for j=1:99
    j
    [evalsorted ja] = sort(evals,'ascend');
    x = x(ja,:);
    xnew = x(1:end-1,:);
    cent = mean(xnew,1);
    evals = evalsorted;
    %Ref
    xr = cent + alpha*(cent-x(end,:));
    fxr = errorSPM(xr);
    if fxr>=evalsorted(1)&&fxr<evalsorted(end-1)
        x(end,:)=xr;
        evals(end)=fxr;
        continue
    end
    % Expa
    if fxr<evalsorted(1)
        xe = cent + gama*(xr-cent);
        fxe = errorSPM(xe);
        if fxe<fxr
            x(end,:)=xe;
            evals(end)=fxe;
        else
            x(end,:)=xr;
            evals(end)=fxr;
        end
        continue
    end
    % contra
    if fxr>=evalsorted(end-1)

```

```

    xc = cent + rou*(x(end,:)-cent);
    fxc =errorSPM(xc);
    if fxc<evalsorted(end)
        x(end,:)=xc;
        evals(end)=fxc;
        continue
    end

end
%Shrink
x(2:end,:) = ones(size(x,1)-1,1)*x(1,:)+sigma*(x(2:end,:)...
        -ones(size(x,1)-1,1)*x(1,:));
evals = errorSPM(x);
end

```

# Appendix B

## Statistical Fitting Cost Function

```
function f = errorSPM(x)
global SPMtot
global what
global Points
Pointsin = Points;
if length(x)==10
x1=[x(1); 0; x(2:end)];
else
    x1=x;
end

C1 =zeros(45,45);
C1(41,41)= x1(5)+sqrt(-1)*x1(6);
C1(41,40)= x1(8)+sqrt(-1)*x1(7);
C1(40,41)= x1(9)+sqrt(-1)*x1(7);
C1(41,42)= x1(10)+sqrt(-1)*x1(7);
C1(42,41)= x1(11)+sqrt(-1)*x1(7);

r=what;
fil = [x1(3)+sqrt(-1)*x1(4) x1(1)+sqrt(-1)*x1(2) x1(3)+sqrt(-1)*x1(4)];
for yy=1:r+1
    pointsrow = [zeros(1,4) Pointsin zeros(1,4)];
    SPM=[];
```

```

start =1+4;
final=length(pointsrow)-4;
for h = 5:final
    Sum = pointsrow(h+0)*C1(41,40)*pointsrow(h-1)*...
        conj(pointsrow(h-1))+...%0-1
        pointsrow(h-1)*C1(40,41)*pointsrow(h+0)*...
        conj(pointsrow(h-1))+... %-1&0
        pointsrow(h+0)*C1(41,41)*pointsrow(h+0)*...
        conj(pointsrow(h))+... % 0&0
        pointsrow(h+1)*C1(42,41)*pointsrow(h+0)*...
        conj(pointsrow(h+1))+...% 1&0
        pointsrow(h+0)*C1(41,42)*pointsrow(h+1)*...
        conj(pointsrow(h+1));
    SPM=[SPM Sum];
end
pointsrow(5:end-4)=pointsrow(5:end-4)+SPM;
Pointsin = pointsrow(5:end-4);
Pointsin=conv(Pointsin,fil);
end
SPMy = ((Points-Pointsin(r+2:end-r-1)));
err=mean(abs(SPMMy(200:length(Points)-200)-...
    SPMtot(200:length(Points)-200)).^2);
f=err;
end

```

# Appendix C

## Algebraic Fitting Cost Function

```
function f = Her2(x)
global C
global center
load('Hicsizinja')
coe = 1;
ch1re =x(1);
ch1im =x(2);
ch2re= x(3);
ch2im =x(4);
C00re = x(5);
C00im = x(6);
C01im =x(7);
c01re =x(8);
c0m1re =x(9);
cm10re =x(10);
C10re = x(11);
c01=c01re+C01im*1i;
c10=C10re+C01im*1i;
c00=C00re+1i*C00im;
c0m1=c0m1re+C01im*1i;
cm10=cm10re+C01im*1i;
alpha= ch1re+ch1im*1i;
beta = ch2re+ch2im*1i;
for i=1:10
```



```

    coetotal2=[];
    coetotal=[];
    coeinja =coe;
    coeinja(ind1{i},:)=[];
    co33con=[];
    sizecoeinja{i} = size(coeinja,1);
    parfor ww=1:sizecoeinja{i}
        co33con = [co33con;repmat(coeinja(ww,:),size(coeinja,1),1).*coeinja];
    end
    co33 =[];
    sizecosscon{i} = size(co33con,1);
    parfor ww=1:sizecoeinja{i}
        co33=[co33;repmat(coeinja(ww,:),size(sizecosscon{i},1),1).*co33con];
    end
    coespm3 = [(c0m1+cm10)*co33;c00*co33;(c01+c10)*co33];
    coetotal = [alpha*coe;beta*coe;beta*coe;alpha*coespm3;...
    beta*coespm3;beta*coespm3];
    coetotal( ind2{i})=[];
    icu = unique( zakhireic{i});
    for gg=1:length(icu)
        coetotal2(icu(gg)) = sum(coetotal(find(zakhireic{i}==icu(gg))));
    end
    coe=conj(coetotal2');
    end
    sumh = sum(H,2);
    ind = find(sumh==3);
    error =0;
    counter = 0;
    for bb=1:size(ind,1)
        templ = H(ind(bb),1:41);
        tempr = H(ind(bb),42:end);
        indinl = find(templ>0);
        indinr = find(tempr>0);
        if length(indinr)>1
            continue
        end
        dif = indinl-21;
        if length(dif)==1

```

```

        dif = [dif dif];
    end
    if indinr == sum(dif)+21
        counter = counter+1;
        indin(counter) = ind(bb);
        coeC(counter) = C(center+dif(1),center+dif(2))+...
            C(center+dif(2),center+dif(2));
    end
end
coeC=conj(-coeC');
coe2 = coe(indin,:);
H2 = H(indin,:);
coe = coe2;
H = H2;
for i=11
    coetotal2=[];
    coetotal=[];
    sumh=sum(H,2);
    coeinja =coe;
    coeinjaC = coeC;
    co33con=[];
    co33conC=[];
    sizecoeinja{i} = size(coeinja,1);
    for ww=1:size(coeinja,1)
        co33con = [co33con;repmat(coeinja(ww,:),size(coeinja,1),1).*coeinja];
        co33conC = [co33conC;repmat(coeinjaC(ww,:),...
            size(coeinjaC,1),1).*coeinjaC];
    end
    co33 =[];
    co33C =[];
    sizeco33con{i} = size(co33con,1);
    [a1 b1] = meshgrid([1:size(coeinja,1)],[1:size(co33con,1)]);
    all1 = [a1(:) b1(:)];
    co33 = coeinja(all1(:,1),:)+co33con(all1(:,2),:);
    co33C = coeinjaC(all1(:,1),:)+co33conC(all1(:,2),:);
    coespm3 = [(c0m1+cm10)*co33;c00*co33;(c01+c10)*co33];
    coespm3C = [(C(center,center-1)+C(center-1,center))*co33C;...
        C(center,center)*co33C;(C(center,center+1)+...

```

```
        C(center+1,center))*co33C];
coetotal = [alpha*coe;beta*coe;beta*coe;alpha*coespm3;...
beta*coespm3;beta*coespm3];
coetotalC = [coeC;coeC;coeC;coespm3C;coespm3C;coespm3C];
end
f = mean(abs([1-alpha^11; coetotalC-coetotal]).^2)
end
```

## Appendix D

# MATLAB Code for Testing Simplex Result

```
string = sprintf('UWCmatrix%dSpanNDSF',9+1);
load(string,'C')
C(find(isnan(abs(C))))=0;
data = randi(16,1,10000);
poi1 =0.2236 ;
poi2 = 0.6708;
Cnst=[ poi1+poi1*1i  -poi1+poi1*1i  poi1-poi1*1i  -poi1-poi1*1i...
        poi1+poi2*1i  -poi1+poi2*1i,...
        poi1-poi2*1i  -poi1-poi2*1i  poi2+poi1*1i  -poi2+poi1*1i...
        poi2-poi1*1i  -poi2-poi1*1i,...
        poi2+poi2*1i  -poi2+poi2*1i  poi2-poi2*1i  -poi2-poi2*1i];

Points = Cnst(data);
Pointsin = Points;
x1=x
C1 =zeros(45,45);
C1(41,41)= x1(5)+sqrt(-1)*x1(6);
C1(41,40)= x1(8)+sqrt(-1)*x1(7);
C1(40,41)= x1(9)+sqrt(-1)*x1(7);
C1(41,42)= x1(10)+sqrt(-1)*x1(7);
C1(42,41)= x1(11)+sqrt(-1)*x1(7);
r=10;
```

```

fil = [x1(3)+sqrt(-1)*x1(4) x1(1)+sqrt(-1)*x1(2) x1(3)+sqrt(-1)*x1(4)];
for yy=1:r+1
pointsrow = [zeros(1,4) Pointsin zeros(1,4)];
SPM=[];
start =1+4;
final=length(pointsrow)-4;
for h = 5:final
    Sum = pointsrow(h+0)*C1(41,40)*pointsrow(h-1)*conj(pointsrow(h-1))+...
    pointsrow(h-1)*C1(40,41)*pointsrow(h+0)*conj(pointsrow(h-1))+...
    pointsrow(h+0)*C1(41,41)*pointsrow(h+0)*conj(pointsrow(h))+...
    pointsrow(h+1)*C1(42,41)*pointsrow(h+0)*conj(pointsrow(h+1))+...
    pointsrow(h+0)*C1(41,42)*pointsrow(h+1)*conj(pointsrow(h+1));
    SPM=[SPM Sum];
end
pointsrow(5:end-4)=pointsrow(5:end-4)+SPM;
Pointsin = pointsrow(5:end-4);
Pointsin = conv(Pointsin,fil);
end
SPMy = ((Points-Pointsin(r+2:end-r-1)));
SPMtoty = SPMcal2(C,Points);
SPMcomp1 = SPMcal2(C,Points-SPMtoty);
total1 = SPMcomp1+Points-SPMtoty;
SPMfinal1 = Points-total1;
SPMtot = SPMcal2(C,Points-SPMy);
total = SPMtot+Points-SPMy;
SPMfinal = Points-total;
mean(abs(SPMtoty).^2)
mean(abs(SPMfinal).^2)
mean(abs(SPMfinal1).^2)
%%
close all
figure
hold all
plot(real(SPMtoty+Points),imag(SPMtoty+Points),'*')
plot(real(total1),imag(total1),'*')
plot(real(total),imag(total),'*')
plot(real(Cnst),imag(Cnst),'*')
legend('Uncompensated', 'Compensated by using C matrix',...

```

'Compensated by using the model', 'Constellation points')

# Appendix E

## MATLAB Code for Compensation with C matrix

```
clear all
clc
close all
global Points
global SPMtot
global what
data = randi(16,1,10000);
poi1 =0.2236 ;
poi2 = 0.6708;
Cnst=[ poi1+poi1*i  -poi1+poi1*i  poi1-poi1*i  -poi1-poi1*i...
        poi1+poi2*i  -poi1+poi2*i,...
        poi1-poi2*i  -poi1-poi2*i  poi2+poi1*i  -poi2+poi1*i ...
        poi2-poi1*i  -poi2-poi1*i,...
        poi2+poi2*i  -poi2+poi2*i  poi2-poi2*i  -poi2-poi2*i];
Points = Cnst(data);
string = sprintf('UWCmatrix%dSpanNDSF',9+1);
load(string,'C')
C(find(isnan(abs(C))))=0;
Pointsin2 = Points;
hold all
for hj = 1:6
    SPMtot2 = SPMcal2(C,Pointsin2);
```

```
plot(real(Pointsin2+SPMtot2),imag(Pointsin2+SPMtot2),'.')
fin = abs(Points-Pointsin2-SPMtot2).^2;
SPMtot = Points-Pointsin2-SPMtot2;
mean(abs(SPMtot).^2)
Pointsin2 = Points-SPMtot2;
end
legend('Uncompensated','Compensated 1 time','Compensated 2 times',...
'Compensated 3 times','Compensated 4 times','Compensated 5 times')
```



# Appendix F

## MATLAB code for Data Masking

```
clear all
clc
close all
string = sprintf('UWCmatrix%dSpanNDSF',9+1);
load(string,'C')
C(find(isnan(abs(C))))=0;
delete(gcf)
parpool(60)
data = randi(16,1,128*100)-1;
%
poi1 =0.2236 ;
poi2 = 0.6708;
Cns=[ poi1+poi1*i  -poi1+poi1*i   poi1-poi1*i  -poi1-poi1*i...
      poi1+poi2*i  -poi1+poi2*i,...
      poi1-poi2*i  -poi1-poi2*i   poi2+poi1*i  -poi2+poi1*i...
      poi2-poi1*i  -poi2-poi1*i,...
      poi2+poi2*i  -poi2+poi2*i   poi2-poi2*i  -poi2-poi2*i];
points = Cns(data+1);
rng(randi(50))
mask1 = randi(16,1,50)-1;
rng(randi(50))
mask2 = randi(16,1,50)-1;
rng(randi(50))
mask3 = randi(16,1,50)-1;
```

```

rng(randi(50))
mask4 = randi(16,1,50)-1;
maskha = [mask1;mask2;mask3;mask4];
%
pointsrowfirst = [zeros(1,2*50) data zeros(1,2*50)];

%
for kk=1:258
kk
datato = [pointsrowfirst((kk-1)*50+1:(kk+2)*50)];
SPM=[];
for tt=1:4
maskchosen = [zeros(1,50) maskha(tt,:) zeros(1,50)];
masks = mod(datato+maskchosen,16);
Points = Cns(masks+1);
Pointsin2 = Points;
for hj = 1:4
    SPMtot2 = SPMcal2(C,Pointsin2);
    SPMtot = Points-Pointsin2-SPMtot2;
    Pointsin2 = Points-SPMtot2;
end
SPMdic(tt)=mean(SPMtot.*conj(SPMtot));
end
[mag(kk) ja(kk)]=min(SPMdic);
maskchosen = [zeros(1,50) maskha(ja(kk),:) zeros(1,50)];
masks = mod(datato+maskchosen,16);
pointsrowfirst((kk-1)*50+1:(kk+2)*50)= masks;
end
%
clc
SPM=[];
pointsrow = Cns(pointsrowfirst(100:end-100)+1);
pointsrow = [zeros(1,200) pointsrow zeros(1,200)];
lengthframe = (size(C,1)-1)/2;
for f = 1+2*lengthframe:length(pointsrow)-2*lengthframe
    Sum=0;
    for m=-lengthframe:lengthframe
        for n = -lengthframe:lengthframe

```

```

                Sum = Sum+ pointsrow(f+m)*...
                    C(m+lengthframe+1,n+lengthframe+1)...
                    *pointsrow(f+n)*conj(pointsrow(f+m+n));
            end
        end
        SPM(f-2*lengthframe)=Sum;
    end
    SPM = SPMcal2(C,pointsrow);
    mean(SPM(100:end-100).*conj(SPM(100:end-100)));
    Pointsin2 = pointsrow;
    hold all
    for hj = 1:6

        SPMtot2 = SPMcal2(C,Pointsin2);
        plot(real(Pointsin2+SPMtot2),imag(Pointsin2+SPMtot2),'.')
        SPMtot = pointsrow-Pointsin2-SPMtot2;
        mean(abs(SPMtot).^2)
        Pointsin2 = pointsrow-SPMtot2;

    end
    mean(abs(SPMtot).^2);
    legend('Uncompensated','Compensated 1 time','Compensated 2 times',...
        'Compensated 3 times','Compensated 4 times','Compensated 5 times')

    % %
    % %
    % %
    % %%
    %
    clc
    SPM = [];
    datatotal = mod(data,16)+1;
    Points = Cns(datatotal);
    pointsrow = reshape(Points,1,size(Points,1)*size(Points,2));

```

```

pointsrow = [zeros(1,200) pointsrow zeros(1,200)];
lengthframe = (size(C,1)-1)/2;

for f = 1+2*lengthframe:length(pointsrow)-2*lengthframe
    Sum=0;
    for m=-lengthframe:lengthframe
        for n = -lengthframe:lengthframe
            Sum = Sum+ pointsrow(f+m)*...
                C(m+lengthframe+1,n+lengthframe+1)...
                *pointsrow(f+n)*conj(pointsrow(f+m+n));
        end
    end
    SPM(f-2*lengthframe)=Sum;
end
SPM = SPMcal2(C,Points);
mean(SPM(100:end-100).*conj(SPM(100:end-100)));

Pointsin2 = Points;
hold all
for hj = 1:6

    SPMtot2 = SPMcal2(C,Pointsin2);
    plot(real(Pointsin2+SPMtot2),imag(Pointsin2+SPMtot2),'.')
    SPMtot = Points-Pointsin2-SPMtot2;
    mean(abs(SPMtot).^2)
    Pointsin2 = Points-SPMtot2;

end
mean(abs(SPMtot).^2);
legend('Uncompensated','Compensated 1 time','Compensated 2 times',...
'Compensated 3 times','Compensated 4 times','Compensated 5 times')

%%
for kk=1:256

```

```

kk
datato = [pointsrowfirst((kk-1)*50+1:(kk+8)*50)];

maskchosen = [zeros(1,200) maskha(tt,:) zeros(1,200)];

masks = mod(datato+maskchosen,16);

maskchosen = [zeros(1,200) maskha(ja(kk),:) zeros(1,200)];
masks = mod(datato+maskchosen,16);
pointsrowfirst((kk-1)*50+1:(kk+8)*50)= masks;
end
dataret = pointsrowfirst;
pointsrowfirst = [zeros(1,2*100) data zeros(1,2*100)];
for kk=1:256
kk
datato = [pointsrowfirst((kk-1)*50+1:(kk+8)*50)];

maskchosen = [zeros(1,200) maskha(tt,:) zeros(1,200)];

masks = mod(datato+maskchosen+maskchosen,16);

maskchosen = [zeros(1,200) maskha(ja(kk),:) zeros(1,200)];
masks = mod(datato+maskchosen,16);
pointsrowfirst((kk-1)*50+1:(kk+8)*50)= masks;
end

%%
maskedfirst = pointsrowfirst(201:end-200);
temp =maskedfirst(1:25);
temptah = maskedfirst(end-25+1:end);
pointsremain = maskedfirst(26:end-25);

pointsrowfirst2 = [zeros(1,2*100) pointsremain zeros(1,2*100)];

%%
for kk=1:floor(length(pointsrowfirst2)/50-8)

```

```

datato = [pointsrowfirst2((kk-1)*50+1:(kk+8)*50)];
SPM=[];
for tt=1:4

% %% XOR
maskchosen = [zeros(1,200) maskha(tt,:) zeros(1,200)];

masks = mod(datato+maskchosen,16);

Points = Cns(masks+1);

pointsrow = reshape(Points,1,size(Points,1)*size(Points,2));

lengthframe = (size(C,1)-1)/2;

parfor f = 1+2*lengthframe:length(pointsrow)-2*lengthframe
    Sum=0;
    for m=-100:100
        for n = -100:100
            Sum = Sum+ pointsrow(f+m)*C(m+101,n+101)...
                *pointsrow(f+n)*conj(pointsrow(f+m+n));
        end
    end
    SPM(f-200)=Sum;
end
SPMdic(tt)=mean(SPM.*conj(SPM));
end
[mag(kk) ja(kk)]=min(SPMdic);
maskchosen = [zeros(1,200) maskha(ja(kk),:) zeros(1,200)];
masks = mod(datato+maskchosen,16);
pointsrowfirst2((kk-1)*50+1:(kk+8)*50)= masks;
end

pointsrowfirst2=[zeros(1,200) temptah...
    pointsrowfirst2(201:end-200) temptah zeros(1,200)];

```

```

%%
SPM = [];
maskropw = maskha(ja,:);
maskrow = reshape(maskropw',1,12800);
datatotal = mod(data,16)+1;
Points = Cns(datatotal);
pointsrow = Cns(pointsrowfirst2(201:end-200)+1);
pointsrow = [zeros(1,200) pointsrow zeros(1,200)];
lengthframe = (size(C,1)-1)/2;
parfor f = 1+2*lengthframe:length(pointsrow)-2*lengthframe
    Sum=0;
    for m=-100:100
        for n = -100:100
            Sum = Sum+ pointsrow(f+m)*C(m+101,n+101)*...
                pointsrow(f+n)*conj(pointsrow(f+m+n));
        end
    end
    SPM(f-200)=Sum;
end

mean(SPM(100:end-100).*conj(SPM(100:end-100)))

%

%
maskedfirst = pointsrowfirst(201:end-200);

pointsremain = maskedfirst;

pointsrowfirst3 = [zeros(1,2*100) pointsremain zeros(1,2*100)];

%%
for kk=1:floor(length(pointsrowfirst3)/50-8)
kk
datato = [pointsrowfirst3((kk-1)*50+1:(kk+8)*50)];
SPM= [];

```

```

for tt=1:4

% %% XOR
maskchosen = [zeros(1,200) maskha(tt,:) zeros(1,200)];

masks = mod(datato+maskchosen,16);

Points = Cns(masks+1);

pointsrow = reshape(Points,1,size(Points,1)*size(Points,2));

lengthframe = (size(C,1)-1)/2;

parfor f = 1+2*lengthframe:length(pointsrow)-2*lengthframe
    Sum=0;
    for m=-100:100
        for n = -100:100
            Sum = Sum+ pointsrow(f+m)*C(m+101,n+101)*...
                pointsrow(f+n)*conj(pointsrow(f+m+n));
        end
    end
    SPM(f-200)=Sum;
end
SPMdic(tt)=mean(SPM.*conj(SPM));
end
[mag(kk) ja(kk)]=min(SPMdic);
maskchosen = [zeros(1,200) maskha(ja(kk),:) zeros(1,200)];
masks = mod(datato+maskchosen,16);
pointsrowfirst3((kk-1)*50+1:(kk+8)*50)= masks;
end

pointsrowfirst3 = [zeros(1,200) pointsrowfirst3(201:end-200) zeros(1,200)];

SPM =[];
pointsrow = Cns(pointsrowfirst3(201:end-200)+1);
pointsrow = [zeros(1,200) pointsrow zeros(1,200)];

```



```

lengthframe = (size(C,1)-1)/2;
parfor f = 1+2*lengthframe:length(pointsrow)-2*lengthframe
    Sum=0;
    for m=-100:100
        for n = -100:100
            Sum = Sum+ pointsrow(f+m)*C(m+101,n+101)*...
                pointsrow(f+n)*conj(pointsrow(f+m+n));
        end
    end
    SPM(f-200)=Sum;
end

mean(SPM(100:end-100).*conj(SPM(100:end-100)))

```

## Appendix G

# MATLA Code for the Modified Version of Tree-code (Based on a code written by Mr. Hamid Ebrahimzadeh)

```
function [child_link,E,EE]=child_link_call(merge_vec,...
    power_set,SUM_N_LUT_INFORMATION,...
                                NBit_LUT)
global may C SPMplot SPMtest count SPMworkkonim SPMworkkiri
poi1 =0.2236 ;
poi2 = 0.6708;
Cns=[ poi1+poi1*i  -poi1+poi1*i   poi1-poi1*i  -poi1-poi1*i...
      poi1+poi2*i  -poi1+poi2*i,...
      poi1-poi2*i  -poi1-poi2*i   poi2+poi1*i  -poi2+poi1*i...
      poi2-poi1*i  -poi2-poi1*i,...
      poi2+poi2*i  -poi2+poi2*i   poi2-poi2*i  -poi2-poi2*i];
m= numel(merge_vec);
child_link = cell(1,m);
E=power_set;
E4 = E.^2;
E6 = E.^3;
ESPMs=0*E;
```

```

flag=0;
count=0;
for ii=1:m
    N = 2^m;
    L=numel(E);
    A=[];
    SPMs=[];
    SPMkarma=[];
    zeroha=[];
    A4 =[];
    A6 = [];
    I_Left=[];
%    flag=1;
    I_Right=[];
    for k1=1:L
        for k2=1:L
            mu =(E(k1)+E(k2))/2;
            if ii==1
                [px py] = meshgrid(cons(k1*4-[0 1 2 3]) ,cons(k2*4-[0 1 2 3]));
                pointsall = [px(:) py(:)];
                SPMsper=[];
                for f=1:size(pointsall,1)
                    pointswork = [zeros(1,2) pointsall(f,:) zeros(1,2)];
                    sumy=0;
                    for g=0:1
                        Sum = 0;
                        for m1=-1:(2^ii)-1
                            for n=-1:1
                                Sum = Sum+pointswork(g+m1+3)*...
                                    conj(pointswork(g+m1+n+3))*...
                                    C(101+m1,101+n)*pointswork(g+n+3);
                            end
                        end
                    sumy = sumy+abs(Sum).^2;
                    end
                SPMsper(f)=sumy;
                end
            SPMs=[SPMs mean(SPMsper)];

```

```

end
if ii==2&flag
tempchild = child_link{1};
[px py pz pk] = ndgrid(cons(tempchild(1,k1)*4-[0 1 2 3]) ,...
    cons(tempchild(2,k1)*4-[0 1 2 3]),...
    cons(tempchild(1,k2)*4-[0 1 2 3]) ,...
    cons(tempchild(2,k2)*4-[0 1 2 3]));
pointsall = [px(:) py(:) pz(:) pk(:)];
SPMsper=[];
for f=1:size(pointsall,1)
pointswork = [zeros(1,6) pointsall(f,:) zeros(1,6)];
sumy = 0;
for g=0:3
    Sum = 0;
for m1=-3:(2^ii)-1
    for n=-3:3
        Sum = Sum+pointswork(g+m1+7)*...
            conj(pointswork(g+m1+n+7))*...
            C(101+m1,101+n)*pointswork(g+n+7);
    end
end
sumy = sumy+abs(Sum).^2;
end
SPMsper(f)=sumy;
end
SPMs=[SPMs mean(SPMsper)];

end
if ii>2&(flag)
    SPMs = [SPMs 1*(ESPMs(k1)+ESPMs(k2)+(0)*ESPMs(k1)*ESPMs(k2))];
end
A=[A,mu*2];
A4 = [A4, E4(k1)+E4(k2)];
A6 = [A6, E6(k1)+E6(k2)];
I_Left=[I_Left,k1];
I_Right=[I_Right,k2];
end
if ii>=1&flag

```

```

    [sorti I] = sort(SPMs);
    A=A(I);
    SPMs = SPMs(I);

elseif 1&(((ii==2)|(ii>2))&flag)
    [sorti I] = sort(SPMs);
    I= I(1:2^(merge_vec(ii))*16);
    Asort=A(I);
    SPMs = SPMs(I);
    [ja maf] =unique(SPMs,'last');
    I2=[];
    tempsort = Asort(1:maf(1));
        [mags jas] = sort(tempsort);
        I2=[I2 jas];
    for bb=1:length(maf)-1
        tempsort = Asort(maf(bb)+1:maf(bb+1));
        indtemp = [maf(bb)+1:maf(bb+1)];
        [mags jas] = sort(tempsort);
        I2=[I2 indtemp(jas)];
    end
    if length(unique(I2))~=length(I2)
        count=count+1;
    end
    I=I(I2);
    A = A(I);
    SPMs = SPMs(I2);
end
if ~flag
    [A I]=sort(A);
end
A4 = A4(I);
A6 = A6(I);
A=A(1:min(2^SUM_N_LUT_INFORMATION(ii),numel(A)));
if flag
    SPMs=SPMs(1:min(2^SUM_N_LUT_INFORMATION(ii),numel(SPMs)));
end
A4=A4(1:min(2^SUM_N_LUT_INFORMATION(ii),numel(A4)));
A6=A6(1:min(2^SUM_N_LUT_INFORMATION(ii),numel(A6)));

```

```

I=I(1:numel(A));
I_Left=I_Left(I);
I_Right=I_Right(I);
if ii < m
    I_Left=reshape(I_Left,2^merge_vec(ii), []);
    I_Right=reshape(I_Right,2^merge_vec(ii), []);
    A4=reshape(A4,2^merge_vec(ii), []);
    A4=mean(A4,1);
    A=reshape(A,2^merge_vec(ii), []);
    if flag
        SPMs =reshape(SPMs,2^merge_vec(ii), []);
        SPMs=mean(SPMs,1);
    end
    A=mean(A,1);
    A6=reshape(A6,2^merge_vec(ii), []);
    A6=mean(A6,1);

    if log2(size(I_Left,2))> NBit_LUT(ii)
        I_Left=I_Left(:,1:2^ NBit_LUT(ii));
        I_Right=I_Right(:,1:2^ NBit_LUT(ii));
        A=A(1:2^NBit_LUT(ii));
        if flag
            SPMs=SPMs(1:2^NBit_LUT(ii));
        end
        A4=A4(1:2^NBit_LUT(ii));
        A6=A6(1:2^NBit_LUT(ii));
    end
end

else

    I_Left=I_Left(1:2^merge_vec(ii));
    I_Right=I_Right(1:2^merge_vec(ii));
    A=A(1:2^merge_vec(ii));
    if flag
        SPMs=SPMs(1:2^merge_vec(ii));
    end
    A4=A4(1:2^merge_vec(ii));
    A6=A6(1:2^merge_vec(ii));

```

```
end
child_link{ii} = [I_Left;I_Right];
E=A;
E4=A4;
E6 =A6;
if flag
ESPMs=SPMs;
end
EE{ii}=E;

end
E=mean(E)/(2^m);
```

# Appendix H

## MATLAB Code for Finding the Average SPM Noise for Energy Vectors

```
clear all
clc
close all
load('Cmatrix_NDSF_SPM','C')
load('Child','child_link')
poi1 =0.2236 ;
poi2 = 0.6708;
Cnst116qam=[ poi1+poi1*i -poi1+poi1*i   poi1-poi1*i -poi1-poi1*i...
  poi1+poi2*i -poi1+poi2*i,...
  poi1-poi2*i -poi1-poi2*i   poi2+poi1*i -poi2+poi1*i...
  poi2-poi1*i -poi2-poi1*i,...
  poi2+poi2*i -poi2+poi2*i   poi2-poi2*i -poi2-poi2*i];
link4d = child_link{2};
link2d = child_link{1};
total=[];
for w= 1:16
templink = link4d(:,w)';
linkleft = templink(1:8);
linkright = templink(9:16);
subleft = link2d(:,linkleft);
```



```

subright = link2d(:,linkright);
total = [total [subleft;subright]];
end
tt2=general_cartesian_prod([1:128],2);
tt = [total(:,tt2(1,:));total(:,tt2(2,:))];
signs = general_cartesian_prod([1 2 3 4],8)-1;
for ja=1:size(tt,2)
    ja
    dataiter = tt(:,ja)';
    signfor = randperm(size(signs,2),4^8);
    parfor sig = 1:4^8
        signiter = signs(:,signfor(sig))';
        Points = Cnst116qam(dataiter*4-signiter);
        SPM = SPMcal(C,Points);
        SPMha(sig)=((SPM));
    end
    SPMvar(ja)=mean((SPMha));
    var(SPMha);
end
SPMworkkonim = zeros(16,16,length(SPMvar));
for yy=1:length(SPMvar)
    tttemp = tt2(:,yy);
    k1 = ceil(tttemp(1)/8);
    k2 = ceil(tttemp(2)/8);
    SPMworkkonim(k1,k2,yy)=SPMvar(yy);
end

```

# Appendix I

## MATLAB Code for Joint Detection

### I.1 Codes for Calculating XPMs

```
function XPMtot = xpmcal1w(pointsA,pointsB,Cxpm)
lengthframe = 0.5*(size(Cxpm,1)-1);
pointsrowA = [zeros(1,2*lengthframe) pointsA zeros(1,2*lengthframe)];
pointsrowB = [zeros(1,2*lengthframe) pointsB zeros(1,2*lengthframe)];
XPM=[];
parfor f = 1+2*lengthframe:length(pointsrowA)-2*lengthframe
    Sum=0;
    for m=-lengthframe:lengthframe
        for n = -lengthframe:lengthframe
            Sum = Sum+ pointsrowA(f+m)*...
                    Cxpm(m+lengthframe+1,n+lengthframe+1)*...
                    pointsrowB(f+n)*conj(pointsrowB(f+m+n));
        end
    end
    XPM=[XPM Sum];
end
XPMtot = XPM;
end
```

```
function XPMtot = xpmcal2w(pointsA,pointsB,Cxpm)
lengthframe = 0.5*(size(Cxpm,1)-1);
```

```

pointsrowA = [zeros(1,2*lengthframe) pointsA zeros(1,2*lengthframe)];
pointsrowB = [zeros(1,2*lengthframe) pointsB zeros(1,2*lengthframe)];
XPM=[];
parfor f = 1+2*lengthframe:length(pointsrowA)-2*lengthframe
    Sum=0;
    for m=-lengthframe:lengthframe
        for n = -lengthframe:lengthframe
            Sum = Sum+ pointsrowA(f+m)*...
                    Cxpm(m+lengthframe+1,n+lengthframe+1)*...
                    pointsrowB(f+n)*conj(pointsrowB(f+m+n));
        end
    end
    XPM=[XPM Sum];
end
XPMtot = XPM;
end
function XPMtot = xpmcal3w(pointsA,pointsB,Cxmpol)
lengthframe = 0.5*(size(Cxmpol,1)-1);
pointsrowA = [zeros(1,2*lengthframe) pointsA zeros(1,2*lengthframe)];
pointsrowB = [zeros(1,2*lengthframe) pointsB zeros(1,2*lengthframe)];
XPMpol=[];
parfor f = 1+2*lengthframe:length(pointsrowA)-2*lengthframe
    Sumpol =0;
    for m=-lengthframe:lengthframe
        for n = -lengthframe:lengthframe
            Sumpol = Sumpol+ pointsrowB(f+m)*...
                    Cxmpol(m+lengthframe+1,n+lengthframe+1)*...
                    pointsrowA(f+n)* conj(pointsrowB(f+m+n));
        end
    end
    XPMpol=[XPMpol Sumpol];
end
XPMtot = XPMpol;
end
function XPMtot = xpmcal4w(pointsA,pointsBx, pointsBy,Cxmpol)
lengthframe = 0.5*(size(Cxmpol,1)-1);
pointsrowA = [zeros(1,2*lengthframe) pointsA zeros(1,2*lengthframe)];
pointsrowBx = [zeros(1,2*lengthframe) pointsBx zeros(1,2*lengthframe)];

```

```

pointsrowBy = [zeros(1,2*lengthframe) pointsBy zeros(1,2*lengthframe)];
XPMpol=[];
parfor f = 1+2*lengthframe:length(pointsrowA)-2*lengthframe
    Sumpol =0;
    for m=-lengthframe:lengthframe
        for n = -lengthframe:lengthframe
            Sumpol=Sumpol+ pointsrowBx(f+m)*...
                Cxmpol(m+lengthframe+1,n+lengthframe+1)*...
                pointsrowA(f+n)*conj(pointsrowBy(f+m+n));
        end
    end
    XPMpol=[XPMpol Sumpol];
end
XPMtot = XPMpol;
end

```

## I.2 MATLAB Code for Calculating Joint Probability Density Functions

```

clear all
clc
close all
poi1 =0.2236 ;
poi2 = 0.6708;
Cnst=[ poi1+poi1*i  -poi1+poi1*i  poi1-poi1*i  -poi1-poi1*i...
    poi1+poi2*i  -poi1+poi2*i,...
        poi1-poi2*i  -poi1-poi2*i  poi2+poi1*i  -poi2+poi1*i...
    poi2-poi1*i  -poi2-poi1*i,...
        poi2+poi2*i  -poi2+poi2*i  poi2-poi2*i  -poi2-poi2*i];
load('datatotal')
XPMall = XPMtot21w1+XPMtot21w2+XPMtot21w3+...
        XPMtot21w4+XPMtot23w1+XPMtot23w2+...
        XPMtot23w3+XPMtot23w4+...
        XPMtot24w1+XPMtot24w2+...
        XPMtot24w3+XPMtot24w4+XPMtot25w1+...
        XPMtot25w2+XPMtot25w3+XPMtot25w4;

```

```

SNRplot =10.5;
sigma2 = mean(abs(PointsAx).^2)/(10^((SNRplot(1)/10)));
noise = sqrt(sigma2/2)*randn(1,length(PointsAx))+sqrt(sigma2/2)*...
        randn(1,length(PointsAx))*1i;
delta = PointsAx+XPMall+noise;
final = [dataAx(1:end-2)' dataAx(2:end-1)' dataAx(3:end)'];
final = [final conj(delta(1:end-2))' conj(delta(2:end-1))'...
        conj(delta(3:end))'];
[p pp ppp] = unique(final(:,1:3),'rows');
XPMpairtemp = cell(1,4096);
parfor h =1:size(p,1)
    h
    temp = find(ppp==h);
    jaha{h} = temp;
    leng(h) = length(temp);
    tt = XPMpairtemp{h};
    XPMpairtemp{h}=[tt ;final(temp,4:end)];
end
for kk=1:4096
    kk
    tempr = XPMpairtemp{kk};
    tempr2 = mean(tempr,1);

    meany = reshape([real(tempr2) ;imag(tempr2)],1,6);
    covsix{kk} = mean(covforsix(tempr,meany),3);
    meansix{kk} = meany;
end

```

### I.3 Matlab Code for Finding Covariance Matrices

```

function out = covforsix(A,meany)
temp = A;
for hjj =1:size(temp,1)
temp2 = reshape([real(temp(hjj,:));imag(A(hjj,:))],1,6);
out(:, :,hjj) = (temp2-meany)'*(temp2-meany);
end

```

```
end
```

## I.4 Matlab Code for Finding Conditional Expected Values for XPM (Only one data symbol)

```
clear all
clc
close all
poi1 =0.2236 ;
poi2 = 0.6708;
Cnst=[ poi1+poi1*i  -poi1+poi1*i  poi1-poi1*i  -poi1-poi1*i...
      poi1+poi2*i  -poi1+poi2*i,...
      poi1-poi2*i  -poi1-poi2*i  poi2+poi1*i  -poi2+poi1*i...
      poi2-poi1*i  -poi2-poi1*i,...
      poi2+poi2*i  -poi2+poi2*i  poi2-poi2*i  -poi2-poi2*i];
load('datatotal')
XPMall = XPMtot21w1+XPMtot21w2+XPMtot21w3+...
        XPMtot21w4+XPMtot23w1+XPMtot23w2+...
        XPMtot23w3+XPMtot23w4+...
        XPMtot24w1+XPMtot24w2+...
        XPMtot24w3+XPMtot24w4+XPMtot25w1+...
        XPMtot25w2+ XPMtot25w3+XPMtot25w4;
delta = PointsAx+XPMall+0;
for tt=1:16
    tt
    tempin = indf{tt};
    tempol = delta(tempin);
    tempr2 = mean(tempol);
    meantemp1(tempin)=tempr2;
    totalck(tempin) = tempol-tempr2;
    var1(tt) =var(tempol-tempr2);
    meany = reshape([real(tempr2) ;imag(tempr2)],1,2);
    meantak{tt} = meany;
end
```

## I.5 Matlab Code for Comparing Joint and Minimum Distance Detection Methods

```
clear all
clc
% load('meanhatak')
%load('whiteandxpmcovs')
%load('vari')
%load('10kwithprobcanandpoints')
%load('sixsnr7p5')
%load('six0sqrt3')
load('eleaf5ch')
load('meantak')
load('six10p5')
close all
%%
poi1 =0.2236 ;
poi2 = 0.6708;

Cnst=[ poi1+poi1*i -poi1+poi1*i   poi1-poi1*i -poi1-poi1*i...
       poi1+poi2*i -poi1+poi2*i,...
       poi1-poi2*i -poi1-poi2*i   poi2+poi1*i -poi2+poi1*i...
       poi2-poi1*i -poi2-poi1*i,...
       poi2+poi2*i -poi2+poi2*i   poi2-poi2*i -poi2-poi2*i];
dataAx = randi(16,1,100000);
dataBx = randi(16,1,100000);
dataCx = randi(16,1,100000);
dataDx = randi(16,1,100000);
dataEx = randi(16,1,100000);
dataAy = randi(16,1,100000);
dataBy = randi(16,1,100000);
dataCy = randi(16,1,100000);
dataDy = randi(16,1,100000);
dataEy = randi(16,1,100000);
for ggg=1:16
indf{ggg}=find(dataAx==ggg);
end
```

```

PointsAx = Cnst(dataAx);
PointsBx = Cnst(dataBx);
PointsCx = Cnst(dataCx);
PointsDx = Cnst(dataDx);
PointsEx = Cnst(dataEx);
PointsAy = Cnst(dataAy);
PointsBy = Cnst(dataBy);
PointsCy = Cnst(dataCy);
PointsDy = Cnst(dataDy);
PointsEy = Cnst(dataEy);

Cspm2 = CMatrix{1};
Cspm2(find(isnan(Cspm2)))=0;
Cxpm12 = CMatrix{2};
Cxpm12(find(isnan(Cxpm12)))=0;
Cxmpol12 = CMatrix{3};
Cxmpol12(find(isnan(Cxmpol12)))=0;
Cxpm23 = CMatrix{4};
Cxpm23(find(isnan(Cxpm23)))=0;
Cxmpol23 = CMatrix{5};
Cxmpol23(find(isnan(Cxmpol23)))=0;
Cxpm24 = CMatrix{6};
Cxpm24(find(isnan(Cxpm24)))=0;
Cxmpol24 = CMatrix{7};
Cxmpol24(find(isnan(Cxmpol24)))=0;
Cxpm25 = CMatrix{8};
Cxpm25(find(isnan(Cxpm25)))=0;
Cxmpol25 = CMatrix{9};
Cxmpol25(find(isnan(Cxmpol25)))=0;

XPMtot21w1 = xpmcal1w(PointsAx,PointsBx, Cxpm12);
XPMtot21w2 = xpmcal2w(PointsAx,PointsBy, Cxpm12);
XPMtot21w3 = xpmcal3w(PointsAx,PointsBx, Cxmpol12);
%%
XPMtot21w4 = xpmcal4w(PointsAx,PointsBx, PointsBy,Cxmpol12);

```



```

XPMtot23w1 = xpmcal1w(PointsAx,PointsCx, Cxpm23);
XPMtot23w2 = xpmcal2w(PointsAx,PointsCy, Cxpm23);
XPMtot23w3 = xpmcal3w(PointsAx,PointsCx, Cxpmpol23);
XPMtot23w4 = xpmcal4w(PointsAx,PointsCx, PointsCy,Cxpmpol23);

XPMtot24w1 = xpmcal1w(PointsAx,PointsDx, Cxpm24);
XPMtot24w2 = xpmcal2w(PointsAx,PointsDy, Cxpm24);
XPMtot24w3 = xpmcal3w(PointsAx,PointsDx, Cxpmpol24);
XPMtot24w4 = xpmcal4w(PointsAx,PointsDx, PointsDy,Cxpmpol24);

XPMtot25w1 = xpmcal1w(PointsAx,PointsEx, Cxpm25);
XPMtot25w2 = xpmcal2w(PointsAx,PointsEy, Cxpm25);
XPMtot25w3 = xpmcal3w(PointsAx,PointsEx, Cxpmpol25);
XPMtot25w4 = xpmcal4w(PointsAx,PointsEx, PointsEy,Cxpmpol25);

XPMall = XPMtot21w1+XPMtot21w2+XPMtot21w3+...
          XPMtot21w4+XPMtot23w1+XPMtot23w2+...
          XPMtot23w3+XPMtot23w4+...
          XPMtot24w1+XPMtot24w2+XPMtot24w3+...
          XPMtot24w4+XPMtot25w1+XPMtot25w2+...
          XPMtot25w3+XPMtot25w4;
databits = de2bi(dataAx-1);
alpha =1;
SNRplot = [7:0.5:20];
for snri=1:length(SNRplot)
    snri
    sigma2 = mean(abs(PointsAx).^2)/(10^((SNRplot(snri)/10)));
    noise = sqrt(sigma2/2)*randn(1,length(PointsAx))+sqrt(sigma2/2)*...
            randn(1,length(PointsAx))*1i;
    noiseplot(snri,:) = noise;
    total = alpha*PointsAx + XPMall+noise;
    if SNRplot(snri)==10.5
    deteced = [];

```

```

parfor nn=1:length(total)-2
    nn
    tempnn = total(nn:nn+2);
    tempnn = reshape([real(tempnn);imag(tempnn)],1,6);
    prob = [];
    for jj=1:4096
        tempnomean = tempnn - meansix{jj};
        prob = [prob exp(-0.5*tempnomean*inv(covsix{jj})*tempnomean')/...
                sqrt(det(2*pi*covsix{jj})))];
    end
    [prob ja]= sort(prob,'descend');
    jaha{nn}=ja;
    sym = p(ja(1),:);
    deteced(nn) = sym(2);
end
detectbits = de2bi(deteced-1);
biterr(snri)=sum(sum(mod((detectbits+databits(2:end-1,:)),2)))...
/(4*(length(dataAx)-2));
symerr(snri) = sum(abs(sign((deteced)-dataAx(2:end-1))))/(length(dataAx)-2);
end
parfor kh=1:length(total)
    kh

    temp=[];
    for jg =1:16
        mrantempy = meantak{jg};
        temp = [temp abs(total(kh)-(alpha*(mrantempy(1)+mrantempy(2)*1i))).^2];
    end
    [detect(kh) detectja(kh)]= min(temp);

end
if SNRplot(snri)==7.5
for tt=1:16
    tt
    mrantempy = meantak{tt};
    tempin = indf{tt};
    meantemp1(tempin)=((alpha*(mrantempy(1)+mrantempy(2)*1i))+Cnst(tt));
end

```

```
total = PointsAx+XPMall - meantemp1;
plot(real(total),imag(total),'*')
end
detectbits = de2bi(detectja-1);
biterr2(snri) = sum(sum(mod((detectbits+databits),2)))/(4*length(dataAx));
symerr2(snri)=sum(abs(sign(dataAx-detectja)))/length(dataAx);
end
symerr2
```